



# PWilliams/JCarruthers/JHEvans/JKinsler



A WIRELESS WORLD PUBLICATION

circuit designs

In October 1972 Wireless World launched a circuit information system called CIRCARDS. The system was based on issued sets of 8" × 5" cards to complement personal filing systems. Published in conjunction with introductory articles in Wireless World, selected and tested circuits. CIRCARDS gave the subscriber descriptions of circuit operation, component values and ranges, circuit limitations, modifications, performance data and graphs.

This book is for those who design, use or understand electronic circuits. It contains information previously included on sets 11 to 20 of CIRCARDS,\* corrected where necessary and supplemented with 10 pages of additional circuits. Introductory articles originally published in *Wireless World* complement each section.

The CIRCARD system won an IPC Business Press award for the best innovation of 1973.

\*Sets 1 to 10 with all information was published in CIRCUIT DESIGNS No. 1, and is still in print.



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# circuit designs Collected Circards



PWilliams/JCarruthers/JHEvans/JKinsler Paisley College of Technology, Renfrewshire

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# Preface

Authors write the books they would like to read themselves. It is easy to start a book with this kind of incentive, but less easy to sustain it. Soon it becomes obvious that the author might end up as the only reader-a receptive but uncritical audience. Circards began as a response to the inordinate amount of time taken to track down a variety of solutions to each problem in the field of circuit design as it occurred. Engineers working in industry must often produce rapid solutions to problems, and rely on the knowledge and skills acquired in related areas. It is valuable to reconsider alternative approaches, since producing solutions in a hurry can so easily become a reflex action with all the possibilities of error that this implies. The question then was how to find a variety of approaches without this time delay. A first possibility was simply to make use of one of the excellent data retrieval systems (INSPEC, ISI, etc.). This would have given a list of articles on the topic together with an abstract of each-but not the circuit diagram. The time taken for the individual to track down each of the references was going to be too long, particularly as he would still have to digest them all sufficiently well to compare their merits and demerits for his particular application.

In our own work we were involved in such searches on a regular basis: in the preparation of lecture notes on unfamiliar topics; as a by-product of research for higher degrees; in the answering of queries passed on to us from local industry entering some area of electronics with which they were not yet familiar. As we had to carry out these functions anyway, it seemed to us that if we performed them in a more efficient and concentrated manner the results would be useful to a wide range of colleagues in the world of electronics.

We talked to colleagues from our own and other colleges at meetings and conferences; we quizzed industrial designers when visiting students carrying out their industrial training; we looked at the needs of amateurs trying to design their own equipment, and of technicians building circuits to often vague and contradictory specifications; and we tried listening to as well as talking to our students on their difficulties in converting theory into practice. To all these we would like to pass on our thanks. Their comments and opinions convinced us that there was a need for a circuit design service of the kind provided by Circards, and on which this book is based.

This need could not be met simply by collecting together and reprinting articles on a given topic drawn from various sources. Firstly there would be no pattern to such a collection and no easy way of comparing the performances or modes of operation. Secondly the viewpoint on each circuit would be that of the original authors; though they would be the obvious source for a discussion of the merits of their circuits, an independent judgement might be lacking. Thirdly, any measured data might be applicable only to a particular configuration or to given devices or component values. To retain the advantages of a collection of alternative solutions, while attempting to avoid the above difficulties something different was needed.

Our proposal was that we would track down the data on a given class of circuits as far as we could, and then

select a number of versions for further study. After carrying out experimental work (to supplement that given in the original reference, to modify the circuit to more familiar devices or to extend its operating range in respect of supply voltages, currents, etc.) these would be written up in a standard format. Particular attention would be given to the range of operation, to the possible variation in component values and to circuit modifications. In the process it was hoped that a fair sprinkling of novel circuits would be uncovered, and that unifying features would emerge to simplify the interchange of design techniques between one class of circuit and another. The format decided on was that of a standard file card. so that once a particular circuit had been identified by the reader as being of interest, that card could be removed for bench use while leaving the rest of the information on file. For more leisurely reference and library use this data is now collected into the present book form, together with an accompanying set of articles describing the principles underlying each topic. The original data is supplemented by a selection from recently published designs in each area, and we have tried to remove those errors and ambiguities discovered on the original cards.

In making these proposals to *Wireless World* we were delighted to find that they were received sympathetically, and would like to record our appreciation of their efforts and skills in bringing this project to fruition. Our thanks must also go to the Principal, Governors and Staff of Paisley College for their continuing encouragement. To our wives and families we can only say that without their support and encouragement this task which has brought us much satisfaction might have been started but could never have been sustained. *Peter Williams, Paisley* 

# Postscript

"Why didn't they ...?" Inevitably this will be the reaction of readers experienced in the design of particular circuit types described in this book. The response is simple ... "We didn't know!".

We try to blend bench investigations with as wide a study of the technical and professional journals as possible. Fortunately we have a varied background experience, and the help and advice of colleagues to fall back on. What none of this can provide is a sense of the present and ever-changing needs of readers. In choosing topics for future publication we would welcome advice from any source—we know that errors have slipped through in the past, and have corrected in this book those that we have traced. It would be easy to minimize these errors, by sticking to standard and well-tried circuits, but that is not enough in our view. For Circards to meet the original aims there has to be a balance between the classical, the industrial standards and the novel; users and designers of electronic circuits can then take this material as the starting point for their own designs.

Information and comments on such developments would help us greatly in planning and writing future "Circuit Design" books, and would make sure that the balance of the material suits the needs of the user.

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# Set 11: Basic logic gates

The W.W. article introducing this subject must have been of limited value for newcomers to logic as a result of a few typographical errors that crept in. These are corrected of course, the main errors being in lines 29 & 30 of Table 1 and in Table 6. The article forms a good introduction to Boolean algebra rules, truth tables and logic symbology (but for details of the Karnaugh map technique see article on page 60). The cards detail the different realizations of logic gates, card 4 being especially useful in summarizing the different kinds of NAND gate (standard, low power, high speed and Schottky). Card 8 is arguably the most useful giving circuits for interfacing between different kinds of logic circuits. Interfacing with analogue circuitry to form shunt or series choppers. as used in multiplexers, is covered in card 11. Three cards deal with newer kinds of logic systems. Card 9 describes the nomenclature used in threshold logic-a generalized approach of which the simple gates form special cases. Optical logic gates, three-state logic and majority gates are covered on cards 10 & 12. Set 16 card 4, and set 18 card 2 give logic circuits using current differencing amplifiers.

Resistor-transistor and direct-coupled gates 1 Diode-transistor gates 2 Basic t.t.l. gate 3 NAND gate variations in t.t.l. 4 Complementary m.o.s. gates—I 5 Complementary m.o.s. gates—II 6 Emitter-coupled logic 7 Interfacing 8 Threshold logic 9 Optical logic 10 Analogue gates 11 Three-state and majority logic 12

# **Basic logic gates**

Logical or arithmetic processes are extensively used in systems such as industrial control, computers, electronic instrumentation and automatic telephone exchanges. These processes often involve complex functions of several variables, the desired functions being realized by switching operations in a logical manner. Although much of the design of these systems now deals with the interconnection of complex functional blocks, successful results also depend on a knowledge of the basic elements that constitute the complex functional blocks.

The basic elements of such systems are logic gates, which may perform combinational operations on their inputs. These inputs will normally be in one of two allowed states that could be, for example, two different voltages, two different currents or two different resistance values such as the limiting cases of open circuit and short circuit. Whatever form the allowed states take, a logic gate is concerned with whether certain statements about its inputs, at a given instant, are true or false. If these statements are made using normal language they become unmanageable as the number of quantities involved increases, making some form of symbolic statement highly desirable.

If a certain statement is true it is assigned the value 1 and if it is false it is given the value 0. For example, if one of the inputs to a logic gate is called A and it can be either at 5 V or 0 V then the statement "input A is at 5 V" may be true or false. If it is true than A =1 and if it is false then A = 0. If this gate has three inputs and its output, D, is only at 5 V (D = 1) when two of its inputs, A and B, are at 5 V and its other input, C, is at 0 V, then D = 1when  $\mathbf{A} = 1$  AND  $\mathbf{B} = 1$  AND  $\mathbf{C} = 0$ . Now C = 0 implies that C is NOT 1 i.e.  $\dot{\mathbf{C}} = 1$ , where the bar indicates NOT or negation, so the above statement could be simply written as D = A AND BAND C. Using the multiplication sign of normal algebra ( $\times$  or .) to represent the AND operation this statement becomes  $D = A \times B \times C$ , or D = A.B.C, or even D = ABC where the "multiplication" (AND) signs are implied. This type of algebra, based on logical statements that

#### TABLE 1. Properties of Boolean algebra.

$ \begin{array}{cccccccccccccccccccccccccccccccccccc$	$ \begin{array}{ll} B &= A \\ B &= A + B \\ + B &= A \\ = A \end{array} $
---	---

# Table 2. Boolean theorems in terms of relay contacts.







are true or false, is called Boolean algebra and it is a very useful tool in the development of logical thinking and in the design of digital circuits and systems.

As well as the AND and NOT operations it is necessary to postulate the OR operation which is represented by the (+)symbol of normal algebra. For example, if a logic gate has two inputs A and B, and its output D is in the logic 1 state when either A or B is in the logic 1 state this statement can be written as D = AOR B which is represented by D = A+B.

A logic gate is an example of a basic logical circuit, called a combinational circuit, the output of which at a given instant is determined by the state of its inputs. Irrespective of its complexity, certain relationships, laws and simplification rules of Boolean algebra can be used to represent or investigate the behaviour of a combinational circuit. Using up to three variables, Table 1 shows some of the properties of this algebra some of which are the same as ordinary algebra. In Boolean algebra division and subtraction have no meaning and the variables can only have the values 0 or 1. Table 2 shows the Boolean algebra theorems relating the values 0 and 1 in terms of relay contacts that are either open (logic 0) or closed (logic 1). Table 3 illustrates the Boolean algebra theorems in one variable A in similar terms, where A can have either of the states 0 (Acontact open) or 1(A-contact closed). In Table 1 relations 26 & 27 together are known as De Morgan's theorem and 29 & 30 are identical with 26 & 27 except that the variables have been negated (or inverted or complemented).

Combinational logic circuits may take many different forms, one of which employs relay contacts which is useful for illustrating some of the simple Boolean relations. For example, in Figs 1 & 2, A, B and C are contacts operated by relay coils (not shown) to complete a path between input and output. Thus, we are concerned with the statement "the connection between input and output is complete".

When this statement is true D = 1and when it is false D = 0. In Fig 1, D = 1 only when contacts A AND B AND C are closed simultaneously so the Boolean representation is D = A.B.C.Hence, series-connected contacts of the same type provide the AND operation. In Fig. 2, D = 1 when contacts A OR B or C are closed so the situation may be represented by D = A + B + C. If more than one contact is closed the above statement is still true, i.e. D = 1. Thus, parallel-connected contacts of the same type provide the OR (or "inclusive" OR) operation and the order in which they are wired or considered does not affect the truth of the statement.

The validity of a Boolean statement representing the behaviours of a combinational logic gate can be checked by means of a truth table, which is a tabular listing of all possible logic combinations of the variables and the resulting output logic. Tables 4 & 5 are the truth tables for Figs 1 & 2 respectively and they show that a complete truth table requires 2<sup>n</sup> rows to represent a gate having n variables. Table 6 is a listing of the truth tables for the commonly-used combinational logic operations and shows the names given to the logic gates used to realize these operations. The NOR (NOT OR) gate performs the complement of the OR function and the NAND (NOT AND) gate the complement of the AND function.

Fig. 1: D = 1 when contacts A AND B AND C are closed, represented by D = A.B.C.



Fig. 2. D = 1 when A OR B OR C are closed, represented by D = A + B + C.

#### TABLE 4. Truth table for Fig. 1

A	B	С	D
0	0	0	0
1	0	0	0
0	1	0	0
1	1	0	0
0	0	1	0
1	0	1	0
0	1	1	0
1	1	1	1



A	8	С	D
0	0.	0	0
1	0	0	1
0	1	0	1
1	1	0	1
0	0	1	1
1	0	1	1
0	1	1	1
1	1	1	1

Unlike the OR gate, the "exclusive" OR gate only makes D = 1 when either A = 1 OR B = 1 but not when A = B= 1. The exclusive OR operation is used so frequently that it is given the symbol +. Thus,  $D = A\overline{B} + \overline{AB} = A \oplus B$ .

Examples have been given of basic logical operations realized by means of relay contacts but this technique can become unwieldly. A more general diagrammatic representation of logic gates is desirable as the logic diagram should be independent of the circuit techniques employed in their realization. Unfortunately, there is no universally accepted symbol<sup>\*</sup> to represent a particular logic gate, some of the different types of symbols that have been used being shown in Fig. 3.

While the operation indicated by a logic gate symbol is independent of the circuitry used, it should be realized that as there are two allowed states the user must decide which state is to represent the logical 1 condition. For example, if the two states are represented by voltage levels, one may be positive and the other 0 V, one may be negative and the other 0 V, one may be positive and the other negative, both may be positive or both negative. Irrespective of the values of these voltage levels, the system is said to use positive logic if the logical 1 state is represented by the more positive level and is said to use negative logic if the logical 1 state is represented by the more negative voltage level

Although all the combinational logic gates appearing in Table 6 are available in various forms of hardware, it is possible to build complete logic systems with either only NOR gates or only NAND gates. Fig. 4 shows how the AND, OR, NOR and exclusive-OR operations may be realized using only NAND gates and Fig. 5 shows the sole use of NOR gates to

\*Following a majority decision of the I.E.C., the B.S.I. have opted for the rectangular logic gate symbols (not shown in Fig. 3). BS3939 section 21 is currently being amended. --- Ed.



Fig. 3. Some of the symbols used for logic gates.

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TABLE 6. Truth tables for common combinational logic operations.

INP	UTS			OUTPUT D =		
A	B	A.B	A+B	A+B	A.B	A B
0	0	0	0	1	1	0
1	Ō	Ö	1	0	1	1
Ó	1	Ō	1	Ō	1	1
1	1	1	1	0	0	0
NAME	OF GATE	AND	OR	NOR	NAND	EXCLUSIVE

realize the AND, OR, NAND and exclusive-OR operations. These illustrations also show the application of some of the relations given in Table 1. Figs 4(a) & 4(b) use relations 28 & 30 respectively on the output function and relation 30 is also used on the output from the threeinput NAND gate in Fig. 4(c). In Fig. 4(d), relations 27, 21 & 11 are used in turn on both inputs to the final gate and relation 30 used on its output function. Figs 5(a) & 5(b) use relations 29 & 28 respectively on the output function, relation 29 also being used on the output of the three-input NOR gate in Fig. 5(c). In Fig. 5(d) relation 29 is used on the input to the final gate and relations 27, 26, 21 & 11 used in turn on its output function.

These examples show that more gates of a given type are required to realize any other particular simple logic function. Although this point has been illustrated by simple Boolean expressions, in the design of more complicated systems the algebra may be cumbersome and other techniques such as Karnaugh mapping







Fig. 4. Logic operations of AND (a), OR (b), NOR (c) and exclusive OR (d), can be realized using only NAND gates. would be used to obtain a minimal solution. To synthesize a complex system it may be advisable to use gates of one type because of their availability and cost.

Many different types of solid-state electronic logic-gate realizations are available such as resistor-transistor logic (r.t.l.), diode-transistor logic (d.t.l.), directcoupled transistor logic (d.c.t.l.), transistor-transistor logic (t.t.l.), emitter-(e.c.l.) and complecoupled logic mentary metal oxide transistor logic (c.m.o.s.). These families of gates have different characteristics and one family may prove to be more suitable than another in a particular application. For example, the prime consideration may be highest possible speed of operation or lowest power consumption or greatest immunity to external noise or the simplicity of interfacing the gates with other circuitry. The successful design of a digital system therefore requires a working knowledge of the capabilities of the various types of electronic gates available.



Fig. 5. NOR gates can realize the logic operations of AND (a), OR (b), NAND (c) and exclusive OR (d).

### **Resistor-transistor and direct-coupled gates**



#### Simple r.t.l.

The simplest resistor-transistor logic (r.t.l.) gate, which performs the positive logic NOR function, is shown left. A positive voltage applied to any input turns Tr1 on, causing Vout to fall from  $V_{cc}$  to a value that depends on the base drive. If sufficient base drive is applied.  $Tr_1$  saturates making  $V_{out} =$  $V_{CEsat}$ , representing the logical 0 state. Positive voltages applied to the other inputs increases the degree of saturation and only change Vout by a small amount. If logical 0 voltages (V<sub>CEsat</sub>) are supplied to all inputs the baseemitter juction of Tr<sub>1</sub> will be only slightly forward biased  $(V_{\text{CEsat}} \approx 0.1 \text{ to } 0.4\text{V})$  and  $V_{\rm out} \approx + V_{\rm CC}$ . For useful logic functions the gate must feed some load, causing an additional current IL to flow in Rc and hence reducing the logical 1 value of  $V_{out}$  below  $V_{CC}$ . The gate is also a negative-logic NAND gate.

#### Improved r.t.l.

+V

Inclusion of a base bias resistor,  $\mathbf{R}_{\mathbf{K}}$  in the middle circuit.

returned to a negative supply ensures that Tr<sub>1</sub> is definitely turned off when all inputs are below the input logical 1 threshold and reduces the transistors turn-off time. Speed-up capacitors can be placed in parallel with each input R<sub>B</sub>to produce resistor-capacitortransistor logic. However, if all inputs are at logical 1 voltages and one of them rapidly switches to the 0-state, its speed-up capacitor couples the negativegoing transition to the baseemitter junction of Tr<sub>1</sub> which can cause the transistor to temporarily switch off. For this reason r.t.l. gates are normally only used at fairly low switching speeds. A clamping diode  $D_1$ , shown

right, can be connected to a supply  $+V_{\rm D} < +V_{\rm CC}$  to make the logical 1 output voltage less dependent on the load current, provided that the drop across Rc does not cause  $D_1$  to become reverse-biased.

#### **Direct-coupled** logic Direct-coupled-transistor logic

is also referred to as direct-

coupled logic and collectorcoupled-transistor logic, but it is strictly incorrect to refer to it as resistor-transistor logic as is done by some manufacturers since the input resistors have no logic function. The base resistors bottom serve only to divide the current between transistors when their inputs are paralleled. The gate is a positivelogic NOR gate which uses the transistorsas summing elements. The transistors also provide input- output isolation and restoration of the logic levels in each gate in a cascade. A positive voltage at any input turns on its associated transistor causing Vout to fall to  $V_{CEsat}$ , the logical 0 level. With all inputs at logical 0 the transistors are virtually cut off causing V<sub>out</sub> to rise towards +Vuntil the transistors in the following gates turn on. For correct operation R<sub>C</sub> and R<sub>B</sub> values must be chosen to ensure that driven transistors turn on when the driving gate transistors turn off. Also, when the driving gate is on the driven transistors must be off, hence the threshold

Typical r.t.l. parameters +Vccmin 20V  $+V_{CCmax}$  28V (4mA)  $-V_{BB}$  OV (with silicon transistors)  $R_B 82k\Omega$ ,  $R_K 30k\Omega$ ,  $R_C 7.5k\Omega$ logical 0 level 300mV max logical I level 14V min Fan-in 4 Fan-out 6 Maximum frequency 10kHz

value of VBE must exceed V<sub>CEsat</sub>. The difference between these two values influences the gate's noise immunity. Discretecircuit versions allow individual trimming of the base resistors to compensate for unequal  $V_{BE}$ values. Integrated circuit versions have closely-matched VBE and VCEsat values due to simultaneous manufacture on the same substrate. Fan-in capability is limited by collector leakage currents which, for several transistors off simultaneously, could cause Vout to fall below the level required to ensure that the following transistors are turned on. This is particularly so in low-power versions of the gate. A typical transfer characteristic is shown below.

#### Further reading

Dokter, F., & Steinhauer, J. **Digital Electronics, chapters** 4 & 5, Macmillan, 1973. Harris, J. N. et al Digital Transistor Circuits, chapters 6 & 7, Wiley, 1966.



Rc	$640\Omega$	$3600\Omega$	
R <sub>B</sub>	450Ω	1500Ω	
Fan-out	5	5	
Gate dissipation	12mW	2.5mW	
Propagation delay	24ns	45ns	
logical I level	1.2V	1.2V	
logical 0 level	200mV	200mV	
Noise margin, min ("1")	400mV	400mV	
Noise margin, min ("0")	350mV	350mV	





# Set 11: Basic Logic Gates-2



Fig. 1 shows a diode-transistor logic (d.t.l.) NAND gate, using discrete components, which is effectively a diode-logic AND gate followed by an inverting transistor. Resistors  $R_A$ ,  $R_B$  and Rc act as a level-shifting potential divider designed to provide enough base drive to allow Tr<sub>1</sub> to saturate, making  $V_{\text{out}} = V_{\text{CEsat}}$  (logical 0), when all input diodes conduct due to logical 1 levels being present at all inputs. If any input is at logical 0 V<sub>CEsat</sub>, its associated diode conducts, causing  $V_1$  to fall to the forward voltage of the diode. The transistor is then held in the cut-off state by the reverse bias obtained by potential division of VBB between  $R_K$  and  $R_B$  and the output goes to the 1-level as its collector rises towards  $+V_{\rm CC}$ . The turn-off of  $Tr_1$  is assisted by the negative base voltage and the turn-on time may be reduced by shunting R<sub>B</sub> with a speed-up capacitor. The fan-in is that of a diode logic gate and the fan-out depends on the current-sinking ability of Tr<sub>1</sub>. Preservation of logic levels may be improved by including a collector clamp diode-see card 1.

Another version of the d.t.l. NAND gate, sometimes called low-level logic, is shown in Fig. 2 where  $R_B$  and its speed-up capacitor are replaced by the input-offset diodes  $D_4$  and  $D_5$ , which are more suitable for monolithic integrated fabrication techniques. Only a relatively small voltage swing is required at the base of Tr<sub>1</sub> to switch it on or off, but in Fig. 1 a relatively large swing in  $V_1$  is required to achieve this due to the large part of  $V_1$  lost across  $R_B$ . The use of  $D_4$  and  $D_5$  in Fig. 2 leads to a much smaller required swing in  $V_1$  to achieve the desired base voltage swing. Hence the signal levels may be lowered to reduce gate dissipation which also falls due to the removal of  $R_{\rm B}$ . Other diodes may be placed in series with  $D_4$  and  $D_5$  to improve noise immunity. While the input diodes should have a very short reverse recovery time, the levelshifting diodes  $D_4$  and  $D_5$ should be slow recovery types to ensure that they do not return to their high-impedance, reversebiased state until  $Tr_1$  has cut off. Elimination of the V<sub>BB</sub> supply can simplify circuitry in many instances, a popular modified form of the d.t.l. NAND gate using a single supply being shown in Fig. 3. In comparison with Fig. 2, the offset diode  $D_4$ is replaced by  $Tr_2$  and  $R_D$ . This transistor provides amplification



Parameter	Fig. 2	Fig. 3	Fig. 4	Fig. 5
+Vcc	4V	5V	15V	15V
$-V_{BB}$	2V			
RA	$2k\Omega$	$1.6 k\Omega$	3kΩ	3kΩ
Rc	$2k\Omega$	6kΩ	$15k\Omega$	$15 k\Omega$
R <sub>D</sub>		$2.15 k\Omega$	$12k\Omega$	$12k\Omega$
RE			_	$1.5 \mathrm{k}\Omega$
Rĸ	$20k\Omega$	5kΩ	$5 k\Omega$	5kΩ
Fan-out	5	8	10	10
Gate dissipation	10mW	10mW	28mW	28mW
Propagation delay	30ns	30ns	125ns	110ns
Noise margin ("1")	0.4V mir	n 0.4V min	5V	5V
Noise margin ("0")	0.35V mi	n 0.35V mii	n 5V	5V

that allows a higher level of base drive to be fed to  $Tr_1$ , achieving a higher fan-out capability, and also permits a reduction in the value of  $R_K$  compared with Fig. 2.

Gates used in industrial logic systems often require high noise immunity, rather than high speed and low power dissipation, as large transients can be produced in supply lines or picked up at inputs due to switching of relays, etc. Fig. 4 shows a modified form of Fig. 3 that exhibits higher noise margins largely due to  $D_5$  being changed from a forward-biased diode to a reverse-biased diode exhibiting a zener-type characteristic when the p.d. across it reaches about 6.7V. Thus the input threshold

amount equivalent to the p.d. that would occur across a further four forward-biased diodes connected in series with  $D_5$  in Fig. 3, but is achieved by using only one such diode operating on its reverse characteristic. A higher supply voltage is required in Fig. 4 but to prevent large increases in currents, and hence gate dissipation, all resistor values are also increased. Fig. 6 shows typical transfer characteristics for the circuits of Figs. 3 & 4. Fig. 5 shows the high noiseimmunity gate of Fig. 4 with an active pull-up transistor Tr<sub>3</sub>. When  $Tr_1$  is off  $R_C$  supplies base drive to Tr<sub>3</sub> which supplies load current via R<sub>E</sub>. With the output in the 0-state Tr<sub>3</sub> is off and Tr<sub>1</sub> sinks load current through D<sub>6</sub> which causes the low logic level to exceed VCEsat of Tr1. The table shows a comparison of some typical parameters for integrated circuit versions of Figs. 2 to 5.

voltage is increased by an

#### Further reading

Dokter, F. & Steinhauer, J. Digital Electronics, chapters 4, 5 and 6, Macmillan, 1973. Meindl, J. D. Micropower Circuits, chapter 11, Wiley, 1969.



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# Set 11: Basic Logic Gates—3

### Basic t.t.l. gate



#### Typical parameters

Construction and a second s

Temperature range: 0 to  $70^{\circ}$ C +V min 4.5V, +V max 5.5V "0" supply current 22mA max "1" supply current 8mA max Fan out 10 t.t.l. loads Inputs: "0" level 800mV max "1" level 2V min "0" level 1.6mA max at +V max "1" level 40 $\mu$ A max with  $V_{IN}$ =2.4V

#### Outputs: "0" level 400mV max at +V min "1" level 2.4V min at +V min "0" level 16mA Short-circuit output current: 18 to 55mA at +V max Propagation delay\* from "1" to "0" 15ns max Propagation delay\* from "0" to "1" 22ns max "1" noise margin 400mV min "0" noise margin 350mV min Gate dissipation 10mW

\*With output loaded by  $400\Omega//15$  pF.

#### **Circuit description**

Circuit shows the form of the basic transistor-transistor logic gate which performs the positive logic NAND function and which may normally have up to eight inputs. If all the inputs are at a high level (logical 1), base drive is provided to Tr<sub>2</sub> through R1 and the base-collector junction of  $Tr_1$ . If any one or more input is at a low level (logical 0), the current in  $R_1$ flows through the base-emitter junction of Tr<sub>1</sub> to ground. The base will then be only  $V_{\rm BE1}$ above  $V_{1N}$  and  $Tr_2$  cut off due to lack of base drive. Transistor Tr<sub>1</sub> thus performs the AND function as its collector is only high if all its inputs are high. Transistor Tr<sub>2</sub> acts as a phase splitter that saturates with only a moderate current gain-note the small ratio of  $R_1/R_2$  with  $R_2 \approx R_3$ . When  $\text{Tr}_2$  is cut off its collector and emitter are approximately at +V and 0Vrespectively. When  $Tr_1$  drives  $Tr_2$  on, its emitter rises to  $V_{BE4}$ and its collector falls to  $(V_{BE_4} + V_{CE_{2}sat})$ . In this state Tr<sub>4</sub> will be saturated so that the output will be at VCE4sat (logical 0) when all inputs are in

the high state. In this condition the gate can sink current through Tr₄ from a number of loads, normally a maximum of 10, in the 0-state, without causing  $V_{CE4sat}$  to rise above the acceptable 0-threshold. If any of the gate inputs is in the low state,  $Tr_4$  will be off as  $Tr_2$  is cut off. Transistor Tr<sub>a</sub> will be on to an extent determined by the emitter current demanded at the output. This current will be small when the gate feeds a number of similar t.t.l. gates and its base current will be smaller still. Hence the p.d. across R<sub>2</sub> due to Tr<sub>3</sub> base drive will be negligible and the output will be in the high state with  $V_{out}$  at approximately +V- $(V_{D_1} + V_{BE_3}).$ 

#### Switching action

When switching the output from the 0- to the 1-state, all inputs are initially high (logical 1). As the potential of one or more input falls, nothing happens until it reaches about 1.4V, when the source current via  $Tr_1$  base-emitter junction prevents base current flowing to

Tr<sub>2</sub> via the base-collector junction of Tr<sub>1</sub>, which rapidly removes the stored charge from Tr, base and switches it off. The collector potential of Tr<sub>2</sub> starts to rise as this transistor turns off, but stops rising as Tr<sub>3</sub> begins to conduct heavily. This conduction occurs because Tr<sub>4</sub> has not yet switched off, as the charge stored in its base decays only relatively slowly through  $R_3$ . Therefore, a large current spike of short duration and limited in amplitude by R<sub>4</sub> occurs in the supply line during the switch-off action due to  $Tr_3$ and  $Tr_4$  being simultaneously on. This conduction in Tr<sub>4</sub> removes some of the stored base charge, allowing the output voltage and Tr<sub>2</sub> collector potential to rise. The rise continues until Tr<sub>3</sub> becomes cut off and the output settles to the 1-level. Typical input, output and transfer characteristics are shown in Figs. 2, 3 & 4 respectively. Width and amplitude of the current spike are virtually unaffected by the rate at which the gate is switched on and off. Hence a side effect of the current spike is an increase in power consumption as the switching frequency increases.

#### Unused inputs

Inputs that are unused in a particular application should be connected in parallel with used inputs for fastest switching speed. Unused inputs can be left open circuit, but excessive pick-up noise may result unless the open circuit is made at the integrated circuit package connection. If unused inputs are connected to the positive supply rail, it is advisable to do so via a resistor of around  $1k\Omega$  to prevent the gate being damaged by a supply line transient that exceeds the maximum rating.

Further reading Scarlett, J. A. Transistor-Transistor Logic and its Interconnections, chapters 1 to 6, Van Nostrand, 1972.



### NAND gate variations in t.t.l.



In the basic, or standard t.t.l. NAND gate (card 3), the resistance values affect performance. Resistor R<sub>1</sub> influences the rate of rise of voltage at Tr<sub>2</sub> base and turn-on time. Gate dissipation, when the output is in the 0-state is affected by the value of  $R_2$ . Stored base charge in  $Tr_4$  is removed via  $R_3$ when the output state changes from logical 0 to 1. Turn-off time of the gate when feeding a capacitive load is influenced by the value of  $R_4$  which provides short-circuit protection.

#### Low-power t.t.l.

For low-power operation the resistor values must be increased to reduce the chargingdischarging currents. But larger resistors imply slower switching speeds unless transistor size is reduced to lower their capacitances. This can be done due to the lower current levels and by reducing the degree of gold doping the transistors achieve higher current gains to better-utilize the smaller currents. The resulting gate, top left, has a power dissipation only one-tenth of that in a standard gate with a speed reduction penalty of only three times.

#### High-speed t.t.l.

To obtain higher switching speeds than are obtainable with standard t.t.l., the chargedischarge rates of the integrated and external capacitances must be increased. This implies larger transistor currents and hence lower resistor values. The higher currents require larger transistors having increased capacitances that tend to offset the speed increase due to higher currents. A distinct speed improvement is obtainable, the high-speed NAND gate shown above having about double the speed and slightly more than twice the dissipation of the standard gate.

The Darlington-connected pull-up transistors Tr<sub>5</sub> and Tr<sub>3</sub> provide higher active-region gain which reduces the output resistance and increases the ability to drive capacitive loads. Resistor R<sub>5</sub> is sometimes returned to the output point, rather than to the 0-V rail, to reduce the gate dissipation. Sometimes a by-pass transistor,  $Tr_6$ , is added to the pull-down transistor, Tr<sub>4</sub>, as shown right. Resistance of the discharge path for stored base charge in Tr<sub>4</sub> is reduced, improving the turn-off time. Transistor Tr<sub>2</sub> cannot conduct through R<sub>3</sub> until its emitter voltage exceeds the turn-on  $V_{BE}$  of  $Tr_6$  which is approximately the same as that of  $Tr_4$ . Hence, the output remains in the 1 state until VIN rises to a level sufficient to turn on Tr<sub>4</sub>, which removes the lowslope region from the transfer characteristic improving noise immunity.

#### Schottky-clamped t.t.l.

Excess base drive is shunted through the diode, which clamps the collector-base junction with a p.d. of 400mV which is insufficient to produce any significant forward conduction. The elimination of gold-doping provides highergain, physically-smaller transistors with very little charge storage and hence much higher switching speeds without the penalty of further increased power dissipation. Use of Schottky-clamped transistors increases the output 1-level improving its noise immunity. Transistor  $Tr_4$  below has very little stored base charge, improving the turn-off time and reducing the supply current spike. As this transistor's V<sub>CEon</sub> determines the output 0-level, the level will be raised by about 100mV compared with the standard gate but its value will be far less temperature dependent. The table shows a comparison of some typical parameters of different t.t.l. NAND gates.

#### Input clamping diodes

Most t.t.l. gates have input clamping diodes to ground, as shown on right to reduce the negative excursions of input signals due to ringing caused by reflection of pulses along the interconnection transmission lines.

#### Further reading

Priel, U. Take a look inside the t.t.l. i.c. *Electron*, pp. 24, 26 & 30, 19 April 1973. Murphy, R. H. Performance and reliability aspects of current trends in t.t.l., *New Electronics*, pp. 30, 33 & 34, 20 April 1971. Clifford, C. Guide to low-power t.t.l., *New Electronics*, pp. 24, 27 & 28, 4 May 1971. Scarlett, J. A. Transistor-Transistor Logic and its Interconnections, chapters 3, 4 & 9, Van Nostrand, 1972.

#### Cross references

Set 11, card 3; set 10, card 11.

Parameter	standard	low- power	high- speed	schottky
+V	5V	5V	5V	5V
"0" current	5.5mA	0.46mA	10mA	9mA
"1" current	1.3mA	0.18mA	4.2mA	4.25 mA
Fan out	10	10	10	10
Gate dissipation	10mW	1mW	22mW	20mW
Output delay "1" to "0"	8ns	30ns	6ns	3ns
Output delay "0" to "1"	12ns	30ns	6ns	3ns
Noise margin "1"	400mV	400mV	400mV	700mV
Noise margin "0"	400mV	400mV	400mV	300mV
VINmax "0"	800mV	700mV	800mV	800mV
V <sub>INmin</sub> "1"	2V	2V	2V	2V
Vout "0"	400mV	300mV	400mV	500mV
Vourmin "1"	2.4V	2.4V	2.4V	2.7V
IIN "O"	1.6mA	0.18mA	2.0mA	2.0mA
I <sub>IN</sub> "1"	40µA	10µA	50µA	100µA
IOUT "0"	16mA	2mA	20mA	20mA



# Set 11: Basic Logic Gates-5

Vop 1

5 0.0

0.

0

### Complementary m.o.s. gates—1



Typical data IC  $\frac{1}{3}$  (CD4007AE) Working voltage range (V<sub>DD</sub>  $-V_{SS}$ ) 3 to 15V Temperature range -40 to  $+85^{\circ}$ C Input capacitance 5pF Input resistance > 10<sup>9</sup>  $\Omega$ Output voltage (high) 9.99V (V<sub>DD</sub> = 10V, V<sub>SS</sub> = 0V) Output voltage (low) 0.01V (V<sub>DD</sub> = 10V, V<sub>SS</sub> = 0V) Fan-out d.c. > 1000 a.c. typically 20



#### Construction

A cross-section depicting the formation of n-channel and p-channel transistors on the same chip is shown below, with associated symbols. The p-channel one is formed directly on an n-type substrate, but the n-channel device is formed in a p-region diffused into the substrate. This process creates parasitic diodes, and their relationship to the inverter terminals is shown centre. As  $V_{DD}$  is normally more positive than  $V_{SS}$ , these diodes are in a reverse-biased state, and their leakage current contributes to quiescent power dissipation. It should be noted that if the voltage level at the output terminal is subjected to any transient condition, it is unable to go more positive than  $V_{DD}$  or more negative than Vss, by more than the forward conducting voltage of these parasitic diodes.

#### Input protection

Because the input resistance of the device is so high, static charges may be sufficient to charge the input capacitance of the gate oxide to a high enough voltage to cause breakdown  $(\sim 100V)$ . The diode protection network, below right is one type designed into some gates. If the gate terminal voltage is greater than  $V_{DD}$ , diodes  $D_1$  and  $D_2$  can conduct, and if less than Vss, D<sub>3</sub> may conduct-the current magnitude should be limited to around 10mA (R may be around  $2k\Omega$ ). For conditions where the diodes are either forward or reverse biased, the voltage across the oxide layer is limited to approximately 1 or 25V respectively.

V<sub>DD</sub> in the ange 3V – 15V

0.6

Vin (volts)



The circuit shows the basic complementary-symmetry isolated-gate inverter stage, which uses both an n-channel and a p-channel enhancementmode m.o.s.f.e.t. in a series-pair configuration. Such circuits can be directly coupled as either transistor will be in its nonconducting or off-state if its gate-source voltage is zero. Individual gates are tied together to form a single signal input gate, and the drains are commoned at the output. Assume that the input signal excursion is from  $+V_{DD}$  to ground potential i.e.  $V_{\rm SS} = 0V$ . When the input is  $+V_{DD}$ , the n-channel f.e.t. is biased to a high conducting state because V<sub>Gs</sub> is a high positive value. Simultaneously, the effective gate-source voltage of the p-channel f.e.t. is zero, and hence this transistor will be off, and the output will be at ground potential. When the input goes to zero volts (the low or 0-state for positive logic), the n-channel f.e.t. is biased off, but the p-channel transistor has now a large negative voltage between gate and source and is therefore biased into conduction, and the level then approaches  $+V_{DD}$ (the high or 1-state). In either logic state, one transistor is conducting and the other is cut-off. It follows that the quiescent power dissipation is exceedingly low—the transistor that is off will only conduct leakage current, typically 1nA.

More significant power dissipation occurs during the switching from one level to the other, due to both a current spike which occurs when the inverter is in its linear region and to the charging and discharging of load capacitance. This depends on the frequency, the value of the capacitance and the square of the supply voltage.

If the p-channel source is connected to ground, the n-channel source should be connected to  $-V_{\rm DD}$  and the



# Set 11: Basic Logic Gates-6

### Complementary m.o.s. gates-2



#### Typical data

 $V_{\rm DD} = +10V, V_{\rm SS} = 0V, T_{\rm amb} 25^{\circ} {\rm C}.$ 

Gate p	drive (source)	) V <sub>out</sub>	drive n(sink)	Vout	quiescent power	t delay
	(mA)	(V)	(mA)	(V)	(µW)	(ns)
Nor	1.0	9.5	2.5	0.5	0.05	25
Nand	1.2	9.5	0.6	0.5	0.05	25
Inverter-pair	2.5	9.5	2.5	0.5	0.05	20
Buffer	2.5	9.5	16.0	0.5	0.5	10 to 25

#### **Basic gate structure**

NOR and NAND functions are formed by series and parallel combinations of p and n pairs. For the NOR gate, the n-type f.e.ts are in parallel and the p-type in series as shown left. The circuit configuration of the NAND is similar but the p-types are in parallel, the n-types are in series, and the supply connections are changed over, see diagram right. The NOR logic action is described assuming  $V_{DD}$  is a positive voltage and Vss is at 0V. Input excursions at A and B will be within the range 0V to  $V_{DD}$ . If either A or B is positive, then one of the p-types will be off and one of the n-types on, thus connecting Vout to 0V via the on-resistance of the conducting transistor. If both A and B are positive, again the output will be at 0V. If A and B are at 0V, then both p-types will be biased on, due to the negative voltage at their gates, and both n-types will be off, and hence Vout will be at  $+V_{DD}$ .

#### General notes

Noise immunity. Typically the input may change by up to  $0.45 V_{DD}$  before the output begins to alter. Over the full range of  $V_{DD} - V_{SS}$  (3 to 15V), a noise immunity of  $0.3 V_{DD}$  is guaranteed.

Unused inputs. NOR gates: Connect input terminals together or to the lower voltage terminal Vss. NAND gates: Connect input terminals together or to the voltage terminal V<sub>DD</sub>.

**Parallel gates.** Gate outputs may be connected in parallel allowing greater output currents at the expense of increased power dissipation—current hogging need not be considered.

**Pulse drive.** Rise and fall times should be less than  $5\mu$ s typically to prevent the device spending too long in the linear region during switching and thus increasing power dissipation.

Output characteristics The two graphs shown left illustrate typical n-device and p-device drain current characteristics for the NAND and NOR gate. Drain currents for the n-type in the NOR gate are much higher than those available in the NAND gate for the same gate-source voltage.

#### **Propagation delay**

Delay periods are usually defined between 50% points on the input and output level transitions, and this will depend to a great extent on the capacitive loading at the output, as this itself affects the transition time. Third graphs show that typical propagation delays depend on supply voltage, though in the 10 to 15V region the delay spread is of the order of 10ns.

As the capacitive loading is increased (each c.m.o.s. gate is an effective 5pF load), it is fairly easy to slow down.the circuits with external capacitance (right). The propagation delay is also temperature-dependent, increasing as the temperature increases due to fall-off in the  $g_m$  of the transistors.

#### Development

Devices have been produced which operate with  $V_{\rm DD}$  values ( $V_{\rm SS}$ =0V) in the range 1.1 to 1.6V, and recently 0.7V. Another technology known as dielectric isolation, applicable to both types of c.m.o.s., promises devices with propagation delays as low as 10ns.

#### Further reading

Harrison, L., CMOS—Tolerant logic, *Electron*, 3 May 1973. Ankrum, P. D., Semiconductor Electronics, Prentice-Hall. Bishop, R. A. Complementary m.o.s. offers many advantages to the digital-systems designer, *Electronic Engineering*, Nov. 1972, pp. 67–70. Funk, R. E. & Bishop, A. C.o.s.m.o.s. simplifies equipment design, *New Electronics*, 1 May 1973, pp. 24–8.



### **Emitter-coupled logic**



#### **Circuit description**

"Emitter-coupled logic" (e.c.l.) describes integrated logic circuits in which the switching transistors do not saturate as in other forms of bipolar transistor logic. Delays due to charge-storage effects in the saturated mode are avoided, leading to faster switching. The above diagram is one form of a three-input basic e.c.l. gate, whose outputs provide an OR and the complementary NOR logic functions. The reference voltage must be well regulated, and of a level midway between the logic swings. For nominal logic high and low output of -0.75V and -1.55Vrespectively, this indicates  $V_{\rm ref} = -1.15$ V. The following evaluation is for guidance only. If inputs A,B,C, are at the low logic level, Tr<sub>1</sub>, Tr<sub>2</sub> and Tr<sub>3</sub> are cut off,  $Tr_4$  is conducting and hence the potential at Q with respect to ground, assuming 0.75V drop across all emitterbase junctions, will be -1.9V, and the voltage across  $R_3$  is  $V_{\rm EE} - V_{\rm Q}$ . Assume  $V_{\rm EE}$  is -5V, then I=2.6mA. This means that the drop across  $R_2$  is 0.78V and therefore the OR output will be -1.53V. As there is no current through  $R_1$ ,  $V_p$  is 0V, and the

NOR output is thus -0.75V due to the base-emitter junction drop.

When a logical 1 (-0.75V) is applied to, say, terminal A transistor Tr<sub>1</sub> will conduct harder than Tr<sub>4</sub>, and the current in  $R_3$  is then supplied via  $R_1$ , i.e. the current has been diverted from  $Tr_4$  because the input voltage to  $Tr_1$  is half a logic level more than that of Vref, and the resultant reduction of the baseemitter drops of Tr<sub>4</sub> is sufficient to decrease its emitter's current to almost zero. Hence the base of  $Tr_5$  is now at 0V and the OR output will be logical 1 at --0.75V. The related NOR is determined from the new V<sub>Q</sub> value of -1.5V, and hence the p.d. across  $R_3$  is -3.5V and thus I=2.9mA. Therefore the p.d. across  $R_1$  will be -0.78V and hence the NOR output is -1.53V.

• A temperature-compensated regulator package of the form shown left is available as the reference voltage, also frequently on the same chip as the gate structure. This minimizes variation of noise margin with temperature.

#### **Faster circuits**

• Centre circuit shows a type

Typical operating data  $R_1 270\Omega$ ,  $R_2 300\Omega$   $R_3 1.2k\Omega R_5$ ,  $R_6 1.5 to 2k\Omega$   $V_{CC}$  0V,  $V_{ret} - 1.15V$ ,  $V_{EE} - 5V$ Logic 1 output - 0.75V Logic 0 output - 1.5V Propagation delay 5 to 11ns Max a.c. fan-out 15 Output resistance < 0.1 $\Omega$ Output current 1.5 to 2mA to maintain logic

0 to 75°C. Gate dissipation ≈ 35mW Worst noise margin 250mV

levels within 3% Temperature range

of e.c.l. gate where the reference is at ground potential. With the supplies shown, logic levels nominally -400mV for logical 1. O and +400mV for logical 1. Suitable for driving into 50  $\Omega$ loads, and up to 25mA d.c. when terminated in 50- and 270- $\Omega$  pull-down resistor to -3.2V.

Fan-out: 12 (a.c.) Propagation delay 2 to 3ns Input level (high) 0.15 to 0.72V Input level (low) -1.5 to -0.15V

Unused inputs: connect to  $-1V \pm 50\%$ .

Noise margin:  $\pm 200$  mV.

• Other e.c.l. gates with a basic configuration similar to the first provide multi-input, multiemitter follower OR/NOR outputs, with optional pull-down resistors.

 $V_{\rm CC}=0V, V_{\rm EE}=-5.2V\pm20\%$ Output (source) current: up to 2.5mA.

Operating temperature range -55 to  $125^{\circ}C$ 

Propagations delay (rising) 3.5 to 5ns

Fall-time propagation delays up to 15ns due to emitter-follower



Set 11: Basic Logic Gates-7

output resistance and capacitance loading. Three outputs tied together (wired OR) gives output impedance of about  $2.5\Omega$ . Suitable for driving  $50-\Omega$  loads (two pull-down resistors only for faster fall times). Noise margin 175mV To maintain high speed, limit interconnection length to <25cm Fan-in: 20; fan-out: 15 (a.c.)

Tail-in: 20, fail-out: 15 (a.c.) Basic form of ECL circuit capable of propagation delay < 1ns is shown right. A separate supply terminal  $V_x$  is used for the output emitter-followers.  $V_{EE}$ : -5.2V. Pull-down resistors of 50 or 2k  $\Omega$  provide a path for leakage currents (unused inputs can be opencircuit) and act as loads for driving gates. Power dissipation  $\approx$  55mW.

Fan-out=70 for  $R_{pd}=50k\Omega$ . Fan-out=7 for  $R_{pd}=2k\Omega$ . Propagation delay:

0.9ns for 510- $\Omega$  load. 1.1ns for 50- $\Omega$  load.

Interconnections should be  $50-\Omega$  microstrip transmission lines and termination connected to -2V supply. Temperature range 0 to 75°C. Logic swing typically -0.9V ("1") to -1.75V ("0").





#### d.t.l./t.t.l.-c.m.o.s.

The minimum t.t.l. 2.4V 1-level output is normally for a load current of 400 $\mu$ A, but as the c.m.o.s. gate input current is approximately 10pA, the more likely 1-output is 3.6V. This is an inadequate noise margin (0.1V) and an active pull-up resistor typically 1 to 10k  $\Omega$  depending on whether high speed or low power is required) is connected from the t.t.l. output to the positive supply rail of +5V. Hence when  $Tr_1$  is off the c.m.o.s. input will be at +5Vgiving a 1.4V noise margin. The threshold values of switching for the c.m.o.s. gate is typically 30% and 70% of the supply voltage, i.e. 1.5V and 3.5V respectively. Note. Unshaded areas represent the 1- and 0-regions, the borders being the minimum 1-level and the maximum 0-level

l-level and the maximum 0-level for minimum and maximum t.t.l. supply voltages.





**c.m.o.s.-h.t.l.-c.m.o.s.** Most high-threshold logic gates operate from a V<sub>CC</sub> supply of  $15 \pm 1V$ . Hence direct connection with c.m.o.s. gates is possible. Noise immunity is high, of the order of 3V for high and low h.t.l. outputs, though this may be improved using a pull-up resistor as shown (in some i.cs this may be internal) which also improves the output rise-time. Rise and fall times of around  $1\mu$ s should be the aim for the h.t.l.-c.m.o.s. interface.

logic '1

-o∙aaV

-1-105V





noise margin of 225mV, and a

logic O

-5·2V

-3·60V

logic 1

Vero

-1·56V

low level of 4.3V.

ransist output

CMOS

logic 'O'

-5-8V

1-475V

utput



The c.m.o.s. gate must sink 1.6mA and source  $40\mu$ A for the 0- and 1-state of the bipolar input respectively. Not all c.m.o.s. devices can cope with one t.t.l. load (1.6mA) but gates on the same package may be paralleled to increase their current sinking capability, or preferably buffers such as CD4009, CD4041, CD4049 should be used. These devices can sink two t.t.l. load currents and still have an output of 0.4V, thus retaining a 0.4V noise margin.

### Low-power t.t.l.-c.m.o.s.

Most c.m.o.s. devices can drive low-power t.t.l. directly as the logic zero-level sink-current is 0.18mA. Again for driving c.m.o.s., the t.t.l. output should have a pull-up resistor for adequate noise margin.



c.m.o.s.-e.c.l.

Both may be operated from

restricted to 1MHz. Speed is increased if Vss is taken to a

separate supply between -5

 $-5V \pm 20\%$ , but speed is

Output swing of e.c.l. (typically -1.55 to -0.75V) is inadequate to drive c.m.o.s. directly, i.e. switching levels are 30% and 70% of -5.2V. One technique

Rpd

DISCRETE 12N2907 MP56516

ECL MC1024 /ss

CMOS CD4001 MC14001



is to use a two-input expandable gate driving a p-n-p transistor, with a pull-down resistor  $(\approx 3 \text{ }\Omega)$  to the negative supply. This provides noise margins amplitudes of 1.56 and 0.66V.

#### h.t.l.-t.t.l.

Circuit shows a technique for interfacing high level logic to t.t.l. This uses a linear voltage comparator LM311, the component values used allowing an input level range of 0 to 30V. Capacitor  $C_1$  may be added to decrease the effects of fast noise spikes.

# Set 11: Basic Logic Gates-9

### Threshold logic



#### **Circuit description**

Threshold logic gates are much more powerful and flexible than are the normal AND, OR gates. Majority, minority, AND, and OR gates are simply particular cases of threshold logic. A threshold gate has inputs A, B, C.... with weights a, b, c...associated with the respective inputs. The output Z from such a gate is then:

Z=1 if < aA+bB+cC+...>≥ some value  $T_1$ , the upper threshold, and Z=0 if

 $\langle aA+bB+cC+\ldots \rangle \leq$  some value  $T_2$ , the lower threshold.  $T_1 > T_2$  and normal arithmetic addition is involved in the above brackets. The output is more precisely written as:  $Z = \langle aA+bB+cC+\ldots \rangle_{T1:T2}$ 

 $T_1 - T_2$  is the threshold gap and is that inadmissible sum which will give an ambiguous output. Generally A, B, C, ... are binary (1 or 0); a, b, c... need not be, but generally are, integers in which case the weighed sum can only take on integral values. The threshold performance is then quoted by those to integers between which switching takes place. We then obtain:

 $Z = \langle aA + bB + cC + ... \rangle_{t_1, t_2}$ where  $t_1$  and  $t_2$  are integers,  $t_1 - t_2 = 1$ , and  $t_1 - t_2 > T_1 - T_2$ . The symbol used is shown (left).

Circuit shows a three-input threshold gate with identical weighting on each input. Basically it comprises three longtailed pairs with a constantcurrent source (e.g.,  $Tr_1$  and  $R_1$ ) in each tail. When A exceeds V<sub>ref1</sub> by 100mV or more the tail current flows through Tr<sub>2</sub> and R<sub>4</sub> and when A is less than Vref, by more than 100mV the current flows through Tr<sub>3</sub> and  $R_5$ . Resistors  $R_4$  and  $R_5$  act as summing resistors, summing the currents from the long-tailed pairs. Transistors Tr<sub>4</sub> and Tr<sub>5</sub> act as emitter-follower output stages for  $V_{C_1}$  and  $V_{C_2}$  so that  $Z = V_{C_2} - V_{be}$  and  $\overline{Z} = V_{C_1} - V_{be}$ . When Z is in the high state it must exceed V<sub>ref1</sub> by 100mV or more so that a succeeding stage will recognize it as logical 1. Likewise in the low state Z must be less than Vref by 100mV or more. The following formulae apply to

the circuit. •  $I_1 = \frac{V_{ref_2} - V_{BE}}{R_1}$ ;  $I_2$  and  $I_3$  are

obtainable similarly.

• When  $I_1$  is switched from  $R_5$  to  $R_4$  on application of logical 1 at A, then the change in  $V_{C_2}=I_1R_5$ . This change should be around 200mV or more to

#### Circuit data

Supply 6V.  $R_1$ ,  $R_2$ ,  $R_3$  1.5k $\Omega$ .  $R_4$ ,  $R_5$  560 $\Omega$ .  $R_6$ ,  $R_7$  3.3k $\Omega$ . V<sub>ref1</sub> 4.9V,  $A = B = C = V_{ref1} + 100$ mV.  $V_{ref2}$  1.8V. Sequentially applying A, B & C, Z changes in 0.4V steps from 3.9 to 5.1V. "1"=5.1V ( $V_{ref1} + 200$ mV), "0"=4.7V or less ( $V_{ref1} - 200$ mV).

 $V_{ref_1}$  can be reduced towards  $V_{ref_2}$  but cannot be increased much beyond 4.90V.  $R_4$  and  $R_5$  can be varied but are generally tied to the values for  $R_1$ ,  $R_2$  and  $R_3$  (ref. 1) and to the voltage swing required.

obtain decisive switching. • Max  $Z = V_{CC} - V_{BE}$  and occurs when no current flows in R<sub>5</sub>.

• Min  $Z = V_{CC} - V_{BE} - 3I_1R_5$ . This assumes that all the tail currents are identical and flowing in  $R_5$ . As shown, all three inputs must be applied before Z goes to

logical 1. Hence:  $Z = \langle A + B + C \rangle_{3:3} \equiv Z =$ A.B.C(Boolean). Clearly if any of the inputs is permanently tied to logical 1 we obtain a two-input AND gate. Moreover, if  $V_{ref1}$  is dropped to 4.5V only two of the inputs are required to be high for Z to be 1 and hence:  $Z = \langle A + B + C \rangle_{2:1} \equiv simple$ majority gate.

If now C (say) is permanently tied to logical 0 we require the two remaining inputs to be high and we have obtained a twoinput AND gate. On the other hand, if C is permanently tied to logical 1 only, one of the remaining inputs requires to be logical 1 for Z to be logical 1 and hence we have obtained a two-input OR gate. Alteration of  $R_4$  and  $R_5$  is more

generally used to alter the threshold<sup>1</sup>. If  $R_4$  and  $R_5$  are different then

the outputs from  $Tr_4$  and  $Tr_5$ 

will not be the logic complement of one another. Any other threshold logic function that one wants can be obtained within the restriction that the weightings will remain the same. Furthermore if, say,  $R_5$  is comprised of a string of series resistors then one can obtain a large number of different functions as well as the basic one<sup>2</sup>.

Reference 3 shows how one can improve the tolerance of the circuit to large input voltages which otherwise can cause saturation and incorrect current summation.

#### Further reading

Hurst, S. L. Introduction to threshold logic, *Radio and Electronic Engineer*, 1969, pp. 339-51.

 Hampel, D. & Winder, R. O. Threshold logic, *IEEE Spectrum*, May 1971, pp. 32–9.
 Hampel, D. Multifunction threshold gates, *IEEE Trans. on*

Computers, vol. C-22, Feb. 1973, pp. 197–203. 3. Hurst, S. L. Improvements in

circuit realisation of threshold logic gates, *Electronic Letters*, vol. 9, 1973, p. 123. See also card 12.



# Set 11: Basic Logic Gates—10

## **Optical logic**



Performance data  $R_1, R_2 100\Omega$   $R 10k\Omega$  and  $500\Omega$  V 6V  $V_{F_1}=0$  and  $V_{F_2}=0 \rightarrow 6V$  giving  $I_{F_1}=0$  and  $I_{F_2}=0 \rightarrow 50mA$ optocouplers TIL112 These figures resulted in graph (right)



#### **Circuit description**

All the simple logic functions can be performed using optical couplers. Fan-out or speed difficulties preclude the use of these gates in complete logic system but in systems where simple logic is required and/or the input is in optical form, they can be very useful and show the usual advantages of optical coupling (cross ref. 1). Circuit shows an OR gate using optical couplers. If  $V_{F_1}$  is large enough (>1.2V) to make D<sub>1</sub> conduct then  $Tr_1$  conducts and V is applied to R. Similarly V is applied to R if  $V_{F_2}$  is high and if both  $V_{\rm F1}$  and  $V_{\rm F2}$ are high. Hence, in Boolean terms,  $V_0 = V_{F_1} + V_{F_2}$ . The transfer characteristic shown right for two different values of R indicates the static performance and shows noise immunities superior to that of t.t.l. For these two values of R with the given V the phototransistors are being operated in their saturated mode (cross ref. 1) which permits a maximum current through each transistor of approximately 15mA. The normal parallel type of fan-out is, therefore, only one since the required IF of succeeding stages is in the range 10 to 50mA. However, serial connection of succeeding stages could yield a fan-out of two or three if the appropriate resistance were chosen. This fan out could be increased if V were increased or, if speed was not essential, by using optocouplers with Darlington output stages. The fan-in can easily be increased. Note that basically the drive signal is the diode current rather than the applied voltage so that current driving can easily be employed. Moreover  $V_{\rm F1}$  and  $V_{\rm F2}$  need not be the same, nor indeed do the two diode currents. All that is necessary is that each transistor

be driven into saturation. With the quoted data pulse repetition frequencies of 40kHz can be handled. Higher frequencies but with lower current handling capacity can be obtained using photo-diodes and lower frequencies with greater current handling capacity with photodarlingtons (cross ref. 2).

#### **Component changes**

V can be increased to 30V. V<sub>F1</sub> and R<sub>1</sub> can be varied so long as  $I_{F1}$  is in the range 10–50mA. Similarly for V<sub>F2</sub> and R<sub>2</sub> Optocouplers: ISO-LIT12, MCT26.

#### Modifications

A NOR gate can be constructed as shown left, in which the load R is placed in the positive supply rail.
An AND gate can be constructed as shown centre. In this case V<sub>o</sub> when in the 1-state will be the supply volts minus

the sum of the two saturated collector-emitter voltages. ● A NAND gate can be

constructed by placing the load R in positive supply line. ● An exclusive-OR gate can be constructed as shown right. Current flows through D and hence in R only when A or B, but not both, are "!". R' serves to limit the current drawn from the supply when both A and B are "!".

• Negative supply voltages can be used if p-n-p optotransistors are used.

Cross references Set 9, cards 8 & 9.



# Set 11: Basic Logic Gates-11

### Analogue gates



#### **Circuit description**

An analogue gate may be considered as an electronic switch which serially connects an analogue signal to a specific input point on the occurrence of a logic or control signal. This is the basis of many multi-channel multiplexers.

One gate that is particularly useful is the c.m.o.s.transmission gate. This comprises a series-pair inverter and two complementary transistors connected in parallel as shown, which allow bidirectional current flow when a control or logic signal is applied to the inverter. High and low signals are applied to the separate gates, and hence both transistors are either on or off simultaneously. The sources and drains of the parallel transistors are tied together, and either terminal may be driven by analogue or digital signals, and the excursion must be within the range of the supply,  $V_{\rm SS}$ to  $V_{\rm DD}$ .

If the control level is lowered to the  $V_{ss}$  rail both transistors are off. This is because the

**Circuit modifications** 

#### **Typical data** IC: 1(CD4016AE), 1(MC14016CL) Supplies (max): $V_{DD} = +7.5V$ , $V_{SS} = -7.5V$ or $V_{\rm DD} = +15V, V_{\rm SS} = 0$ Input analogue signal range: $\pm 7.5$ V or 15V peak 'on' resistance: 300 to $1k\Omega$ (typical) 'off' resistance: $1000M\Omega$ (typical) Feedthrough capacitance: 0.2pF Transfer function linearity: <0.5% distortion into $10k\Omega$ load and $(V_{DD} - V_{SS}) > 10V$

gate-source voltage of the p-type transistor will be zero or positive, and the n-type transistor gate-source voltage either zero or negative for a drive-signal swing limited to  $V_{\rm DD} - V_{\rm ss}$ . When the control signal is at  $V_{DD}$ , both transistors tend to be in the on-state. When the drive signal rises towards  $V_{\rm DD}$ , the p-type f.e.t. will conduct harder because its gate-source voltage will increase negatively. At the same time the n-type f.e.t. conductance will begin to fall, as its positive gate-source voltage is decreasing. The resultant effect is for the parallel arrangement to exhibit a fairly constant conductance and hence constant resistance. The graph is an idealized characteristic and assumes first-order linearity of n-m.o.s. and p-m.o.s. device conductance g<sub>DS</sub> against input voltage, to give a constant  $g_{DS}$  for the parallel connection. The actual non-linearity, but the effect on transfer function is most evident

õ, c



#### **Circuit modifications**

Circuit left is a shunt-series chopper circuit using p-channel junction f.e.ts to gate the analogue signal. The f.e.ts must be fed in antiphase and can be driven from t.t.l. logic provided the f.e.t. pinch-off voltage is less than 4V. When Q is high  $(\overline{Q} \text{ low})$ , Tr<sub>2</sub> is conducting and  $Tr_1$  is off. As  $Tr_2$  is connected to a virtual earth point E, then the signal level at this f.e.t. input is minimum and hence the gate voltage exceeds the sum of the signal voltage and the f.e.t. pinch-off value, a necessary condition for switching. When  $\overline{\mathbf{Q}}$  is high,  $\mathrm{Tr}_2$  is off, and the signal is grounded via the on resistance of  $Tr_1$ . The analogue swing is limited by the maximum signal swing of the i.c. amplifier, and the speed of switching will depend on the slew rate of the amplifier.

Tr<sub>1</sub> & Tr<sub>2</sub> 2N5461, IC 741 or LM301A,  $R_1 = R_f = 10$  to  $33k\Omega$ The junction f.e.ts may be



virtual earth point to provide a signal multiplexer (middle). Signals  $V_1$ ,  $V_2$  and  $V_3$  can be gated in turn by applying complementary control signals to the  $\overline{Q}$ , Q terminals in sequence, say from a ringcounter.

The arrangement right, a series-shunt chopper, is best suited to voltage sources as neither the op-amp or Tr<sub>2</sub> draws significant current when Tr<sub>1</sub> is on, or when  $Tr_1$  is off. The current from the source is negligible. Hence the input signal is gated to the op-amp with almost zero attenuation.

#### Further reading

CD4016A Data Sheet, RCA Solid State Databook Series SSD 203A. Johnson, P. A. Complementary m.o.s. integrated circuits, Wireless World, vol. 79, August 1973, pp. 395-400. Givens, S. FETs as analog switches, Electronic Components, 26 January, 1973. Honey, F. J. DTL/TTL controls large signals in commutator, Electronics, March 1970, p. 90. Jenkins, J. O. M. Interface circuits drive high-level switches from low-level inputs, Electronic Engineering, May 1971.







### Three-state and majority logic



#### Three-state gates

Three-state logic (t.s.l.) is now available from at least three companies: National Semiconductor, Texas Instruments and Signetics. Flip-flops, multiplexers, demultiplexers, line drivers, counters and r.o.ms are among the devices available in this form. The three states are normal 1 and 0 levels plus an off state which represents a high impedance condition in which the gate can neither sink nor source current-effectively an open circuit. Referring to the diagram, if the enable signal is logical 0 then normal logic inversion of the input signal is performed. However, if the enable signal is logical 1, then the input signals are overridden and the device goes into its hi-Z or off state as point A and with it point B are grounded. Hence, both Tr<sub>1</sub> and Tr<sub>2</sub> are switched off and present a high impedance to the load. This means, for example, that a large number of gates can be connected by means of a bus to a single load, only one gate at a time being connected to the load, all others being in the hi-Z state. If the number of gates connected to the bus is high then the leakage current (typically  $40\mu A$ ) of the off devices must be taken into account as the single on-device is supplying the load plus the leakage of all the off-devices. For this reason t.s.l. gates normally have a Darlingtonconnected upper output stage and this increases the source current capability by an order of magnitude over that for normal t.t.l. This in itself is an

advantage of t.s.l. which in addition has the usual advantages of t.t.l. with respect to other logic families. The increased source current capability also carries with it a much reduced one-level output impedance which gives a onelevel noise immunity an order of magnitude better than that for t.t.l.

To ensure in a bus-organized system that no two devices are ever on at the same time, all t.s.l. devices are arranged such that the time delay from on to off is less than that from off to on. Nevertheless, overlaps can occur and although no damage is done to the devices transients resulting from this or any other source can be longer than in a t.t.l. system, principally because more gates will probably be connected to the output of a t.s.l. device and this gives rise to increased load capacitance. Note that all t.s.l. devices are fully t.t.l. compatible and that three-state buffer gates are available to convert any d.t.l. or t.t.l. device into a t.s.l. element.

#### Further reading

Calebotta, S. *Electronic Design*, vol. 20, no. 14, 6 July 1972, pp. 70–2. National Semiconductor, Digital Integrated Circuits Data Book.

#### Majority logic gates

A majority logic gate is a form

of threshold logic gate where the numerical "weight" assigned to each of its inputs is unity. Majority logic uses combinational gates that provide an output in the 1-state (true) only when more than half of the inputs are in the 1-state. To realize this requirement when the number of inputs in the 1-state exceeds the number of inputs in the 0-state by only one (an input majority of one), majority gates normally have an odd number of inputs. It has only recently become possible to design systems based on standard integratedcircuit packages employing majority logic. These packages are of the 16-pin dual-in-line type and contain two identical majority logic gates, each having five inputs and using c.m.o.s. technology. Such majority logic gates allow the design of certain functions, such as those required for communication in the presence of noise and correlation methods, that are difficult to implement with other types of gate. The flexibility of the majority gate can be seen by comparison with the normal AND, OR, NAND and NOR gates which select one input combination from a possible 2<sup>n</sup> combinations of n inputs, whereas the majority gate can select  $2^{n-1}$  of the 2' inputs. A majority gate with only three inputs is possible, as shown left, having an output function K = AB + AC + BCand, on inversion, this can produce the NOR majority function  $K = \overline{AB} + AC + BC$ . The output function of the 5-input majority gate, shown centre, is more complex and is: K=ABC+ABD+ABE+

K = ABC + ABD + ABE + ACD + ACE + ADE + BCD + BCE + BDE + CDE.

By feeding this function to an exclusive-NOR gate together with another function, that may be at logical 0 or logical 1, further flexibility is introduced. When the W-input is at logical 1 (right) K provides the above 5-input majority logic function, K = M5 (say), and when W = 0inversion occurs making  $K = \overline{M5}$ . With D=1 and E=0, K provides the majority of the three inputs A, B and C when W=1 (K=M3) and the NOT majority function K = M3 when W=0. With D=E=1, K provides the three-input OR function when W=1 and the three-input NOR function when W=0. With D=E=0, the three-input AND function is realized when W=1 and this becomes the three-input NAND function when W=0.

#### **Further reading**

Garrett, L., C-MOS may help majority logic with designers' vote. *Electronics*, vol. 46, no. 15, 19 July 1973, pp. 107–12.

Cross references Set 11, cards 5, 6 & 9.





# Set 11: Basic Logic Gates Up-date

1. A novel form of logic gate has been proposed recently that allows relatively complex logic functions to be constructed using a small number of components. They are compatible with existing i.c. processes and can be implemented using devices from standard i.cs. No power gain is available as the active devices are used as series switches, i.e. cascaded gates suffer from loss of signal level as with earlier diode gating systems. In the example shown, two p.m.o.s. enhancement-mode transistors are used. With both inputs

2. Integrated injection logic I<sup>2</sup>L has the lowest speed-power product of presently available gates (1975), and a much higher packing density. The structure is such that the n-p-n transistor Tr<sub>1</sub> is vertical and the lateral p-n-p transistor serves as both a current source and as an active load. Tr<sub>1</sub> has multiple collectors  $c_1$  and  $c_2$ . With the input of the I<sup>2</sup>L gate high (or floating, corresponding to the previous gates being off), then  $Tr_1$  is biased on by the p-n-p current source. Then its outputs are capable of sinking the currents from the current sources at the inputs of the gates to which they are connected.

3. D.m.o.s. transistors are double-diffused metal-oxide devices, in which the channel length L of this n-type enhancement type can be controlled very accurately during production. Since Ron on-resistance, and input capacitance are proportional to L, while  $g_m$  and cut-off frequency are proportional to 1/L and  $1/L^2$  respectively, then these devices provide better performance than conventional n-type f.e.ts. Digital and analogue switch configuration are shown,

For the digital switch.  $(V_{\rm DD} - V_{\rm SS})$  can go up to +30V. With an input signal at Vin,  $V_{\rm out} = \left(\frac{R_{\rm on}}{R+R_{\rm on}}\right) V_{\rm DD}$ 



the output is low. With both inputs low, the devices are off because again each device has  $V_{gs}=0$ . Only if one input is high and the other low is one

device put into a conducting state is the output taken high. The symmetrical nature of the



circuit shows that it does not matter which is taken high, e.g. if A is high and B low. then Tr<sub>2</sub> conducts raising the output close to A (the difference depending on the on-resistance of Tr<sub>2</sub> compared with  $\mathbf{R}_{\mathbf{I}}$ ). The circuit thus implements the exclusive-OR function, and the reference describes other arrangements both m.o.s. and bipolar for producing logic functions with fewer-than-normal components. Reference

Edwards, C. R. Some novel exclusive-or/NOR circuits, Electronics Letters, 1975, 11, pp. 3/4.

If  $V_{in}$  < 700mV, the injected current will be diverted through the low output of the driving gate; hence no base current to p-n-p transistor and the collectors will depend on potential to which they are connected. The OR/NOR gate shown provides  $F = \overline{(A+B)}$  and Z = (A+B). If either input A or B is high,  $Tr_1$  or  $Tr_2$  conducts and F is low. Also, as then the wired-OR connection to the base of Tr<sub>3</sub> is low, Tr<sub>3</sub> is off. and Z tends to be high. Note that I<sup>2</sup>L input/output connections are not connected directly to the package pins, but via appropriate buffer networks.



V <sub>DD</sub>	$R(k\Omega)$	t <sub>r</sub>	ta .	
5 10 15	0.68 0.68 1.0	0.7 0.8 1.0	0.6 0.7 0.9	

The threshold voltage at which the transistor turns on is dependent on  $V_B$  when source and substrate are not connected as in the analogue switch application, and may have to be accounted for to avoid shift in operating point. The input signal is transmitted through the analogue switch when  $V_{\rm C}$  > (most positive peak of  $V_{in}$ ) and is disconnected when  $V_{\rm C} <$  (most negative peak of  $V_{in}$ ).

#### Reference

Theory and applications of DMOS, Electronic Industry, December 1975.

notes

# Set 12: Wideband amplifiers

After considering the suitability of various transistor equivalent circuits, the accompanying article describes the many ways to get extended h.f. performance from transistor circuits. As with valve circuits, there is the well-known limitation to the number of stages that can be cascaded for optimum gain-bandwidth product. To improve on it shunt and series peaking with inductors is the first line of attack, which soon runs into problems of instability and interaction when stages are cascaded. Incidentally, card 2 recalls the dodge, widely used in pulse circuits, of making a parallel RC input network equal in time constant to  $R_{in}C_{in}$ . A popular method of reducing the effect of Miller (or should we say Blumlein) multiplication of input capacitance is the cascode arrangement (cards 5 & 7), considered as a common-emitter, commonbase cascade.

There are other arrangements of two transistors (see also set 20) that can provide good bandwidth; card 12 shows a common-base, common-collector arrangement, related to the d.c. feedback pair of card 9. (For readers who are unclear of the various feedback arrangements around transistor pairs, see Fig. 8, the table and the text on pages 28 and 29.) Integrated circuits offer good solutions in many situations, useful types being variants of the standard 741, 301 devices, and compensation points to modify h.f. characteristics are provided on i.cs such as the LM318 (card 11). A dodge not to be forgotten is to use digital i.cs in a linearly-biased mode; cards 6 & 8 show how, the e.c.l. circuits offering bandwidths up to a few hundred megahertz.

High input impedance c.m.o.s. amplifier 1
Shunt-peaked wideband amplifier 2
High-gain wideband amplifier 3
Wideband voltage followers 4
Bipolar cascode wideband amplifier 5
Wideband amplifier using e.c.l. 6
FET cascode amplifiers 7
Wideband amplifiers with t.t.l., r.t.l., d.t.l. 8
Amplifiers using d.c. feedback pair 9
Gated video amplifier 10
High-speed operational amplifiers 11
Common-base wideband amplifier 12

# Wideband amplifiers

Wideband amplifiers are extensively used in instrumentation and communication systems where the signals to be handled may be of an analogue or a digital nature. Such amplifiers are required to provide fairly equal amplification of a large range of frequencies with a lower frequency limit of zero or nearly zero. The high-frequency behaviour of the active devices must be considered in conjunction with the passive network elements to design an amplifier with required characteristics.

Many different circuit models are available for such devices, but not all are necessarily useful. Consider, for example, the bipolar junction transistor models shown in Fig., 1. The three versions of the intrinsic low-frequency equivalent circuit shown in (a), (b) and (c) are highly-idealized models that focus attention on the basic active properties of the device. The low-frequency T and h-parameter models shown in (d) and (e) are more useful, but are unsatisfactory for predicting amplifier performance in the region of its upper cut-off frequency.

Models that are useful for high-frequency design must provide a more realistic representation of the transistor as a network element. The high-frequency model one chooses is often determined by availability of data, personal preferences and experience or whether the parameters of interest are readily determined by measurement. Fig. 2(a) shows the hybrid equivalent circuit which is a reasonable compromise between accuracy and complexity and which may be reduced to simpler forms for low-frequency and high-frequency calculations.

In high-frequency transistors,  $r_{ce}$  and  $r_{b'e}$ are often sufficiently large to be neglected. the former being much larger than the load impedance and the latter much greater than the reactance of  $C_{b'c}$  at high frequencies. The simplified form of Fig. 2(b) is often sufficiently accurate for assessing high-frequency performance and this may be reduced to that shown in Fig. 2(c), where  $C_{in}$  consists of  $C_{b'e}$  in parallel with the Miller effect equivalent of  $C_{b'c}$ . The base spreading resistance  $r_{bb'}$  and the product  $g_m r_{b'c}$  may normally be assumed to be independent of the operating point,  $g_m$  increasing and  $r_{b'e}$ decreasing as  $I_E$  increases. The value of  $C_{b'e}$  increases with  $I_E$  and the depletionlayer capacitance  $C_{b'e}$  varies as  $1/(V_{CB})^{\frac{1}{2}}$  to  $1/(V_{CB})^{\frac{1}{2}}$ .

To obtain a wide bandwidth with the simple cascade of common-emitter stages shown in Fig. 3, the collector coupling re-



sistors must be made small compared with. the input impedance of the following stage, the capacitive component of which causes the gain to fall at high frequencies. Further reduction of  $R_c$  to exchange gain for bandwidth is limited by the presence of  $R_{hh'}$ . Also, if the gain per stage is reduced, more stages must be cascaded to achieve a desired amplifier gain and it becomes increasingly difficult to maintain the overall bandwidth which shrinks as the number of cascaded stages increases. The gain-bandwidth product of the transistors  $(f_T)$  attains a maximum value at a particular value of emitter current, which is often small. Adjusting the emitter current of each stage to its optimum value may result in a small signal-handling capability if significant distortion is to be avoided.

Several techniques are available for improving the achievable stage gain-bandwidth product, the simplest of which is the inclusion of an inductor to compensate for the falling response due to transistor input capacitance. The stages in Figs. 4(a) and 4(b) are said to be shunt-peaked and seriespeaked respectively, the latter being far less effective in improving gain-bandwidth product. The effect of the shunt-peaking inductor is illustrated in Fig. 5 and by correct design the stage bandwidth, for a given gain, can be improved by a factor of about two without lifting the high-frequency gain above its low-frequency value. Too large a value of L results in overcompensation which produces overshoot and ringing in ' the transient response. Amplifiers using a number of these stages in cascade may suffer from instability and prove difficult to align.

This problem may be alleviated by making the effective load on each transistor resistive. Referring to Fig. 6, this can be achieved by considering L and  $R_c$  to be in parallel with the series equivalent of the CR input network, making  $\mathcal{R}_{c}$  equal to the equivalent series input resistance and designing L to produce the same short-circuit time constant for each of the parallel branches. A disadvantage of this constantresistance cascade is that it is no longer possible to vary a network element to adjust amplifier gain or bandwidth, which must be attempted by variation of the transistor parameters, e.g. by adjusting the individual collector currents. The foregoing techniques have the disadvantage that all the cascaded stages interact, any change in the design of one stage normally requiring changes in the others.

As the input capacitance, including that due to Miller effect, plays an important part in limiting the achievable bandwidth, a design approach that attempts to eliminate the effects of internal feedback is useful. The cascode amplifier shown in Fig. 7 employs this technique and it may be considered as a common-emitter common-base cascade. The common-base stage has a very low input impedance, so the common-emitter stage has a current gain approaching hre and a very small voltage swing at its collector, resulting in a large reduction of the internal feedback between collector and base. The bandwidth of the common-emitter stage approaches  $f_{\theta}$  and as that of the commonbase stage is much larger, the cascode pro-







Fig. 4. Simplest way of improving bandwidth is to add shunt (a) or series (b) compensation. Effect of shunt peaking—the most effective—is shown in Fig. 5.



Fig. 6. To avoid instability in cascaded circuits of Fig. 4(a), the transistor load can be made resistive.



Fig. 7. Cascode circuit minimizes effect of internal feedback.



Fig. 8. Four basic ways of increasing bandwidth using negative feedback.



Fig. 9. Series and shunt feedback applied to a single stage.



Fig. 10. Peaking capacitors improve h.f. response of Fig. 9.



Fig. 11. Deliberate mismatching of impedances can improve stability.

vides the voltage gain and current gain of a common-emitter stage with a wider bandwidth than that obtainable with a simple common-emitter amplifier.

Satisfactory design of wideband amplifiers usually requires the interaction between individual stages or elementary building blocks to be negligible, or definable, the mid-band gain to be stable and input and output impedances to be adjustable to desired values. Use of feedback in the design allows these criteria to be approached without undue concern for the variations in transistor parameters and permits bandwidth to be extended at the expense of gain in a controllable manner. While the reduction in gain is a disadvantage it is not an expensive price to pay, bearing in mind the benefits obtained and the relatively low cost of adding extra feedback stages to meet the overall gain requirement.

Another disadvantage of feedback is the increased possibility of oscillation, which may be avoidable at the design stage by using a sufficiently accurate circuit model. In a multi-stage amplifier designed for the highest possible bandwidth before the application of feedback, the cut-off frequencies of all stages will normally be similar. Hence there is a near certainty that the combined phase shift can reach 180° while the magnitude of the gain is well in excess of unity. To remove this possibility, by deliberately setting one of the cut-off frequencies much lower than the others, negates the original requirement for maximum pre-feedback bandwidth. General-purpose operational amplifiers, such as the 741-type, have internal stages with high cut-off frequencies, but a dominant lag at about 10Hz has to be introduced to cope with the possibility of 100% feedback.

Four basic feedback configurations may be used to create elementary building blocks. Fig. 8 shows these configurations which may be described in terms of the method in which the feedback is derived and applied. Thus, (a) is series-derived seriesapplied, (b) is shunt-derived series-applied, (c) is shunt-derived shunt-applied and (d) is series-derived shunt-applied. Alternative descriptions of these configurations are in common usage e.g. (a) transimpedance feedback, series-series or simply series, (b) voltage-ratio feedback or series-shunt, (c) transadmittance feedback, shunt-shunt or simply shunt, and (d) current-ratio feedback or shunt-series. Other descriptions include the use of the terms current feedback or voltage feedback. In the former case the signal fed back is proportional to the output current but may itself be a current or a voltage. With voltage feedback the fed-back signal is proportional to the output voltage. All four arrangements have the property of increased bandwidth and reduced gain compared with the open-loop values. The input and output impedances become modified as shown in the table, shunt derivation (application) reducing the output (input) impedance and series derivation (application) increasing the output (input) impedance.

The two types of single-stage feedback, series and shunt, are shown in Fig. 9. In Fig. 9(a) the load impedance should be low and the input supplied from a voltage source

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whereas in Fig. 9(b) the load impedance should be high and the amplifier input should be from a current source. The highfrequency response of these elementary blocks may be improved by the addition of peaking capacitors as shown in Fig. 10.

Source and load impedance requirements for stable gain with single-stage feedback can be met by purposely creating a large mismatch between cascaded stages, i.e. by alternating series and shunt feedback stages as shown in Fig. 11. With this arrangement gain-bandwidth product is high, lowfrequency gain may be determined with reasonable accuracy by multiplication of individual stage gains, and there is little interaction between stages. Feedback applied to two stages can offer similar merits, Figs. 12(a) and 12(b) showing series-derived shunt-applied and shunt-derived seriesapplied configurations respectively. Both networks have widely differing input and output impedances and are therefore attractive as basic building blocks for cascaded stages

Although discrete devices may be used in a multistage realization, the availability of integrated-circuit transistor arrays, containing about five transistors with parameters inherently reasonably matched, are very attractive for many designs. Integrated circuits are available in the form of a longtailed pair which can be operated as singleended or differential-input wideband amplifiers by the addition of external passive components that allow a high degree of flexibility in the selection of gain, bandwidth and signal-handling capability.

Other integrated circuit versions provide emitter-follower input and output transistors to make the input and output impedances high and low respectively. Integrated wideband power amplifiers can be obtained providing bandwidths up to about 8MHz and outputs of about 14V pk-pk. Integrated circuits containing pairs of cascode amplifiers and some with gating facilities are also very useful as wideband amplifiers. Modern integrated-circuit logic gates designed for high-speed switching applications may also be usefully employed as low-cost wideband amplifiers by setting the quiescent conditions in a linear part of the transfer characteristic and employing feedback to define the gain.

Feedback type	Zout	Z <sub>in</sub>
Fig. 8(a)	increases	increases
Fig. 8(b)	decreases	increases
Fig. 8(c)	decreases	decreases
Fig. 8(d)	increases	decreases

### High input impedance c.m.o.s. amplifier



#### **Circuit description**

C.m.o.s. circuits have been designed for digital systems. Basic building block is the inverter consisting of a complementary pair of enhancement-mode m.o.s. devices with threshold voltages in the region of 1.7V and peak current capabilities of a few tens of milliamperes at maximum forward bias. Input resistance is extremely large with the leakage currents of the protection diodes as the limiting action. The individual device gm is around 1 to 2mS for forward gate-source voltages in excess of 5V. Biased in the linear mode, the devices each have  $|V_{gs}| \approx V_s/2$ , so maximum forward bias is  $\approx 7.5V$  for the rated supply voltage of 15V. This restricts the upper value of gm that can be attained. In the absence of an external load resistance the voltage gain is limited by the finite slope resistance. An unusual property of this circuit is that the voltage gain falls with increasing supply voltage, as the relatively small rise in g<sub>m</sub> is more than offset by the fall in output slope resistance at the higher resulting currents. The disadvantage of operating at low supply voltages (and hence currents) is that the voltage gain is more loaddependent and the upper cut-off frequency falls sharply. To maximize the input resistance while using d.c. negative feedback to stabilize the operating point, the feedback resistor is decoupled (see Fig. 2). Gate current is negligible and the input and output quiescent voltages are equal. Other c.m.o.s. devices using different fabrications, e.g. silicon gate, ion implantation, offer greater bandwidths and/or lower threshold voltages. Complementary m.o.s. circuits designed for the linear market have been recently announced.

#### **Component changes**

IC<sub>1</sub>: Any c.m.o.s. inverter including CD4001 NOR gate, CD4009/CD4049 inverting buffers, MC1407, MC14009 and other equivalents. R<sub>1</sub>: 220k to 22M $\Omega$ . C: Sets the lower frequency at which the negative feedback becomes operative, reducing the input impedances. 0.1 to 10 $\mu$ F (time constant R<sub>1</sub>C/2). C<sub>2</sub>: Depends on load which, with this circuit, cannot be very low unless transconductance properties alone required with little voltage gain. 0.1 to  $100\mu$ F. V<sub>s</sub>: 3 to 15V. At voltages below 5V the quiescent current becomes very small and output impedance is very high.

#### **Circuit modifications**

• Negative feedback may be applied over any odd number of inverting stages, e.g. over three as in the circuit shown left. Multiple phase shifts within the amplifier, together with the high voltage gain available, make the method difficult to apply without precautions. Open-loop gains of 80dB and greater may occur. Even decoupling the feedback still allows oscillation at low frequencies via the coupling/decoupling/supply time-constants.

• A better method of controlling gain than simply attenuating it is to cascade stages with shunt and series negative feedback as shown centre where inserting a resistor in the source of each device in stage 1 reduces and defines the transconductance. Shunt feedback across stage 2 defines its transresistance and the combination has a lower but well-defined voltage gain.

• Because the above amplifier is non-inverting, d.c. feedback may be applied via the third stage used purely as a d.c. coupling element linking A-A', B-B', with the overall a.c. feedback decoupled.

#### Further reading

RCA, COS/MOS Digital Integrated Circuits, 1973, pp. 24–30, pp. 37–42, p. 346, p. 355. Wujek, J. H. Field-effect transistor circuits, in Electronic Circuits Manual by J. Markus, McGraw-Hill 1971, p. 22. Ferranti Ltd., FET Source Follower and Bootstrapped FET Source Follower, FETs and Applications, no. 22, 1965.

#### **Cross references**

Set 12, cards 4 & 7. Set 11, cards 5 & 6.



# Set 12: Wideband amplifiers-2

graph | L(µH)

3.33

(1) (2) (3) (4) (5) (6)

### Shunt-peaked wideband amplifier



Circuit data/ Supply: +6V, 4.5mA Tr<sub>1</sub>:  $1/5 \times CA3046$ C:  $10\mu$ F tantalum R<sub>1</sub>:  $68\Omega$ ; R<sub>2</sub>:  $100k\Omega$ R<sub>3</sub>:  $680\Omega$ L:  $1.83\mu$ H (38 turns 28 swg, close wound on 0.2in dia.)

#### **Circuit description**

Bandwidth of a commonemitter amplifier may be extended by including an inductor in shunt with its input producing a shunt-peaked amplifier. The circuit is fed from a current source, or alternatively in a cascaded amplifier  $R_1$  and L may be used as the collector coupling network of each stage which is loaded by the input impedance of the following transistor.

Because the gain-bandwidth product of a transistor varies with its collector current, R<sub>2</sub> was chosen to maximize the gain-bandwidth product for the device used. Resistor R<sub>3</sub> was chosen to make  $V_{\rm CE} \approx + V/2$ and  $R_1$  to provide the desired low-frequency current gain. Source resistance was  $100\Omega$ , so a current source was simulated by a chain of ten  $1-k\Omega$  resistors in series to reduce the effect of the shunt capacitance of a single 10-k $\Omega$  resistor on the measured high-frequency response. With L=0, amplifier input voltage falls with increasing frequency due to transistor input (and Miller effect) capacitance. With L in circuit, low-frequency response is the same but a low-Q resonance occurs between L and Cin of the transistor, causing the response to peak as the frequency is raised towards its original cut-off value.

By designing L on the basis of available transistor parameter data and the required current gain, it is possible to produce a maximally-flat gain response while maximizing stage gainbandwidth product. Alternatively, a good starting point is to use the measured low frequency input resistance and upper cut-off frequency with L=0 to estimate C<sub>in</sub> and then design L to resonate with this value in the region of the desired cut-off frequency with the peaked response. (See curve 3 above.)

#### **Component changes**

Useful range of supply +3 to +30V. Values of R<sub>2</sub> and R<sub>3</sub> should be adjusted to produce f<sub>T</sub> max. for transistor if maximum stage gain-bandwidth product is to be achieved. Select R<sub>1</sub> to give desired low-frequency gain, and C to give desired lower cut-off frequency. Highfrequency response may be tailored to suit requirements by choice of L value. (See curves 2, 4, 5 & 6.)

#### **Circuit modifications**

• Up to five cascaded stages may be obtained with the same integrated circuit package but it may prove easier to obtain the desired response experimentally than by calculated design. Problems of alignment and possible instability due to inter-action between cascaded shunt-peaked stages can be alleviated if each transistor feeds

a resistive load. The load on each device consists of L and R<sub>1</sub> in parallel with the input impedance of the following stage, as shown left. Load impedance may be made purely resistive by adjusting the values of  $R_1$  and L to be  $r_{bb}'(r_{bb}' +$  $R_{\rm in})/R_{\rm in}$  and  $C_{\rm in}r_{\rm bb}'$  respectively, where  $r_{bb}$  the transistor base spreading resistance. In such a constant-resistance cascade, L and R<sub>1</sub> cannot be used to adjust the gain and bandwidth of the amplifier which must be trimmed by adjusting the collector currents of the transistors.

• A series-peaking inductor may be connected between a lowimpedance source and the base, as shown centre, to extend the bandwidth. Stage gainbandwidth product is less than for the shunt-peaked arrangement and its maximum value cannot be achieved with a maximally flat response.

• Another high-frequency compensation network, widely used in pulse applications, is shown right. Bandwidth may be extended by making  $R_1C_1 = R_{in}C_{in}$  but the stage gainbandwidth product, while constant, does not approach that of the shunt-peaked stage. However the constancy of the gain-bandwidth product does allow exchange of gain for bandwidth by adjusting  $R_1$ .

### Further reading

FREQUENCY (Hz)

Ghausi, M. S. Electronic Circuits, Van Nostrand, 1971, chapter 4. Shea, R. F. (Ed.), Amplifier Handbook, McGraw-Hill 1966, section 25.4.2. Joyce, M. V. & Clarke, K. K. Transistor Circuit Analysis, Addison-Wesley, 1963, chapter 8. Markus, J. Electronic Circuits Manual, McGraw-Hill, 1971, p. 913.

#### **Cross references**

Set 12, cards 5, 9 & 12. Set 7, card 9



# Set 12: Wideband amplifiers-3

### High-gain wideband amplifier







#### **Circuit description**

This amplifier uses an integrated circuit developed for consumer applications. It is simple in design, consisting of three common-emitter transistors directly coupled so that the collector potentials of Tr<sub>1</sub>, Tr<sub>2</sub> are defined by the base potentials of Tr<sub>2</sub>, Tr<sub>3</sub> respectively, i.e. they are at  $\approx 0.6V$ . Resistor R<sub>5</sub> has a low value with only a marginal effect on these potentials. Output potential is forced by the feedback effect via R<sub>6</sub>, R<sub>7</sub>,  $\mathbf{R}_{\mathbf{8}}$  to be some defined multiple of the base potential of Tr<sub>1</sub>. If  $R_8 < R_6$ ,  $R_7$  then the output p.d. to ground is  $\approx (R_6/R_7 + 1)$  $V_{\text{be}_1}$  and would typically be set near the supply mid-value. Current in each stage increases with increasing supply voltage, and as a result both open-loop gain and bandwidth increase. Increased current also reduces the input impedance. Resistor R<sub>9</sub> determines quiescent output current provided output voltage has been fixed by feedback as outlined above. As there are three stages all contributing phase shift, the total phase shift is in excess of 180° before the magnitude of the gain falls

towards unity. Hence it is difficult to apply heavy negative feedback to extend the bandwidth at the expense of gain as can be done with amplifiers using common-emitter stages. (For audio frequency applications it is possible to control the feedback to produce defined characteristics as for tape recording/playback amplifiers.) As the high gain extends to high frequencies,  $A_v \approx 100$  at f = 4MHz, proper grounding, decoupling and use of short leads, screened where appropriate are all important.

#### **Component changes**

IC: No direct replacements but groups of three transistors from CA3046 and similar multitransistor packages may be used.  $R_6$ ,  $R_7$ : Ratio fixes output quiescent voltage  $\approx (R_6/R_7+1)$ 0.6V. Resistors may be increased to minimize loading on output, but base current of  $Tr_1$  becomes significant factor in determining quiescent conditions. Typical range 5k to 50k  $\Omega_1$ .

 $R_s$ : Also limited by base current effects, but may be increased up to  $10k\Omega$  to maximize input

resistance of circuit if required.  $R_9$ : Determines output stage quiescent current and hence maximum load current. No advantage to making too high since current in preceding stages fixed by internal resistors. 2.2k to  $22k \Omega$ .

 $+V_s$ : 2 to 18V. At low voltages, gain falls sharply Choose R<sub>6</sub>, R<sub>7</sub> in conjunction with V<sub>s</sub> to bias output near mid-point of supply for maximum output swing.

 $C_1$ ,  $C_3$ ,  $C_4$ : Determine low frequency cut off. Maintain ratio increasing values for pro rata fall in cut-off.  $C_4$  may be reduced greatly if feeding high impedance load.

#### **Circuit modifications**

• Any group of three transistors may be used in similar configurations. Simple arrangement (left) for lowvoltage applications where output quiescent voltage of 0.6V would be adequate. Again decoupling of feedback essential since gain/phase properties too complex for application of heavy negative feedback. Resistors  $R_1$  to  $R_5$ may be comparable value resistors (1k to 10k  $\Omega$ ) though  $R_4$ ,  $R_5$  may be higher if somewhat higher output direct voltage desired.

• By using complementary transistors, each transistor has current defined by  $V_{be}$  of succeeding stage. This makes current and gain/ bandwidth less dependent on supply voltage (middle). Complementary versions of each form are equally possible, reversing the supply voltage.

• The good high frequency gain allows its use in 10.7MHz i.f. applications (right) with the d.c. feedback applied via the input tuned circuit and the output taken to a ceramic filter. Similar principles apply at 470kHz but the high voltage gain makes careful design obligatory. In all r.f. applications drive from a 50 $\Omega$  source is normal, but the output impedance of the circuit is relatively high.

#### Further reading

Motorola Linear Integrated Circuits Data Book, 2nd edition 1972, pp. 7–561 to 7–564. CA3095E Wideband Amplifier, RCA Databook SSD-201A, 1973, p. 249.

Cross references Set 12, cards 5, 10 & 11.



# Set 12: Wideband amplifiers—4

### Wideband voltage followers



#### **Circuit description**

A specially-designed monolithic voltage-follower, the 310 has a combination of highly desirable parameters: high input impedance, low output impedance, and wide frequency range. The negative feedback is connected internally but some modifications to performance that can be made externally include offset zero by means of R<sub>1</sub>, increased output current capability by preloading with  $R_4$  at the expense of increased dissipation, and a.c. coupling to source, bootstrapping the bias resistors R2, R3 to minimize loading effects.

Slew-rate limitations imposed by the maximum charging rate of an internal compensation capacitor, mean that the large signal bandwidth is much less than small signal bandwidth true of amplifiers having heavy negative feedback in general.

#### Performance

Upper cut-off frequency for small signals. 15MHz for zero source resistance, 2MHz for 10k $\Omega$ . At 1MHz, output impedance is 25 $\Omega$ , output voltage swing is  $\pm 3V$ , and input impedance is 3pF plus strays.

#### **Component changes**

IC: Direct replacements from many manufacturers. Any op-amp compensated for 100% negative feedback may be used. In general it will not be possible to achieve the same spread of parameters, i.e., high input impedance may preclude wide bandwidth.

 $R_2$ ,  $R_3$ : Increase for higher  $Z_{in}$  at expense of increased offset.

C<sub>2</sub>, C<sub>1</sub>: Determine lower cut-off frequency (with R<sub>1</sub>, R<sub>3</sub>). Combination gives inductive term to input impedance at low frequencies. 0.1 to  $10\mu$ F. R<sub>4</sub>: Increases negative current capability at expense of dissipation.

V:  $\pm 5$  to  $\pm 15V$  (down to  $\pm 3V$  in extreme cases).

#### Further reading

LM110—An Improved IC Voltage Follower, National Semi-conductor Linear Applications Handbook, 1972, p. LB11–1 and 2. Campbell, T. C. Application note ICAN-5213, in RCA Databook SSD-202, 1972, pp. 203–4. Markus, J. Electronic Circuits Manual, McGraw-Hill, p. 13.

#### Cross references

Set 12, cards 1 & 11. Set 7, card 3.



#### **Circuit description**

Using the five transistors in an i.c. package (CA3046), this is a simple wide-band voltage follower. A similar circuit could be used with discrete transistors replacing current mirror Tr<sub>4</sub>, Tr<sub>5</sub> by a similar current source or by a single resistor particularly if the supply voltages were higher. Transistors Tr<sub>1</sub>, Tr<sub>2</sub> are the differential pair comparing input with output, the use of a monolithtic pair giving good temperature drift. Transistor Tr<sub>3</sub> is an emitter follower within the feedback loop. Resistor R<sub>4</sub> may be required to suppress very high frequency oscillations (dependent on circuit layout). Not shown, but generally important, is the decoupling of the supply voltages as close to the i.c. as possible—typically 0.01 to  $1\mu$ F.

#### Performance

Upper cut-off frequency  $\approx 4.5$  MHz. Output impedance  $\approx 10\Omega$ . Input impedance:  $\approx 100 k \Omega$ . Output voltage swing  $\approx 2.5$  V pk-pk at 100 kHz.

#### **Circuit modifications**

● Replace Tr<sub>4</sub>, Tr<sub>5</sub> by any other constant current source (or even resistor, with reduced stability against supply changes).
 ● Increase open-loop gain output capability, by replacing R<sub>2</sub>, R<sub>3</sub> with constant current sources.

• Replace  $Tr_3$  by common emitter p-n-p transistor taking signal from  $Tr_1$  collector instead. Higher open-loop gain, greater output swing capability, greater likelihood of oscillations requiring compensating Circuit data  $Tr_{1-5}$ : CA3046  $R_2$ : 3.9k $\Omega$   $R_1$ : 4.7k $\Omega$   $R_3$ : 1k $\Omega$   $R_4$ : 1k $\Omega$ Supply:  $\pm 2.5V$ 

capacitor, e.g., collector-base of  $Tr_3$ .

#### **Component changes**

transistors used.

IC: CA3046, CA3045, CA3086.  $R_1$ : 1k $\Omega$  to 100k $\Omega$ . Lower values increase bandwidth and dissipation.  $R_2$ : Set to carry half current in  $R_1$  (assuming unity ratio for current mirror). Typically  $R_1 \approx R_2$ .  $R_3$ : Low values maximize output swing and dissipation 330 $\Omega$  to 33k $\Omega$ . V:  $\pm 2$  to  $\pm 7\frac{1}{2}$ V. Higher voltages possible if discrete

# Series 12: Wideband amplifiers—5

FREQUENCY (MHz)



### **Bipolar cascode wideband amplifier**

Circuit data Supply: +6V, 1.2mA Tr<sub>1</sub>, Tr<sub>2</sub>:  $1/5 \times CA3046$ R<sub>1</sub>:  $39k\Omega$ ; R<sub>2</sub>:  $3.3k\Omega$ R<sub>3</sub>:  $33k\Omega$ ; R<sub>4</sub>:  $2.2k\Omega$ R<sub>5</sub>:  $1k\Omega$ C<sub>1</sub>, C<sub>2</sub>, C<sub>3</sub>:  $1\mu$ F V<sub>1n</sub>: 10mV r.m.s.



The bipolar transistor cascode amplifier offers very good isolation between its input and output ports, so that the possibility of instability is virtually non-existent for all types of load. Input transistor Tr<sub>2</sub> acts as a common-emitter stage which is loaded by the upper transistor connected in the common-base configuration and having a very low input impedance. Thus the commonemitter stage has a current gain approaching h<sub>fe</sub> and the voltage swing at its collector is very small, producing a significant reduction of the internal feedback between collector and base. Bandwidth of the common-emitter stage approaches  $f_{\beta}$  and as that of the common-base stage is much larger, the complete circuit behaves like a commonemitter amplifier as far as voltage gain and current gain are concerned, but has a wider bandwidth than is achievable with a simple common-emitter stage providing the same high voltage gain.

A reasonable estimate of the 3-dB bandwidth obtainable with the cascode amplifier may be made using available data and the relationship: gainbandwidth product  $(f_{\rm T}) \approx h_{\rm fe} \cdot f_{\beta}$ if it is assumed that  $h_{fe}$  and  $f_{\beta}$ are the approximate current gain and bandwidth values for the whole circuit. Each of the transistors in the integrated circuit array has typical hre and  $f_{\mathbf{T}}$  values of 110 and 450MHz respectively, with  $I_{\rm C} \approx 1.2 {\rm mA}$  giving  $f_{\beta} \approx$ 4.1MHz. (Measured upper

3dB cut-off frequency was approximately 3.7MHz for the circuit.)

#### **Component changes**

Useful range of supply +2 to +30V.

 $V_{in}(max) \approx 40 \text{mV}(\text{pk-pk}).$ Midband gain falls by 3dB when output is loaded with approximately 6.8k  $\Omega$ (capacitively coupled). Upper cut-off frequency largely determined by  $\mathbf{R}_4$  for given transistors. Lower cut-off frequency adjustable by changing  $C_1$ ,  $C_2$ and  $C_3$ .  $Tr_1$  and  $Tr_2$  may each be replaced by  $\frac{1}{4} \times CA3018$  or equivalent discrete devices e.g. 2N706-type.

#### **Circuit modifications**

•One of the differentiallyconnected pair of transistors in the CA3046 package may be connected as an output emitter-follower stage to provide a low output impedance cascode amplifier. If a CA3018 package is used, all four transistors may be used by following the cascode with a

pair of cascaded emitter followers as shown left. In this form, with  $R_6 = 3.3 \mathrm{k} \Omega$  and  $R_7 = 2.7 \mathrm{k} \Omega$  a voltage gain of about 36dB is obtainable with an upper cut-off frequency of approximately 11 MHz. •A pair of cascode stages with a common tail resistor may be used as a differential cascode amplifier. This arrangement is the basis of the CA3040 integrated circuit wideband amplifier, which employs high-input-impedance emitterfollower buffer stages between the input terminals and the common-emitter sections of the cascodes. Outputs from the cascode common-base stages are fed to the output terminals via emitter followers to produce a reasonably low output impedance.

•Centre circuit shows the CA3040 connected to a pair of equal load resistors ( $R_L$ ) which receive antiphase outputs. With the amplifier connected to a 50- $\Omega$  source and  $R_L$ =50 $\Omega$  the voltage gain to each output is about 22dB with a bandwidth of approximately 30MHz. The values increase to about 32dB and 50MHz respectively when  $\mathbf{R}_{\mathbf{L}}$  is raised to  $1k\Omega$  and the bandwidth may be increased to about 90MHz by including the input series peaking circuit shown right. Supplies must be decoupled with capacitors  $C_1$ and  $C_2$  at the integrated-circuit package pins and ferrite beads on the supply leads and careful printed circuit layout are necessary. Supply  $\pm 6V$ , R<sub>1</sub> 50 $\Omega$ , C<sub>1</sub>, C<sub>3</sub> 100nF, C<sub>2</sub> 1nF, C<sub>4</sub> is a small trimmer to adjust highfrequency gain and phase balance.

#### Further reading

Theriault, G. E. et al. Application of the RCA-CA3018 Integrated-Circuit Transistor Array, RCA Databook SSD-202 1972, pp. 68–70. Austin, W. M. Principal Features and Applications of the RCA-CA3040 Intregrated-Circuit Wideband Amplifier, RCA Databook SSD-202, 1972, pp. 171–81.

Cross references Set 12, cards 2, 3, 7 & 12. Set 7, card 5.



Wideband amplifier using e.c.l.

# Set 12: Wideband amplifiers—6

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 Typical performance

  $V_{EE}$  -5.2V, 20mA

  $R_1$ ,  $R_2$  390  $\Omega$ 
 $C_1$  100nF,  $C_2$  100pF

  $C_3$  1 $\mu$ F tantalum

 IC
 MC1001

 Voltage gain (1.f.) 2.89

  $V_{in}$  73mV r.m.s.



#### **Circuit description**

The integrated circuit is an emitter-coupled logic gate designed for high-speed digital applications. This type of gate has an input stage consisting of a long-tailed pair with multiple inputs to one half of the pair via separate transistors having their collector-emitter paths in parallel. Each half of this differential stage feeds an emitter follower output stage, a number of which may be paralleled within the package. Thus, at least one inverted and at least one non-inverted output is available and the logic gate may be used as a linear, wideband amplifier provided it is correctly biased, for example by means of  $R_1$  and  $R_2$  as shown above. This simple arrangement of connecting one of the logical NOR outputs back to the input provides self-biasing of the gate that ensures that the d.c. input and output voltages are the same except for the small drop across  $R_1$  and  $R_2$ . This is only possible if the gate is operating in the middle of its transfer characteristic, the arrangement automatically compensating for changes in bias and offset voltages.

Only the d.c. component of the

output signal is fed back to the input,  $C_2$  and  $C_3$  decoupling the a.c. component. Capacitor C<sub>1</sub> prevents any d.c. component of the input signal from disturbing the self-biasing conditions. The a.c. signal voltage available at the NOR output used to derive the feedback, is typically 3 to 4% lower than that obtainable from other NOR outputs, and is approximately 9 to 10% higher than at the unloaded NOR outputs. Careful printed circuit layout is required if a smooth, highfrequency roll-off is to be obtained in the frequency response.

#### **Component** changes

Useful range of  $V_{\rm EE} \approx -3$  to -8V (see graph). Minimum load resistance for 10% fall in mid-band gain  $\approx$  270 $\Omega$  (a.c. coupled via 100nF capacitor).

Distortion is less than 1% for  $V_{in}$  in the range approximately 1 to 20mV and useful  $V_{in}$  (max) without significant distortion is approximately 200mV pk-pk. For bandwidths in excess of 300MHz use MC1660.

#### **Circuit modifications**

• More than one input of the gate may be used to increase gain by connecting others in parallel, but the offset voltage between output and input rises as the number of paralleled inputs increases. This may not be a severe disadvantage in certain applications.

• The simple self-biasing arrangement may be applied over several e.c.l.-gate amplifying stages, as shown left. • A given total bias network resistance  $(R_1 + R_2)$  may be decoupled asymmetrically. For example, if it is required that the bias-network loading of the output be reduced without changing  $(R_1 + R_2)$ , values such as those shown centre may be used. This will cause the input impedance of the amplifier to fall from about  $390\Omega$  to about  $220\Omega$ 

• The biasing arrangement may be changed to allow an increase in both the input and output impedances without changing the d.c. input and output voltages. One such arrangement is shown right, where the junction of  $R_1$  and  $R_2$  may be returned to the 0-volt rail through a suitable-value, additional bias resistor  $R_3$ . For example, with  $R_1$ ,  $R_2$  2.2k $\Omega$ , a suitable value for  $R_3$  would be around  $47k\Omega$ . This arrangement slightly increases voltage gain.

#### Further reading

Using e.c.l. gates as wideband amplifiers, *Electronic Engineering*, June 1973. MECL Integrated Circuits Data Book, Motorola, 2nd edition, 1972, pp. 5-3 to 5-6 and pp. 4-17 to 4-20.

Cross references Set 11, card 7. Set 12, cards 8 & 10.

Circuit modifications  $R_1$   $C_2$   $C_2$  $C_2$
## Set 12: Wideband amplifiers—7

### **FET cascode amplifiers**



### Circuit data

 $\begin{array}{l} Supply: \pm 7.5V\\ IC: CD4007AE\\ R_1, R_2: 1M\Omega\\ C_1: 0.1\mu F; C_2: 0.33\mu F\\ Voltage gain: \approx 26(R_1 \ 33k\Omega)\\ Cut-off \ frequency: \ 320kHz\\ Alternative bias \ for single \ supply: R_3, R_4: 1M\Omega\\ C_3: 1\mu F; C_4: 0.01\mu F \end{array}$ 



#### **Circuit description**

The basic principle of a cascode stage is that of a commonemitter or common-source amplifier feeding directly into a common-base or common-gate stage, so that the first stage feeds into a near short-circuit. This allows it to develop its maximum current gain or transconductance, but more important the voltage swing at the collector or drain is small. Large voltage swings produce relatively large currents in the inevitable feedback capacitance, with a corresponding heavy shunting effect at high frequencies. This reduction in feedback (commonly called the Miller effect though first discussed by Blumlein) is of particular importance in wideband and r.f. amplifiers. In particular, the better the isolation between input and output the higher the gain that can be achieved consistent with stability. The devices need not be in series for d.c. purposes though this is more common. In the circuit shown, based on a low-cost c.m.o.s. i.c., Tr<sub>2</sub> is a common-source stage whose output signal current is diverted into  $Tr_3$  provided that  $Tr_1$  has a high dynamic impedance at a.c. (see circuit modifications). The voltage gain of the cascode pair is equal to or greater than the highest voltage gain obtainable from a single stage, while coupling between output and input is minimized. An alternative version is shown in which a decoupled d.c. feedback path via R<sub>3</sub>, R<sub>4</sub> gives

self-biasing (with a.c.-coupled source). The high input resistance of the m.o.s. devices together with the possibility of operation at  $V_{dg}=0$  allows for simple biasing networks and offsets the disadvantage of the wider tolerance on m.o.s. devices as against bipolar transistors.

#### Component changes

IC: Equivalents of CD4007AE including MC14007, etc. Other combinations of enhancement-mode m.o.s.f.e.ts may function but simpler circuits can be devised for these. This circuit was intended to make best use of this low-cost i.c.

Supply voltage: Positive voltage may need to be increased with some devices; negative rail is then reduced accordingly to stay within rating.  $R_1, R_2, R_3, R_4$ : Not critical. 100k  $\Omega$  to 22M  $\Omega$ . With low resistances, low-frequency time constants are raised unless large capacitances used, while input impedance falls for second version. Very high resistances bring noise/hum/leakage

#### problems.

 $C_1$ ,  $C_2$ ,  $C_3$ ,  $C_4$ : High values to improve low-frequency response if required.

### **Circuit modifications**

• The high impedance of the circuit is possible because the p-channel enhancement-mode m.o.s.f.e.t. can still have the drain well beyond the knee of its characteristics if biased with drain and gate at equal quiescent potentials. Provided the a.c. applied to the drain is adequately decoupled by  $R_1C_1$ , the dynamic impedance approaches that of the drainsource  $(R_1 \text{ may be very large and})$ still achieve correct bias). The current mirror approach retains the high dynamic resistance down to d.c. at the expense of a second m.o.s.f.e.t., Fig. (b). Cascode circuits may have the devices in series, sharing the same d.c. current (rather than in parallel as with the complementary stage above). If the input device is a junction f.e.t., the simplest circuits demand only a bipolar transistor with base potential set to a value in excess of the pinchvoltage of the f.e.t. to achieve good cascode performances. The f.e.t. determines the input performance including very high input impedance at low frequencies, while the cascode configuration extends the high impedance to higher frequencies while allowing higher overall gain.

• All f.e.t. circuits are equally possible provided  $Tr_2$  has a higher  $V_p$  than  $Tr_1$ . Transistor  $Tr_3$  is a source follower to isolate the load  $R_2$  from stray and load capacitances, i.e., increasing bandwidth of output circuit.

#### Further reading

FET Circuit Applications, National Semiconductor Linear Applications Handbook, 1972, p. AN32–2. Schade, O. H. Jr. Application note ICAN-6080, in RCA Data Book SSD-203A, 1973, pp. 299–300.

Cross references Set 12, cards 1 & 5. Set 11, cards 5 & 6. Set 7, card 5.



## Set 12: Wideband amplifiers—8

### Wideband amplifiers with t.t.l., r.t.l., d.t.l.



#### T.t.l. gate description

Without the use of negative feedback, the normal transfer characteristic of the t.t.l. gate is designed to provide welldefined logic levels with a highgain transition region between them causing the gate to rest in either the on or off condition. Compared with the static on or off state, the supply current drain increases sharply as the gate is switched through the narrow transition region. Although designed for mediumspeed digital switching applications a t.t.l. NAND gate can be converted into a linear wideband amplifier. By addition of  $R_1$  and  $R_2$ , as shown, the shunt-derived shunt-applied negative feedback has the effect of linearizing the overall d.c. transfer characteristic. eliminating the well-defined logic states, widening the transition region, reducing the gain, and increasing the standing supply current.

Thus, a suitable operating point may be established, with a quiescent supply current of about 19mA, which will allow linear amplification of input signals having levels up to about 800mV pk-pk in the frequency range of approximately 100Hz to 12MHz. Unused gate inputs may be paralleled with the used input rather than being left open-circuit. Alternatively, a logic signal may be applied to the unused input to provide a gating facility on the used input's signal.

#### **Component changes**

Useful range of supply +3 to +5.5V.

Changing R<sub>2</sub> varies gain, shape

Circuit data Supply: 5V, 19mA IC:  $1/4 \times SN7400$ C<sub>1</sub>:  $10\mu$ F tantalum R<sub>1</sub>: 220 $\Omega$ ; R<sub>2</sub>: 560 $\Omega$ V<sub>in</sub>: 50mV r.m.s.



R.t.l. and d.t.l. gate descriptions. A dual NAND/NOR r.t.l. gate, contained in a single 914 package (shown dotted), has Tr<sub>1</sub>, Tr<sub>2</sub> and Tr<sub>3</sub>, Tr<sub>4</sub> connected as a long-tailed pair with  $R_5$  returned to the -Vrail and serving as an approximation to a constant-current Source below. Transistor Tr5 is used as an inverting output stage having a gain of two, with the output at its collector fed back to  $Tr_1$  via  $R_4$  and  $R_3$ . Thus, a non-inverted output is available for signals applied to



 $Tr_4$  base. The d.c. output voltage can be set to zero with the input grounded, by adjusting R<sub>5</sub>. A gain of 10 with bandwidth greater than 10MHz is obtainable with a supply of  $\pm 6V, Tr_{5} 2N3702, R_{1} 640\Omega, R_{2}$  $450\Omega$  (internal to 914), R<sub>3</sub>, R<sub>5</sub>,  $R_6$ ,  $1k\Omega$ ,  $R_4$   $10k\Omega$ ,  $R_7$   $2.2k\Omega$ . Circuit below right Shows a d.t.1. buffer gate connected in the collector circuit of Tr<sub>1</sub> using the former's +V and extender pins. The gate is operated at its logic switching threshold where it provides high gain. Negative feedback is provided by R<sub>2</sub> and  $R_3$  which determine the overall gain. By decoupling a section of  $R_2$ , the gain may be increased without affecting the d.c. operating conditions. A gain of 10, a bandwidth in excess of 500kHz and a load-driving capability of 1 volt into  $300\Omega$  is achievable with a supply of

+5V, IC  $\frac{1}{2}$  × 932, Tr<sub>1</sub> 2N2222, R<sub>1</sub> 120kΩ, R<sub>2</sub> 120Ω, R<sub>3</sub> 1.1kΩ, C<sub>1</sub> 10nF, C<sub>2</sub> 100pF, and C<sub>3</sub> 10 $\mu$ F.

#### Further reading

Kolataj, J. H. Linearize your t.t.l. gates—then build useful circuits with them, in 400 Ideas for Design, Hayden 1971, pp. 2-3.

Klipstein, D. L. Build an operational amplifier from a dual NAND/NOR gate, in 400 Ideas for Design, Hayden 1971, p. 21.

Jones, D. Wideband amplifier uses a single d.t.l. gate, in 400 Ideas for Design, Hayden 1971, p. 23.



### Amplifiers using d.c. feedback pair



Circuit data Supply: +5V, 6mA Tr<sub>1</sub>, Tr<sub>2</sub>:  $1/5 \times CA3046$ C<sub>1</sub>:  $10\mu$ F tantalum R<sub>1</sub>:  $3.9k\Omega$ ; R<sub>2</sub>:  $10k\Omega$ R<sub>3</sub>:  $150\Omega$ V<sub>C1</sub>: 1.5V; V<sub>B1</sub>: 0.67VV<sub>E2</sub>: 0.77VSource e.m.f.: 50mV r.m.s.



#### **Circuit description**

Transistor Tr<sub>1</sub> is connected in the common-emitter configuration and feeds the emitter follower Tr<sub>2</sub> with overall shunt-derived shunt-applied feedback via R<sub>2</sub>, which allows the overall gain and input impedance to be adjusted. If the circuit is supplied from an ideal voltage source the gain will be high and the bandwidth relatively small. Such a source could be heavily loaded when connected directly to the circuit shown above. This is due to the equivalent load on the source of  $h_{ie}$  of  $Tr_1$  in parallel with  $R_2/(1+|A_v|)$ , where  $|A_v| \approx g_m R_{L_1}; R_{L_1}$  being the

equivalent load resistance seen by  $Tr_1$ . The signal generator available had an output resistance of  $100\Omega$  which approximately matched the input resistance of the amplifier. This can be seen to be reasonable assuming  $h_{ie}$  of  $Tr_1$  to be approximately  $2.5k\Omega$  and  $|A_v| \approx 80$  with  $R_{L_1} \approx 3 k \Omega$ . To define the overall gain and input impedance and to provide a wider bandwidth a resistor may be included in series with  $C_1$ . Response shown applies with a 900- $\Omega$  resistance included and the overall gain  $= V_{out}$ /source e.m.f. If it is desired to match the amplifier input resistance to a defined source resistance, this

can be achieved using the circuit as shown above and adjusting the value of  $R_2$ . The integrated-circuit package used contains a five-transistor array, two connected as a longtailed pair. One of the single transistors has its emitter connected to the integrated circuit substrate which must be connected to the most negative supply rail, hence this transistor can be conveniently used as  $Tr_1$ and any other as Tr<sub>2</sub>. A discrete component version of the circuit is also viable.

#### **Component changes**

Useful range of  $V_{ec} \approx +3$  to +15V.



#### **Circuit description**

Another amplifier based on the configuration shown top, and also using an integrated-circuit transistor array is shown above. Transistors are contained in a single CA3018 package with  $Tr_3$  emitter and  $Tr_3$  base internally connected. The circuit arrangement of  $Tr_1$ ,  $Tr_2$  serving as a common-emitter common-collector pair followed by  $Tr_3$ ,  $Tr_4$  in the same

configuration is designed to reduce internal capacitive feedback. The  $Tr_2$ -emitter follower acts as a low output impedance source to drive the  $Tr_3$  commonemitter stage and also provides a low-capacitance high-impedance loading on the  $Tr_1$  commonemitter stage.

All stages are d.c. coupled with two negative feedback paths. Feedback from  $Tr_2$  emitter to  $Tr_1$  base is effective at d.c. and

low frequencies while that from Tr<sub>3</sub> collector to Tr<sub>1</sub> collector functions at all frequencies. Feedback provides stability of d.c. operating conditions and allows a gain-bandwidth tradeoff to be made. When supplied from a 50- $\Omega$  source, a voltage gain of 49dB with a bandwidth of approximately 32MHz is obtainable with a supply of +6V, C<sub>1</sub>, C<sub>2</sub> 1 $\mu$ F, C<sub>3</sub> 470nF, R<sub>1</sub> 3.5k $\Omega$ , R<sub>2</sub> 8.2k $\Omega$ , R<sub>3</sub> 22k $\Omega$ , R<sub>4</sub> 18k $\Omega$ , R<sub>5</sub> 1.8k $\Omega$ , R<sub>6</sub> 2.5k $\Omega$ , R<sub>7</sub> 1k $\Omega$ , R<sub>8</sub> 2.7k $\Omega$ , using input signals in the range  $7\mu V$ to 4mV (r.m.s.). Lower cut-off frequency largely depends on the capacitors used and is about 800Hz with the above values.

#### **Further reading**

RCA Application note ICAN 5296, in Databook SSD-202, 1972, pp. 68/9. Cherry, E. M. and Hooper, D. E. Design of wide-band transistor desired overall gain, bandwidth of input impedance, e.g. with  $R_2$ =47k $\Omega$  and circuit as shown above: overall gain  $\approx 22$  and bandwidth  $\approx 3.4$ MHz. Resistance values may be scaled by a given factor, e.g. increasing values by factor of 10: overall gain  $\approx 28$  and bandwidth  $\approx 1$ MHz.

Maximum source e.m.f. 72mV

r.m.s. with  $V_{ce} = +5V$ .

R<sub>2</sub> adjustable to provide

feedback amplifiers, *Proc.I.E.E.*, vol. 110, 1963, pp. 375–87. Tuil, J. Transistor-equipped aerial amplifiers—wideband amplifiers, *Electronic Applications*, vol. 28, 1968, pp. 75–8.

### Cross references

Set 12, cards 2, 8 & 12. Set 7, card 9. Set 4, card 4.

## Set 12: Wideband amplifiers-10

### Gated video amplifier



#### Circuit data

Supply: ±5V, 8mA IC: MC1445L Voltage gain: 19dB Bandwidth: 60MHz Output attenuation: 60dB with input gated off Input impedance:  $\approx 8k\Omega$ Output impedance:  $\approx 22\Omega$ Input common-mode range:  $\pm 2.5V$ Quiescent output voltage:  $\approx 0.18V$ Gate voltage range: 0.35 to 2.0V

#### **Circuit description**

The long-tailed pair is the key to the operation of this circuit, as of so many linear i.cs. The  $V_{be}/I_{C}$  characteristic of a bipolar transistor is given by  $I_{\rm C} = I_{\rm S} \exp{(qV_{\rm be}/kT)}$ . Transconductance is obtained by differentiating  $I_C$  with respect to  $V_{be}$ , and is found to be proportional to Ic. Thus controlling Ic gives proportional control of the voltage gain of an amplifier provided that the collector load is low enough to allow the full transconductance to be developed. Used with a single transistor an output signal may be gated on and off by such a means, but the shift in d.c. level carries the gate signal through into the output circuit. With long-tailed pairs as shown, Tr<sub>3</sub> sustains a constant total current using common-mode feedback via the emitters of  $Tr_8$ ,  $Tr_9$ . If  $Tr_3$  conduction increases the collector potentials of  $Tr_4$ ,  $Tr_6$  must fall regardless of which the current is channelled through. This fall in potential at each collector is coupled via the emitter followers  $Tr_8$ ,  $Tr_9$  to close the d.c. feedback loop and stabilize

operating conditions. The base of  $Tr_2$  is fixed while that of  $Tr_1$ is normally more positive, ensuring that Tr<sub>4</sub>, Tr<sub>5</sub> are the conducting transistors, i.e. having sufficient gm to give an overall voltage gain of 18 to 20dB. Transistor Tr<sub>2</sub> has negligible current and  $Tr_6$ ,  $Tr_7$ contribute nothing to the output. If the potential at the base of  $Tr_1$  is lowered, either by placing a low direct voltage at the gate input  $(D_1)$ , or by taking the input to the negative supply rail through a resistor, then the currents in Tr<sub>6</sub>, Tr<sub>7</sub> increase at the expense of  $Tr_4$ ,  $Tr_5$ . Hence the signals applied to these differential inputs are gated by a d.c. input. This input may be changed rapidly to allow gating for very short durations or may have intermediate values that allow the inputs to be summed in varying proportions.

#### **Component changes**

IC: MC1445, 1545; similar i.cs with different bias arrangements include Silicon General SG1402/2402/3402. "Discrete" versions based on transistor arrays are also possible but no particular advantages would be expected except possibility of operating at unusual supply voltage/current levels. Balanced modulators/demodulators (MC1496) have similar input stages with free collectors suitable for coupling into tuned circuits.

If the frequency range is more limited, it is possible to adapt analogue multipliers with one input switched between zero and some finite value. This also allows reversal of phase of the output if the control input polarity is reversed.  $V_s: \pm 3.5$  to  $\pm 12V$ .  $R_L:$  At low supply voltages the output quiescent current may be

output quiescent current may be increased by placing external resistors in parallel with the  $5.0k\Omega$  internal resistors. This increases the signal current that may be capacitively coupled into a load, while lowering the output impedance. 10k to  $1k\Omega$ at  $\pm 5V$  supply. Observe maximum dissipation limits of device (dependent on package, ambient temperature, but up to 400mW at < 50°C).



#### **Circuit modifications**

• By applying the mean value of the output to a standard op-amp, the output can be made to provide a sensitive error signal for departures of the mean i.c. output potential from ground. If the op-amp is operated from a higher positive voltage than that required by the wide-band i.c., then d.c. feedback ensures that the positive voltage applied to the i.c. forces the mean output close to zero (depending on the op-amp offset of a few millivolts). This allows d.c. coupling of signal from input to output of wideband amplifier with negligible offset.

• For single-ended supplies the unused inputs are taken to a decoupled potential divider, with the option of a  $50\Omega$  resistance added to define the input impedance of any signal input, e.g. capacitively coupled as shown. The gate terminal then becomes compatible with a t.t.l., c.m.o.s. logic gate sharing a common ground line.



## Set 12: Wideband amplifiers—1



R<sub>1</sub>, R<sub>2</sub>: 10kΩ Slew rate:  $80V/\mu s$ Small-signal bandwidth 14MHz

**Circuit description** 

The bandwidth available from operational amplifiers has been extended by many manufacturers to the level where they can be applied to problems previously requiring speciallydesigned wideband amplifiers. The circuits (a)-(d) use a monolithic i.c. the LM318. showing how manufacturers are able to provide compensation points to modify the highfrequency characteristicstrading in stability margins for increased slew rates, minimum settling times, etc. The internal structure of any particular i.c. at this level of performance is immensely complex and most users will have to treat them as black boxes.

Using hybrid techniques as well as discrete circuitry, amplifiers are available with slew rates in excess of  $1000V/\mu s$  and bandwidths greater than 100MHz. Methods of construction are more costly than monolithic circuits and they are relevant to specialized applications. In general to achieve such very high bandwidths, the open-loop voltage gain has to be restricted to the range 1 to 5000, while the monolithic amplifier above is optimized for best bandwidth consistent with voltage gains > 100,000.

#### Further circuits

• As a standard operational amplifier, circuits commonly used with 741, 301 and similar devices may be adapted to high  $R_1$ : 5k $\Omega$ ,  $R_2$ : 10k $\Omega$ C<sub>1</sub>: 5pF Voltage gain: -2 Comparable figures to (a)

frequency units such as the LM318. Layout and decoupling are important, but high frequency amplifiers, oscillators, filters and the like follow similar configurations, generally with comparable resistance values and reduced capacitances. To combine the wide frequency range with high input resistance the technique shown may be adopted. At very low frequencies the high input impedance amplifier-LM312 or similardevelops its full gain applying its inverted output to the noninverting input of the highfrequency amplifier. The overall characteristic is inverting and the feedback is negative, while the minimal input current of the high-impedance unit permits R<sub>1</sub> and  $R_2$  to have megohm values. At high-frequencies the

integrator configuration of the

LM312 leaves the non-inverting

input of the LM318 virtually

R<sub>1</sub>, R<sub>2</sub>:  $10k\Omega$ , R<sub>3</sub>:  $5k\Omega$ C<sub>1</sub>:  $0.1\mu$ F, C<sub>2</sub>: 5pFShortens settling time to  $1\mu$ s for 0.1% accuracy after 10V step

grounded. The signal appears on the inverting input via the CR network and the feedback remains negative. The two time constants define the cross-over regions the overall gain being sustained to 1MHz. • Non-linear circuits may be extended to higher frequencies, and for example precision halfand full-wave rectifiers operate to an order of magnitude higher in frequency than with 741, etc. At these higher frequencies, Schottky diodes with their smaller forward voltage drop and absence of charge storage are worthwhile alternatives to even high-speed conventional diodes. Component values for the voltage follower mode are

similar to those for the unitygain inverter. Not all amplifiers are compensated for 100%feedback, but may be optimized for higher gains instead.



#### **Further reading**

LM318 data sheet, National Semiconductor Linear Integrated Circuits Databook, 1972, pp. 185–9. Young, R. L. Lift i.c. op-amp performance, *Electronic Design*, vol. 4, no. 21, 15 Feb, 1973, pp. 66–9. Motorola application note— AN276, 14MHz wideband amplifier using the MC1530 op-amp.

Cross references

Set 12, cards 3 & 4.



## Set 12: Wideband amplifiers—12

### Common-base wideband amplifier



Circuit data Supply: +5V, 8mA Tr<sub>1</sub>, Tr<sub>2</sub>:  $1/5 \times CA3046$ C<sub>1</sub>, C<sub>2</sub>:  $10\mu$ F tantalum R<sub>1</sub>:  $3.9k\Omega$ ; R<sub>2</sub>:  $10k\Omega$ R<sub>3</sub>:  $150\Omega$ ; R<sub>4</sub>:  $470\Omega$ Source e.m.f.: 30mV r.m.s. Source res.:  $100\Omega$ 



#### **Circuit description**

The input stage consists of  $Tr_1$ in the common-base configuration with this transistor feeding  $Tr_2$  acting as an emitter follower with overall shuntderived shunt-applied feedback through  $R_2$ . This arrangement is similar to the circuit shown on card 9, except that the source feeds the emitter instead of the base of  $Tr_1$ , indicating that the d.c. feedback pair is a convenient method of biasing a commonbase stage.

Capacitor  $C_1$  prevents any d.c. component present in the input signal from affecting the biasing, or it may be thought of as necessary if the source cannot sink current from Tr<sub>1</sub>. cannot sink current from Tr<sub>1</sub> emitter. Capacitor C2 grounds the base of  $Tr_1$  to a.c. signals. The input impedence of the circuit is low and may be adjusted or matched to a source, by means of  $R_1$  which controls the collector current and hence the emitter current of  $Tr_1$ . This type of circuit is useful for applications where the common-base configuration offers advantages such as small non-linearity of its transfer characteristic and relatively

constant gain with frequency, while providing moderate gain due to the impedance transformation between input and output.

The integrated circuit used is the same as that for card 9 where one transistor has its emitter connected to the substrate which must be connected to the most negative supply rail (0V in this case). Because neither  $Tr_1$  nor  $Tr_2$ has its emitter grounded in the above circuit the substrateconnected transistor cannot be used but its emitter must still be grounded.

#### **Component changes**

Useful range of  $V_{cc}$ : +3 to +15V.

Maximum source e.m.f.  $\approx 37 \text{mV}$ r.m.s. (with  $V_{cc} = +5 \text{V}$ ). With other resistors in the same ratios  $R_1$  may be: (a) raised as high as  $100 \text{k}\Omega$  if low quiescent power is a major factor and maximum bandwidth less important.

(b) chosen to provide maximum gain-bandwidth product in  $Tr_1$ . (c) reduced to about  $100\Omega$  if maximum output current is required.

R<sub>3</sub> may be reduced towards

 $R_1/h_{\rm FE}$  and  $R_2$  increased towards  $h_{\rm FE}$ .  $R_1$ .

#### **Circuit modifications**

• If the signal source can sink direct emitter current of  $Tr_1$ ,  $C_1$  and  $R_4$  may be omitted and the source connected directly to  $Tr_1$  emitter as shown left, the rest of the circuit remaining the same.

• Where d.c. isolation of the source and  $Tr_1$  emitter is required, an alternative input arrangement to the original circuit using a transformer may be adopted as shown centre, again the remaining circuitry being unchanged. This input transformer may be used to match the source to the input impedance of Tr<sub>1</sub> if desired. • If it is required to use a common-base stage that provides an output current which is as nearly identical to the input current as possible, a compound transistor may be used as shown right. The base current of  $Tr_1$  is a very good measure of the extent by which its collector current falls short of its emitter current. By feeding back the base current of  $Tr_1$  to the common-base transistor  $Tr_3$  and adding its

collector current to the input current, the output current more nearly approaches I<sub>in</sub>. This technique also raises the output impedance from  $Z_{out_1}$ to approximately  $Z_{out_1}$ . $h_{tb_1}/(1-h_{tb_3})$  and reduces the distortion from  $D_1\%$  to approximately  $D_1(1-h_{tb_3})\%$ compared with Tr<sub>1</sub> alone.

#### Further reading

Miller, J. R., Solid-State Communications, McGraw-Hill, 1966, chapter 15. Boxall, F. S. Base current feedback and feedback compound transistor, *Semiconductor Products*, vol. 1, no. 5, 1958, pp. 17–24.

Cross references Set 12, cards 2, 5 & 9. Set 7, card 9. 41



1. Both input and output compensation of an operational-amplifier (with 100mA capability) to provide high slew-rates are considered. Fig. 1 (a) shows input compensation, where  $R_{\rm C} = 47\Omega$  and  $C_{\rm C} = 0.01 \mu F$ , for unity gain. Table 1 indicates variations required to these components at higher gains. For gains greater than 5, a single capacitor across the output is adequate for stability with a sacrifice in bandwidth Fig. 1(b). The slew-rates possible are quoted

2. This circuit provides a pulse amplifier with a high voltage capability. The circuit is driven from t.t.l. levels, and can operate at frequencies up to 100kHz. High-voltage transistors Tr<sub>3</sub> and Tr<sub>4</sub> are alternately switched on and off via transistors Tr, and Tr<sub>2</sub>. The bias supply for transistors  $Tr_1$  and  $Tr_3$  is obtained from a 4.5V battery and an optical coupler allows  $Tr_1$  to be switched by the t.t.l. gate. No isolation is required for the lower transistor network which can be driven by t.t.l. directly. When the input is high, Tr<sub>2</sub> is hard on, removing

3. Figs. 3 (a) and 3 (b) are circuits for wideband amplifiers using computer aided design techniques. The first provides an 11dB gain over the range 100-500MHz, and the last provides 14dB gain over the range 100MHz to 1GHz. When two transistors are placed in parallel for medium power (10 to 100mW), equal emitter resistors are included for d.c. balance, and also provide series feedback for the stage. The emitter capacitor C<sub>E</sub> provides a gain compensation. Resistor R<sub>1</sub> is the shunt feedback resistor, with an inductor in series to reduce the feedback at the high frequency end of the response, and thus maintain flat gain. The reactive elements being selected at input and output for matching purposes. The single transistor amplifier



+3.5V +3.5V t.L. input  $Tr_1$   $Tr_2$  2N 2905  $Tr_3$   $Tr_4$  MJ 105  $Tr_1$   $Tr_2$   $Tr_2$   $Tr_2$   $Tr_2$   $Tr_3$   $Tr_4$  MJ 105  $Tr_1$   $Tr_2$  2N 2905  $Tr_3$   $Tr_4$  MJ 105  $Tr_1$   $Tr_2$  2N 2905  $Tr_3$   $Tr_4$  MJ 105 in the lower part of the table and little change occurs for capacitive loads up to 50nF.

#### Reference

Wyland, J. Ideas for Design, Electronic Design 10, May 10, 1975.

Gain	C <sub>C</sub> nF	$R_C$ $\Omega$	Slew rate V/µs	Bandwidth kHz
1	10	47	50	600
10	4.7	100	30	600
100			100	600
5	200		0.75	12
10	100		1.5	24
50	20	_	7.5	120
100	10		15	240

the bias from  $Tr_4$  which is convincingly off because its base-emitter junction will be, reverse-biased by about -1.5V.  $Tr_3$  will be biased on giving a high output voltage depending on power supply level.

#### Reference

Limuti, D. Ideas for Design, Electronic Design 20, Sept. 27, 1975.



is designed for discrete components, but for microstrip and thin-film hybrid amplifiers, it is suggested that a transmission line approach is better, for the necessary inductance values. It is suggested that the design in Fig. 3 (b) can be packaged in a standard TO-8 can.

#### Reference

Steivin, P. Use c.a.d. to optimize broadband amp design, *Electronic Design* 10, May 10, 1974.

## Set 13: Alarm circuits

Rather than dwelling on circuit techniques, the background article brings the concept of data domains to the fore. This interesting approach is an adaptation of that by Malmstadt and Enke, referenced in the article. Card 9, on signal domain conversion, loosely connects with the theme in presenting voltage-tocurrent and voltage-to-frequency converters and their converse.

Transducer circuits included in this set measure pressure and temperature on card 10, displacement, velocity and acceleration on card 12, and light on card 1. Another important group of circuits, for sensing variations in conductance, is applied to flame, smoke and gas detection on card 1, and as a water-level or moisture-sensitive alarm on cards 10 & 11. Most resistance sensors are bridge circuits, of course, & card 2 shows some useful variants, including a bridge/ amplifier combination with both input and output referred to earth. The security alarm of card 11 is one application of the bridge. Cards 4 & 5 are of simple hysterectic level sensors.

Information on driving filament lamps and relays is provided on card 8, while card 4 gives various output stage configurations. Further load-driving circuits can make direct use of the 555—see cards 3 & 5, which also detail other uses of the i.c.

Flame, smoke and gas detectors 1 Bridge circuits 2 Time delay and generator circuits 3 Level sensing and load driving 4 Applications of 555 timer 5 Frequency sensing alarm 6 Digital alarm annunciators 7 Filament lamps and relays 8 Signal domain conversion 9 Pressure, temperature and moisture-sensitive alarms 10 Security, water level and automobile alarms 11 Electromechanical alarms 12

## **Alarm circuits**

In this topic the viewpoint changes. Previously in Circards, circuits have been described as separate entities, with the articles laying a foundation, and the cards showing the practical alternatives. The dilemma is that the title used here, though commonly applied to the topic, can be misleading. This is because so many "alarm circuits" have several identifiable subsections each of which can be readily classified under headings such as those of previous series e.g. Schmitt trigger and astable circuits. Even heading this article "Alarm systems" might confuse since it could convey the idea of a conglomeration of alarms.

A typical alarm arrangment is shown in block diagram form in Fig. 1. An alarm signal is required from some output transducer when the signal from an input transducer indicates a particular fault condition. The intervening blocks are required to process the signal representing the fault condition, to detect a particular voltage/current level which is an analogue of the input parameter, and to deliver power to the output transducer. Before considering each of these sub-sections individually, there is a general principle which can be very helpful in considering such systems—the concept of data domains.

#### **Data domains**

Information on physical systems is obtainable in the form of physical quantities such as force, distance, energy. The required information may often have to be obtained by monitoring some more complex property involving a function of the more common parameters. In few cases is the information in such a form that it can be conveyed and displayed directly. It has to be transmitted through some medium or series of media first. To the practising engineer the medium is not the message but rather the barrier that hinders the appearance of that message.

In passing through these media the information takes on new forms or dimensions. For each new set of dimensions we may postulate a domain in which the data exists. Within that domain there may take place further conversions without going outside the domain. Thus the first main division is between inter-domain and intradomain conversions. The division is arbitrary since the selection of domain interfaces is arbitary. One possible division is discussed below but readers may have their own ideas on this.\*

For the purpose of this article, the domains are determined by the following considerations: is the data (a) continuous or discontinuous and (b) instantaneous or not. These conditions applied to electrical phenomena provide us with four domains. In addition there is the much larger physical domain (P) containing all non-electrical data. For other disciplines it is this domain that would be sub-divided into more convenient packages, as for example the heat/light/sound divisions of physics. The electrical domains are thus

	lnstan- tan <del>c</del> ous	Non- instan- taneous
Continuous	A	t
Discontinuous	$D_{\rho}$	D <sub>s</sub>

An A or amplitude function e.g. voltage/ current/resistance, is an electrical signal precisely and continuously related to some other function which may or may not be within the A domain itself. It has an instantaneous value which is a measure of some property of the unknown. An example is the electrical resistance thermometer where the resistance (A domain) is a continuously variable function of the instantaneous temperature (P domain) i.e. a P-A domain conversion occurs.

A t function is also continuously variable. but to represent the data a finite time must elapse i.e. it is not instantaneous. This property should not be confused with the finite delays imposed by the physical limitations of systems, which prevent the instantaneous change in an A function. A t function will have corresponding delays in responding to changes in the data, but even with a fixed input requires a finite time to complete the conversion. An example is an oscillator whose frequency is proportional to a voltage. To determine that frequency at least one period must elapse (often many) and the data conversion is non-instantaneous. Such a voltage-controlled oscillator is performing an A-t conversion, the amplitude of the output waveform being irrelevant as all the data resides in the time-function.

Conversions may take place through more than one domain, and the shortest route in a system is not necessarily the best. If we wish, for example, to convert from temperature to frequency (a *P-t* conversion) we can do so by constructing an oscillator whose frequency is temperature dependent<sup>†</sup>, or we can use a thermocouple to generate a direct voltage that, amplified, controls a v.c.o. The latter can be considered as a P-A-A-t conversion with the voltage amplifier being an A-A converter i.e. input and output both existing in the amplitude or Adomain. Better linearity of frequency against temperature could be achieved in this second approach.

Where the data is required in digital i.e. discontinuous form, a similar distinction can be made as to whether the data appears simultaneously at input and output (within the delay constraints mentioned above) or whether a finite time is required for the data conversion. The two categories resulting are the parallel and series modes respectively  $(D_{a} \text{ and } D_{b})$ . They may also be thought of as a spatial and temporal ordering of the data-a pulse train representing the data in serial form conveys that information correctly regardless of the frequency if the order pattern is correct. In a digital voltmeter the data might be converted into serial form following an initial voltage-tofrequency/time conversion, while it would be stored and displayed in parallel form. The data domain conversion pattern would then be  $A-t-D_s-D_p$ . Within each domain, there may be a

Within each domain, there may be a great variety of possible forms for the data, and multiple conversions can and do take place. Even a "simple" amplifier may have individual stages best viewed as V-I and I-V converters, while a voltage amplifier can be regarded as a V-V converter.

#### Transducers

These are the interfaces between the physical (P) and electrical domains  $(A, t, D_p, D_s)$ . The range and variety is too large to cover in such an article as this, but some obvious

<sup>†</sup>This can also be considered as a hidden form of P-A-t conversion since the temperature affects some A parameter such as R, C etc.



Fig. 1. Typical alarm circuit with input and output transducers.

<sup>\*</sup>This approach was prompted by the excellent book "Digital Electronics for Scientists" by Malmstadt & Enke (Benjamin) which proposed a slightly different division.

examples are worth discussing. If an electrical conductor is subject to temperature variations its conductivity will vary. For metallic conductors the temperature coefficient of resistance is normally positive and the characteristic is sufficiently welldefined to allow precision thermometers to be based on it (platinum resistance thermometers). For semiconductors the coefficient may be negative or positive and of much greater magnitude though generally less well defined. This makes devices such as thermistors, which depend on this property, particularly useful in alarm circuits as a relatively sharp transition takes place in the resistance value and switching of a load is simplified.

If such a resistance which depends on a physical parameter is incorporated in a bridge circuit (Fig. 2) then by suitable selection of the other resistors the critical resist-



Fig. 2. Bridge circuit for including a sensing resistor.

ance value of the transducers may be made to correspond to the bridge balance point. Any high gain differential input amplifier may be used to detect this change of polarity about the balance point, providing a large output swing. Addition of positive feedback provides hysteresis, minimizing the output switching that would occur from noise or other stray input signals, when close-tobalance comparators or general-purpose operational amplifiers may be satisfactory in such applications.

Other physical parameters may affect the resistance of particular conductors and semiconductors. For example a polycrystalline film of cadmium sulphide in darkness has a very high resistance (>1M $\Omega$ ), while exposure to sunlight may drop that resistance to a few hundred ohms. Where the changes are as extreme as this, the variable resistance could simply be placed in series with a supply voltage and the load to provide a direct if somewhat imprecise alarm.

Other semiconductors when exposed to particular gaseous impurities show similar large variations in resistance and are now used in gas and smoke detectors, though they require a separate power source to raise their operating temperature. Even the basic resistance thermometer mentioned above can be adapted to detect other physical parameters; for example the flow of air or other fluid across a heated filament removes the heat more rapidly causing the resistance to fall. Thus detection of fluid velocity is a possibility.

Other transducers give a voltage or current that is a function of a physical parameter; the e.m.f. of a thermocouple and the current flow in a reverse-biased photodiode are examples. Yet others may involve the variation of electrical parameters such as capacitance or inductance, coupling between coils, etc. In such cases a common alternative to the bridge technique, still viable with the substitution of a.c. drive to the bridge, is to make the frequency of an oscillator depend on the variation of the reactance used, and follow the oscillator by some form of frequency-sensitive switch.

Following the input transducer the signal may need to be amplified, filtered or modified (domain conversion of some form) in some signal-processing stage prior to being fed to a level-sensing stage. In some cases the two functions can be combined, as operational amplifiers having very high gain can suffice. If the output of the levelsensing device is insufficient in magnitude to drive the required load then a further power stage may have to be substituted (Fig. 1).

Additionally it may be required to cause this output signal to be an audible tone or an interrupted voltage for flashing a lamp. Either case could require an astable oscillator or similar form of generator (Fig. 3). A monostable circuit may be interposed to delay the onset of the alarm output for some period after the appearance of the fault signal and logic gating would be added in more complex systems to generate alarm outputs that depend on a particular combination of input parameters.

Thus most alarm systems can be broken down into simpler blocks and the block diagram of a burglar alarm could be identical with that for a circuit intended to sense icy conditions on a road. By appreciating and making use of this principle, it is often possible to make very economical designs of alarm circuits by adapting the best individual blocks from previously published alarm circuits. The major design problem is then that of making the blocks compatible in respect of supply voltage load requirements and the like.





## Set 13: Alarm Circuits—

### Flame, smoke and gas detectors



**Component** values Tr<sub>1</sub>: TIS34 Tr<sub>2</sub>: BC126 Tr<sub>3</sub>: BC125 Tr<sub>4</sub>: BC125 Tr<sub>5</sub>: BC126 **D**<sub>1</sub>: 1N914 D<sub>2</sub>: 1N4002  $C_1$ : 1nF $\mathbf{R}_1, \mathbf{R}_2: 15 \dot{\mathbf{M}} \Omega$  $R_3: 2.2k\Omega$ **R**₄: 12kΩ

 $R_s: 6.8k\Omega$  $R_6: 1.5k\Omega$  $R_7, R_9: 12k\Omega$ R<sub>8</sub>: 820Ω  $R_{10}$ : 15k $\Omega$  $R_{11}$ : 2.7k $\Omega$  $R_{12}: 4.7 k\Omega$  $R_{13}$ : 22k $\Omega$ Semiconductors not critical but TIS34 may need selection because of parameter spread.

#### Flame detector

A flame offers a low conductance path to ground. In series with  $R_1$ ,  $R_2$ , that conductance defines a range of potentials on the gate of  $Tr_1$ , that leaves the emitter of  $Tr_2$  at a high enough potential to keep  $D_1$  out of conduction, but not so high as to bring Tr<sub>4</sub> into conduction via R7. Hence  $Tr_3$ ,  $Tr_5$  conduct holding on the relay-interlocked with the supply for fail-safe operation. If the flame is extinguished  $Tr_1$ gate goes high, driving Tr<sub>4</sub> on via  $Tr_2$ ,  $R_7$ . This removes the

drive from Tr<sub>3</sub>, Tr<sub>5</sub> and the relay. A short circuit to ground at the input reduces the base potential of Tr<sub>2</sub> bringing D<sub>1</sub> into conduction and cutting off  $Tr_3$  and hence the output. The mid-section of the circuit offers a window action with the relay being held on for a restricted range of flame resistances, higher and lower values giving drop-out. The resistance is high requiring a high input resistance buffer; the output is conventional.



Tr<sub>1</sub>: LS400  $R_2, R_9: 100k\Omega$ Tr<sub>2-4</sub>: 2N712  $R_3: 15k\Omega$ Tr<sub>5</sub>: C106F R<sub>4</sub>: 470kΩ  $R_1: 1k\Omega$  $R_{5-7}, R_{10}: 10k\Omega$ 

C1: 16µF  $C_2, C_4: 22nF$  $C_3: 0.1 \mu F$ 

C<sub>5</sub>: 50µF C<sub>6</sub>: 4.7nF Transistor types not critical.

#### Smoke detector

When detecting the interruption of light by smoke, to avoid the effects of ambient illumination, etc., the light beam may be chopped at source and the resulting a.c. from Tr<sub>1</sub> above used via buffer Tr<sub>2</sub> to trigger the monostable circuit around Tr<sub>3</sub>,  $Tr_4$ . This prevents the potential

applied to R<sub>10</sub> from rising sufficiently to fire the thyristor. If the load is a horn having an interrupter switch in series with its coil, the thyristor can cease conduction on removal of the gate drive (alternatively a.c. drive to the load would be required).



Tr<sub>1</sub>: BC126 Tr<sub>2</sub>: TIS43  $C_1: 0.22 \mu F$ LS: 8 to  $80\Omega$  $\begin{array}{l} R_1: 470 \Omega \\ R_2: 3.3 k \Omega \end{array}$ R<sub>3</sub>, R<sub>4</sub>: 10kΩ R₅: 100kΩ  $R_6, R_7: 1k\Omega$ 

#### Gas detector

A particular gas-sensor (TGS from Figaro Engineering, Shannon, Ireland) has two fine wires embedded in a semiconductor. One is used to heat the material, with the resistance between it and the second being reduced on-the absorption of deoxidizing gas or smoke. The sensor is sensitive to concentrations of < 0.1%, with resistance falling from many tens of kilohms to as low as  $1k\Omega$  at high gas concentrations. Response is non-linear and with a recovery time in excess of one minute. Bridge unbalance is

detected on M<sub>1</sub> and though repeatable has to be interpreted qualitatively unless special calibration procedures are available. When the unbalance brings Tr<sub>1</sub> into conduction, C<sub>1</sub> charges until the unijunction Tr<sub>2</sub> fires and the cycle recommences. The audible note in the loudspeaker rises from a succession of clicks to a continuous tone as the gas concentration increases. A Schmitt trigger would allow relay drive, while the audible alarm could be transferred to the flame-detector circuit, for example.

### **Bridge circuits**



Components ICs: 741, V<sub>s</sub>  $\pm$ 15V R<sub>1</sub> to R<sub>4</sub>: 10k $\Omega$ , R<sub>5</sub>: 1M $\Omega$ Bridge voltage: 1.5V (Fig. 2)

#### **Circuit description**

Three bridge configurations are shown. In each case the bridge is composed of four resistors,  $R_1$  to  $R_4$ , and the circuits are basically Wheatstone bridges with balance occurring for  $R_1/R_2 = R_3/R_4$ . Substitution of impedances  $Z_1$  to  $Z_4$  would leave the balance requirements unchanged, and other variants such as the Wien bridge can be produced. For resistive elements it may be possible to supply the bridge and amplifier from a common d.c. supply and a high-gain op-amp detects departure from balance. A small amount of positive feedback via  $R_5$  helps reduce jitter in the output when close to balance, but gives hysteresis to the balance sensing.

• If a separate supply is required for the bridge, one bridge balance point may be grounded, removing the need for high common-mode rejection for the amplifier (Fig .2) The errors in all these circuits include voltage offset of the amplifier, 1 to 5mV for untrimmed general-purpose op-amps, and input currents/ offset, 10nA to 1µA for conditions as before. For balance

detection to within 0.1% this implies bridge voltages in excess of 1V and currents of up to 1mA.

• By opening the bridge and embedding the amplifier in the network as shown, balance is achieved for the same relationship between the resistances, but with input and output both with respect to ground. Fig. 3 has an output that is a linear function of the departure of R<sub>2</sub> from the balance condition (R1, R3, R4 assumed constant as reference resistors). For d.c. applications the input may be one or other of the supply voltages. In all cases best sensitivity is achieved for  $R_1/R_2 \rightarrow 1$ . If the resistor whose value is being sensed has to have a low resistance, power wastage is avoided by keeping the other pairs of resistances high.

• Another method of achieving input and output as groundreferred signals, is to use an amplifier with push-pull outputs and single-ended input. A simple case is the single transistor, Fig. 4, where the power supply, if properly by-passed, closes the bridge when used for a.c. measurement/sensing. The example shown would pass all frequencies except the notch frequency by  $1/2\pi RC$ , though with appreciable attenuation near the notch.

• For many purposes, the availability of a centre-tapped supply provides a "phantombridge" action. If the ratio of positive to negative supplies remains constant then taking one input of the sense amplifier to the centre-tap leaves only a half-bridge externally. Used for example with photodiodes, the output voltage is proportional to the unbalance currents in the diodes, i.e., to the degree of unbalance in the illumination of the diodes. Because the diodes act as constant-current devices the circuit Fig. 5, is much more tolerant of drift in the centre-tap than for purely resistive elements. The negative feedback gives a linear outputunbalance characteristic. Reversal of the amplifier input terminals would give positive feedback, introducing a switching action and hysteresis as in the first diagram.

• Some i.cs have internal potential dividers which can effectively form part of a

bridge. The 555 timer, for example, has its two comparators tapped at 1 and 1 of the supply voltage via a resistor chain with very good stability to the ratio of their values; the absolute values are not important for such an application (Fig. 6). The lowerthreshold detector ("trigger") when held high prevents any output change (input 1 is assumed high) regardless of the status of the reset terminal. The reset terminal regains control only when the trigger input falls below the level accurately defined by the potential divider. With the trigger taken from an external potential divider containing the required sensing element the bridgebalance sensing can be obtained.

#### Further reading

Markus, J. (ed.), Bridge circuits, in Electronics Circuits Manual, McGraw-Hill, 1971, pp. 84-9. Graeme, Tobey & Huelsman, Operational Amplifiers, McGraw-Hill, 1971.

Cross references Set 1, cards 9 & 10. Set 9, cardr 1 & 11. Set 13, cards 1 & 3.



## Set 13: Alarm Circuits-2

### Time delay and generator circuits





#### **Circuit description**

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An i.c. such as the 555, with internal comparators driving a set-reset flip-flop offers great flexibility in the design of alarm systems. With pin 2 high, the capacitor is held low via pin 7. A negative-going edge on 2 allows  $R_1$  to charge  $C_2$  until the potential on 6 passes  $2V_{\rm S}/3$ , when the original state is restored.

• Linking the inputs of the two comparators (2 and 6) to the discharge path (7) causes the potential at the common point to cycle between  $V_{\rm S}/3$  and  $2V_{\rm S}/3$ , set by an internal Potential-divider (Fig. 2). For both inverters can use very highcircuits the output has switching characteristics comparable to a t.t.l. gate because of a similar totem-pole output stage. An audible alarm is available by connecting a loudspeaker  $(3-25\Omega)$  between V<sub>s</sub> and pin 3. If  $V_s$  is +5V, the on/off condition of the alarm may be controlled by driving pin 4 from the output of a t.t.l. gate. • An astable can also be constructed by feedback from the output to the paralleled comparator inputs. When the

output is high,  $C_1$  is charged (Fig. 3) through  $R_7$  until the upper threshold is passed; the output switches low and  $C_1$  is discharged until the trigger value set by pin 2 is passed. Timing is set by the less welldefined output amplitude, and the frequency is less stable than the basic circuit. Addition of R<sub>2</sub> varies mark-space ratio.

If the reset terminal 4 is coupled to an RC network (Fig. 4), then a time-delay can be introduced at switch-on, before which firing of the circuit as a monostable cannot be achieved.

A monostable using c.m.o.s. value resistors, giving time delays of > 1s with capacitors of  $< 1\mu$ F. In Fig. 5, a shortduration excursion of the input from  $+\ensuremath{ \mbox{to ground sets the}}$ output to zero for the monostable period (about 3s) because the output of the first inverter is high, as is the input of the second until R<sub>2</sub> can pull the gate down by charging  $C_1$ . The high impedance makes such monostables useful as touchoperated circuits. • A related astable circuit Fig. 6 shows an additional resistor R<sub>1</sub> which isolates  $C_1$  from the rapid charge/discharge imposed by the gate protection diodes in both these circuits. The resistor improves the timing stability. • The output stage of an astable/monostable circuit is important where high voltage/ current/power is required. For the 555 timer, the output stage is similar to the typical t.t.l. output (Fig. 7). but with a Darlington-connected top section. The positive output is thus at least 1V below supply while the low output can be to within 0.1V of ground at low currents. Above 50mA the voltage drops may reach 2V and 1V respectively.

• For some applications the open-collector output of t.t.l. devices such as SN7401 gives convenient driving of loads, while other devices such as SN7406 will withstand collectoremitter voltages of up to 30V. (Fig. 8).

#### Further reading

Three articles, by Robbins, Orrel and De Kold, in Electronics, 21 June, 1973, pp. 128-32.

Application note for XR-2556 timing circuit, Exar, 1973.

Cross references Set 3, card 9. Set 13, card 5.



## 

#### **Circuit description**

The basic level-sensing circuits shown may be used with or without positive feedback, to obtain an output change as the input passes a defined level or levels. For  $R_2 \rightarrow \infty$ ,  $R_1 \rightarrow 0$ , amplifier gain determines the range of input voltages for which the output is not switched hard to one or other extreme. (Typically 1 to 20mV for comparators, required to operate at high speeds; 0.1 to 5mV for op-amps where accuracy of level-sensing makes their slower operation an acceptable penalty.) Hysteresis introduced by positive feedback allows the circuit to latch into a final state after the first excursion through a given level, provided the input cannot reverse its sense sufficiently to pass back through the other switching level. These circuits can thus perform the combined functions of level-sensing and set-reset action required in many alarms if for example the signal is a positive-going voltage initiating the set action, while the reset action is a negativegoing pulse over-riding the

former, e.g., a resistor taken from the non-inverting input to the negative rail.

An adaptation of the output stage shown in Fig. 5 gives an output when the p.d. across either  $R_1$  or  $R_2$  exceeds about 0.6V. In the former case this corresponds to a positive input voltage defining sufficient positive supply current via R<sub>3</sub>, i.e.,  $V_{in}R_1/R_3 \approx 0.6V$ . Similarly a negative input voltage switches the output via Tr<sub>1</sub>. The switching action is not particularly sharp as it uses only the gains of the transistors. A standard window comparator gives sharper switching but requires two amplifiers/comparators and still

requires an additional transistor if an output swing comparable to supply voltage is required, e.g. for efficient switching of lamps relays, etc. particularly at higher currents.

• A previously-described output stage (set 2) gives push-pull drive using one op-amp as driver. Resistors  $R_1$ ,  $R_3$  are selected to keep  $Tr_1$ ,  $Tr_2$  out of conduction in quiescent state. The op-amp is



used in any of the sensing/ oscillating modes that result in p.ds across  $R_3$  sufficient to drive  $Tr_1$ ,  $T_2$  into conduction. Either may be used alone for driving lamps, relays, or the circuit as shown may be capacitively coupled to a loudspeaker for a.c. power drive.

• An output stage using a bridge configuration requires antiphase switching at the inputs, but gives a load voltage whose peak-peak value is twice the supply voltage. This is equally applicable to audio alarms or to driving of servo systems for which it was designed.

• Complementary m.o.s. buffers may be used to drive complementary output transistors as shown and with the aid of an additional inverter a similar stage provides a bridge output. The transistor base current is limited to a few milliamp res but in all these output spikes may occur during the output transitions. Diode protection against inductive voltage spikes as in Fig. 5 should be used for loudspeaker, relay and solenoid loads. ● Any of the output transistors may in principle be replaced by the compound transistor pairs if higher peak currents are needed. To reduce the above requirements it is worth considering the use of f.e.t.

devices as the input transistor

stages, short-duration current

Set 13: Alarm Circuits-4

#### Further reading

of the pair.

Electronic Circuits Manual (Markus, McGraw-Hill, 1971): Main circuits—pp. 1-6; lamp control circuits—pp. 344-9; trigger circuits—pp. 889-907. Linear Integrated Circuits Handbook, Marconi-Elliot, pp. 165-170. Industrial Circuits Handbook, SGS-Fairchild, pp. 6-13.



## Set 13: Alarm Circuits-5

## Applications of 555 timer





#### Typical performance

IC: NE555V (Signetics), V<sub>s</sub> +10V. R<sub>1</sub>: 2.2k $\Omega$ , R<sub>2</sub>: 10k $\Omega$ . k=0.6, D<sub>1</sub>: 5.6-V Zener diode. Upper set point: 5.7 ( $V_z$ ).

#### **Circuit description**

The 555, designed as a timing circuit with either monostable or astable operation, has internal circuit functions that allow it to be used for many other purposes. In alarm systems, the power output stage that permits currents of either polarity of up to 200mA (though 50mA minimizes voltage losses) means that lamps and relays can be driven quite readily. When used as an astable circuit the output square wave can be applied to a loudspeaker to give an audible alarm, while a voltage fed to the control terminal modulates the frequency for warble or two-tone effects. As a monostable circuit it can be used to provide delays from microseconds to minutes, allowing, for example, a warning alarm to be held for a defined period of time after the appearance of the condition being detected. In such cases the condition (closure of a switch in a burgular alarm for example) is converted into a negativegoing pulse, applied to the trigger input. A further application for the device involves the controlled hysteresis provided by the two comparators biased from an internal potential divider. With  $V_1 > V_{ref1}$  the output is driven negative via the flip-flop which ignores any further excursions of  $V_1$  about  $V_{ref1}$  in either sense. When  $V_2$  falls below  $V_{ref2}$  the flip-flop is reset, the output going positive. In the astable

Lower set point: 4.75V ( $V_z/2k$ ). Output swing: 9V for  $R_L \ge 250\Omega$ . If  $R_1$ ,  $D_2$  omitted,  $V_{ref1}=2V$ ,  $V_{ref2}=V$  and set points become 2V and V/k.

circuit  $V_1 = V_2$ ,  $V_{ref_1} = 2V_S/3$ ,  $V_{ref_2} = V_S/3$  and the capacitor is charged and discharged between  $V_S/3$  and  $2V_S/3$ .

### Component changes

IC: Motorola MC1455. Separate comparators could be used with independent reference voltages or a single comparator with hysteresis defined by feedback-see Series 2.  $V_{\rm S}$ : 4.5 to 18V. At low voltages the saturation voltages at the output may not allow adequate drive to electromechanical/ filament lamp loads.  $R_1$ ,  $D_1$ : Any network to provide constant voltage at control input. Voltage may be to within 1V of common line or positive supply, but for optimum performance should be close to  $2V_8/3$ .  $\mathbf{R}_2$ : 1k to 1M $\Omega$ . At low values, excessive loading of source; at high values inaccuracies due to threshold current of up to 0.25µA.

#### **Circuit modifications**

Use as battery charger illustrates method well (Fig. 1). Upper threshold when  $k_1 V_L =$  $V_z$ ; lower threshold when  $k_2 V_1 = V_z/2$ . When upper threshold is exceeded output at pin 3 reverse-biases diode D<sub>2</sub> and battery discharges into load when present. As voltage  $V_L$ falls below lower threshold, voltage at pin 3 rises and charges battery through limiting resistor R<sub>2</sub>. Hysteresis may be reduced towards zero for  $V_z/k_1 \rightarrow V_z/2k_2$ . To increase hysteresis, the potential at pin 5 may be reduced following a transition through the upper threshold. This may be done as in Fig. 2 by using output pin 3 via a diode-both thresholds are varied if the diode is replaced by a resistor.

• The increased swing simplifies the triggering of a following 555 used as a Schmitt trigger, as the capacitor voltage in Fig. 2 can approach zero. Complete alarm systems can be based on such circuits combining level sensing, time delays and waveform generation, as well as audible alarms.

#### Further reading

Four articles, by De Kold, McGowan, Harvey & Pate, in *Electronics*, 21 June 1973, pp. 128–32.



## Set 13: Alarm Circuits-6

### Frequency sensing alarm



#### **Circuit description**

The circuit is a monolithic m.o.s. i.c. which uses external RC elements to fix the frequencies at which the circuit provides a switching action. It does so via two separate switching times defined by  $C_1R_1$  and  $C_2R_2$ , as from a pair of monostable circuits with the second time interval being initiated at the end of the first. The input may be a repetitive signal of arbitrary waveform, provided the amplitude is in excess of 100mV pk-pk (though it should not exceed 20V pk-pk). Internally this is presumably squared by a Schmitt type of circuit to trigger the monostables. Three distinct conditions may exist; if the period of the received signal is t=1/f and the two delays are  $t_1 = k/C_1R_1 t_2 = k/C_2R_2$ , then  $t < t_1, t_1 < t < t_1 + t_2, t > t_1 + t_2$ These conditions are distinguished by additional internal circuitry that allows sensing of frequencies above a given datum or within a given band with a switched output that can be made to latch on or off, toggle at a lower frequency (f/20), and hold on during signal failure or for temporary interruptions of the signal. The upper frequency in the band mode or the datum in the datum mode is set by  $t_1$  and the lower band-frequency by  $t_1 + t_2$ . The circuit provides frequency-sensing function similar to comparators Schmitt-triggers and windowcomparators.

#### Component changes

Vs: -12 to -22V some samples operate with reduced accuracy down to -8V.

Vin: 0.1 to 20V pk-pk. freq. set points: 0.01Hz to 50kHz.

response time: within 5 to 10 cycles of receipt of correct frequency.

 $\mathbf{R}_1, \mathbf{R}_2$ : 100k to 1 M  $\Omega$ .

 $C_1, C_2$ : 250pF to 1 $\mu$ F.

 $C_3$ : 10nF to 1 $\mu$ F (not critical).

#### **Circuit modifications**

• As the lower frequency in the band mode is affected by time constant  $C_2R_2$  in the original circuit while the upper frequency is not, variation of  $R_2$  increases the band by variation of its lower bound only. For  $C_1=C_2=C$ , variation in the tapping point of  $R_C$  in (a) at left leaves the sum of the time constants unchanged at  $(R_A+R_B+R_C)C$ , i.e., it is the lower frequency that remains constant while the upper frequency is changed. • Variation in both frequencies

• Variation in both frequencies while retaining a reasonably constant ratio of  $f_2: f_1$  (the

**Circuit modifications** 

(8)

(a)

#### Typical performance

IC: FX101 (Consumer Microcircuits) -12V supply, -3mA + load current.  $V_{in}$ : 250mV pk-pk to pin 1. R<sub>1</sub>, R<sub>2</sub>: 470k  $\Omega$ . C<sub>1</sub>: 22nF, C<sub>2</sub>: 10nF, C<sub>3</sub>: 0.1 $\mu$ F. Ground pins: 2, 3, 9. Output on for: f > 150Hz ( $f \approx 1/0.6C_1R_1$ ). Pin 1 signal input. 2 grounded, holds switch state during signal loss. open, switch off. ground via 'C', switch off after signal break of

equivalent of a constant Q), can be achieved by varying the common bias applied to the resistors. If strong dependence on supply voltage is to be avoided the bias voltage should be supply-proportional as in (b). Constant-current sources allow linear control of period against a separate reference voltage, which may be supplyproportional (centre). Filament lamps may be driven via an additional transistor, currents up to 100mA or so being provided by circuit on right. Direct drive of reed relays, l.e.ds is possible though current is marginal.

#### Further reading

Volk, A. M. Two i.c. digital filter varies passband easily, *Electronics*, 15 Feb. 1973, p. 106. McKinley, R. J., Versatile digital circuit filters highs, lows or bands. *Electronics*, 21 June 1971, p. 66. FX101: Consumer Microcircuits data sheet D/026. **Cross references** Set 1. cords 6 & 7

Set 1, cards 6 & 7.

(ь)





## Set 13: Alarm Circuits-7

### Digital alarm annunciators



#### **Circuit function**

It is assumed that a fault condition is the opening of relay contact RL<sub>1</sub>, though any other sensor that maintains the NAND-gate input terminal at a low ('0') level is adequate. A fault will turn off a "safe' green light and illuminate a 'danger'' red light, and operate an audible alarm. When the "recognize" push-button is depressed, the red light stays on, but the alarm is silenced. When the fault clears, the alarm is restarted, the green light comes on and the red light goes off. The "recognize" button is again pushed to reset circuit to its normal state.

#### **Circuit** operation

Consider the circuit in its normal state where inputs R, T and F are at zero volts (or binary zero), i.e. R=T=F=0and X=0 (X=1), Y=0 ( $\overline{Y}=1$ ) and hence LED<sub>1</sub> is energized (green) and LED<sub>2</sub> (red) is off.

#### **Circuit modification**

As X,  $\overline{X}$ , Y,  $\overline{Y}$  are available, the exclusive-OR function of A can be obtained as shown.



If a fault occurs,  $RL_1$  opens, F goes high (or binary one), i.e. F=1, causing X=1 ( $\overline{X}=0$ ), but the state of Y (and  $\overline{Y}$ ) remains as before. Hence A=1, and triggers audible alarm. Pushing the recognize button causes R=1, and as F=1, T=0, then Y=1 ( $\overline{Y}=0$ ), but X does not change. LED<sub>1</sub> remains on, but A=0, and alarm stops. This state will be maintained until the fault is cleared.

When the fault is cleared, R = F = T = 0, Y does not change, but X=0 (Y= $\overline{X}=1$ , X= $\overline{Y}=0$ ). Hence LED<sub>2</sub> is illuminated, A=1, and the alarm operates.

Final recognition of the fault clearance is obtained from R=1, which will return circuit to its normal state, i.e. for R=1, F=T=0, Y=0 and X=0. Depression of the test button will check LED<sub>1</sub> and the alarm, when started from normal state with LED<sub>2</sub> on.





#### **Circuit description**

Complementary-m.o.s. devices may be used in the circuit above to minimize stand-by power consumption.

Normal safe condition obtains with L=A=0,  $\overline{F}=1$ . When the fault-switch closes,  $F\rightarrow 1$  and since L is already low,  $X\rightarrow 0$ . Memory circuit IC<sub>2</sub>, IC<sub>3</sub> does not change. Also since  $\overline{F}=0$ ,  $Y\rightarrow 1$ , and hence  $\overline{A}$  is forced to zero, therefore A=1. This transition may be used to switch an audible alarm. Simultaneously the oscillator gate is opened which will cause lamp flashing at a rate determined by the astable frequency. If the fault is rectified, the alarm condition is maintained until the clear button is pressed causing C to below. Hence  $L \rightarrow 1$ , and will latch in this condition via memory circuit IC<sub>2</sub> and IC<sub>3</sub>. Also  $\overline{L}=0$ , thus A=0, this condition being maintained via IC<sub>4</sub> and IC<sub>5</sub>, and the alarm is silenced.



**Circuit description** Arrangement right allows detection of first-fault occurrence from three sensors  $S_1, S_2, S_3$ , this number being restricted by the number of inputs available per NAND-gate. Outputs  $Q_4, Q_5, Q_6$  are set to zero when the reset button is depressed. The  $\overline{Q}$  output of each flip-flop is applied to the other

two NAND gates, but not to the one associated with itself. Hence two of the three inputs of each gate are high. If  $S_2$  closes, for example, IC<sub>2</sub> output goes low, and this negative-going edge being applied to IC<sub>5</sub> preset terminal sets  $Q_5=1$  (and hence  $\bar{Q}=0$ ). Therefore IC<sub>1</sub> and IC<sub>3</sub> are now inhibited and cannot respond to a fault condition.

### Filament lamps and relays



Filament lamps are widely used as visual alarm indicators and often connected in the collectoremitter circuit of a bipolar transistor that is switched on and saturated under alarm conditions. These lamps have a positive temperature coefficient of resistance with a large difference of resistance between the cold and hot states-see graph 1 which is typical for a 6V, 100mA panel lamp. When switched on across a voltage source, a large current surge flows in the lamp, and switching transistor, which then decays exponentially to its normal or rated value in the hot state. This surge may be ten times the rated current, or even higher, shortens the life of the lamp, may destroy the switching transistor or blow the power supply fuse. graph 2 shows the typical initial surge current characteristic of a 6V, 60mA panel lamp having a thermal time constant of about 2ms

When lamps are used as flashing alarms, the initial surge current is as shown in graph 2 but the surge current on successive pulses depends on the thermal time constant and the time between flashes. Graph 3 shows the typical variation in surge current when a 6V, 60mA panel lamp is switched on for 5 s then off for t<sub>off</sub> seconds.

If the p.d. applied to the lamp is gradually increased the current rises in a controlled manner to its normal operating value, prolonging the life of the lamp and reducing the probability of transistor damage. A simple arrangement is shown Fig. 4, where  $Tr_2$  is normally held on and saturated with a low value of  $V_{CE(sat)}$  holding  $Tr_1$  and the lamp off. Under alarm conditions, the base drive to  $Tr_2$ is removed and the capacitor charges through  $R_B$ . The base voltage of  $Tr_1$  rises exponentially so that the lamp surge current is avoided.

To prevent damage to Tr<sub>1</sub> should the lamp become shortcircuited, a resistor Rc could be included in Tr<sub>1</sub>'s free collector, but this would reduce the lamp voltage in normal operation. Circuit Fig. 5 shows a modification that allows an almost normal lamp voltage and also limits the short-circuit current to the desired value by using only a small  $R_{\rm C}$  value and a saturating transistor Tr<sub>a</sub>. Relays are used to actuate alarm devices that need to be isolated from their control circuitry for various reasons such as their current, voltage or power requirements being incompatible with the electronic circuitry. Circuit Fig. 6, of a commonly-used relay drive circuit which takes into account both the resistive and inductive

properties of the relay coil. When actuated, the steady-state coil current is fixed by the coil resistance and supply voltage, but when  $Tr_1$  is turned off the inductance of the coil causes the collector voltage to rise towards a level greatly exceeding  $V_{CC}$  if the protective diode  $D_1$ were omitted. Diode  $D_1$  allows  $V_{CE}$  to rise only slightly above  $V_{\rm CC}$  before the diode conducts to dissipate the energy stored in the relay coil. When Tr<sub>1</sub> turns on  $D_1$  is reverse-biased and does not affect the operation. The diode must be able to withstand a reverse voltage slightly greater than  $V_{CC}$  and be able to conduct the relay-coil discharge current for a brief time. Transistor  $Tr_1$  must have a  $V_{CE}$ rating exceeding  $V_{CC}$  and be capable of carrying the relay operating current. If a relay is required to operate when an input level exceeds a certain predetermined value, it may be included in a Schmitt trigger circuit; e.g., the relay coil and protective diode could replace R<sub>4</sub> in the basic circuit of series 2, card 2.

If the alarm indication uses a l.e.d. or alpha-numeric array of l.e.ds consult series 9, cards 2, 5 & 6.

#### Further reading

Shea, R. F. (Ed.), Amplifier Handbook, section 31–6, McGraw-Hill, 1966. Egan, F. (Ed.), 400 ideas for design, vol. 2, pp. 18/9, Hayden, 1971. Cleary, J. F. (Ed.), Transistor Manual, pp. 202, General Electric Co. of New York, 1964. Industrial Circuit Handbook, section 2, SGS-Fairchild, 1967.

#### **Cross references**

Series 2, card 2. Series 9, cards 2, 5 & 6. Series 13, cards 4 & 7.





Voltage-to-current conversion It is often required to supply signals to relatively long transmission lines in which case the signal is more convenient in current form rather than as a voltage. Thus, voltage-to-current converters are useful and may be realized using operational amplifiers especially if the load is floating. Figs. 1 & 2 show the more common forms the former being an inverting type and the latter non-inverting. In both Figs,  $i = V_{in}/R$  and is independent of the load impedance, but the source and operational amplifier must be able to supply this load current in Fig. 1, whereas little source current is needed in Fig. 2 due to the high input impedance of the amplifier. Fig. 3 shows another floating-load V-to-I converter which requires little source current if R<sub>1</sub> is large and allows  $i_{\rm L}$  to be scaled with  $R_{3}$ , the operational amplifier supplying the whole of the load current;  $i_{\rm L} = V_{\rm in}(1/R_1 +$  $R_2/R_1R_3$ ). The circuit of Fig. 4 is suitable for V-to-I conversion when the load is grounded. When  $R_1R_3 = R_2R_4$  the load current is  $i_{\rm L} = -V_{\rm in}/R_4$  and the current source impedance seen by the load very high.

#### Current-to-voltage conversion

If a device is best operated when fed from a voltage source but the available signal is in the form of a current, a current-tovoltage converter will be required, one example being shown in Fig. 5. Current is fed to the summing junction of the operational amplifier which is a virtual earth so that current source sees an almost-zero load. Input current flows through  $R_1$ producing an output voltage of  $V_{out} = -R_1$  volts/amp. The only conversion error is due to the bias current of the operational amplifier which is algebraically summed with i<sub>in</sub>. The output impedance is very low due to the use of almost 100% feedback.

#### Voltage-to-frequency conversion

Many voltage-to-frequency converters exist, the circuit complexity often being a guide to the degree of linearity and maximum operating frequency. Fig. 6 shows one form of V-to-f converter (a v.c.o.) suitable for use at frequencies below about 10kHz, each amplifier being of the current-differencing LM3900 type. Amplifier A<sub>1</sub> is connected as an integrator with A2 acting as a Schmitt trigger which senses the output from A<sub>1</sub> and controls the state of Tr<sub>1</sub> which either shunts the input current through R<sub>2</sub> to ground, making Vout1 run down linearly, or allows it to enter A<sub>1</sub> causing

 $V_{out_1}$  to rise linearly with  $R_1 = 2R_2$ . So  $V_{out_1}$  is a triangular wave and  $V_{out_2}$  a square wave having a frequency that is linearly dependent of  $R_1$ ,  $C_1$  and the threshold levels selected for the Schmitt trigger.

#### Frequency-to-voltage conversion Diode-pump, transistor-pump

and op-amp pump circuits are widely used for low-cost frequency to voltage conversion. Another circuit, using a single LM3900 quad currentdifferencing amplifier package, is the phase-locked loop shown in Fig. 7 which uses the v.c.o. of Fig. 6. Amplifier  $A_3$  is in the LM3900 package used as a phase comparátor having a pulse-width modulated output depending on the phase difference between Vin and Vout<sub>2</sub> of the v.c.o. Resistor R<sub>a</sub> and C<sub>2</sub> form a simple low-pass filter which makes the d.c. output vary in the range +V to +V/2 as the phase difference changes from 180° to 0°. This direct voltage controls the

frequency of the v.c.o. and its lock range may be increased by using the fourth amplifier in the package as a d.c. amplifier between the filter and the integrator. Centre-frequency of the p.l.l. is about 3kHz with: R<sub>1</sub>, R<sub>3</sub>  $1M\Omega$ ; R<sub>2</sub>  $510k\Omega$ ; R<sub>4</sub>, R<sub>8</sub>, R<sub>9</sub>,  $30k\Omega$ ; R<sub>5</sub>, R<sub>6</sub>  $1.2M\Omega$ ; R<sub>7</sub>  $62k\Omega$ ; C<sub>1</sub> 1nF; C<sub>2</sub> 100nF; V= +4 to +36V.

#### **Further reading**

Graeme, J. G. & Tobey, G. E. Operational Amplifiers, chapter 6, McGraw-Hill, 1971. Linear Applications—application notes AN20 and AN72, National Semiconductor, 1973.

Cross references Set 3, cards 3, 5 & 10. Ser 13, cards 1 & 6.



## Set 13: Alarm Circuits-10





#### Pressure-sensitive alarm

A pressure-sensitive alarm may be made using a speciallymodified transistor known as the Pitran. It is a planar n-p-n transistor having a diaphragm mounted in the top of its metal can which is mechanically coupled to its base-emitter junction. When a pressure is applied to the diaphragm a reversible charge is produced in the transistor characteristics. The mechanical pressure input can be used to directly modulate the electrical output of the transistor which may be fed to the alarm circuitry, e.g., via a comparator or Schmitt trigger which switches state when the input pressure to the Pitran either exceeds or falls below some critical level. The Pitran may be connected as a singleended-input single-ended-output Stage, as shown in Fig. 1 or as a differential-input balancedoutput stage, Fig. 2 Conventional transistor circuit design techniques may be used for the Pitran stages. Linear output voltages of up to onefifth of the total supply voltage are obtainable.

#### Temperature-sensitive alarm

Circuit Fig. 3, shows the input circuitry of an alarm which may be operated by the output signal from the operational amplifier when the temperature monitored by the probe transistor exceeds a pre-determined value. The temperature-sensing transistor is a low-cost n-p-n type that can produce a resolution of less than 1 deg C in a temperature range of 100 deg C. If the operating current of the probe transistor is made proportional to

temperature, the non-linearity of its base-emitter voltage may be minimized, being less than 2mV in the temperature range -55 to +125°C. Zener diodes set the input voltage to 1.2V and this is applied through  $R_2$  to fix the operating current of the probe transistor. Resistor R<sub>4</sub> may be adjusted to make amplifier's output zero at 0°C and R<sub>5</sub> is used to calibrate the output voltage to 100mV/ deg C, or any other scaling factor, independently of the  $V_{out}=0$  condition. R<sub>1</sub>, R<sub>3</sub>  $12k\Omega$ ;  $R_2 3k\Omega$ ;  $R_4 5k\Omega$ ;  $R_5, R_6 100k\Omega; D_1, D_2 LM113;$ Tr<sub>1</sub> 2N2222; A<sub>1</sub> LM112;  $V\pm 15V.$ 

### Moisture-sensitive alarm

A low-cost audible alarm which operates when the electrodes of the input sensor become damp due to increase in humidity, direct contact with water, rain or snow is shown in

Fig. 4. The sensor is conveniently made from parallel-strip printed circuit board or commercial equivalent, so that increase in moisture at the strips produces a very small current to  $Tr_1$  base via  $R_1$  which forms a high-gain compound pair with  $Tr_2$  which switches hard on. Transistors  $Tr_3$  and  $Tr_4$  form the audible alarm multivibrator, that acts as a load on the compound pair, having a repetition rate determined by the  $C_1R_3$  time constant. A piercing note at about 2.5kHz is produced with  $\mathbf{R}_1$ ,  $\mathbf{R}_2$  100k  $\Omega$ ;  $\mathbf{R}_3$  1k $\Omega$ ; C<sub>1</sub> 10nF; Tr<sub>1</sub>, Tr<sub>2</sub>, Tr<sub>4</sub> ZTX300; Tr<sub>3</sub> OC71; LS 8- $\Omega$  loudspeaker; V+9V.

A flashing display with a rate of about 2Hz may be obtained by replacing the loudspeaker with a 6V, 60mA panel lamp and changing the values of  $R_2$  to 470k  $\Omega$  and  $C_1$  to 2.2 $\mu$ F.

#### Further reading

Tingay, E. The Pitran—a new concept in pressure measure-

ment, International Marketing News, p. 8, 1970. Linear Applications—application notes AN31, AN56 and AN72, National Semiconductor, 1973. Brown, F. Rain warning alarm,

Everyday Electronics, pp. 208-11, 1972.



### Set 13: Alarm Circuits—11

### Security, water level and automobile alarms



#### Circuit description

Component R<sub>3</sub> is the resistance in the search loop which if obtained using two  $100k\Omega$ resistors allows one to include switches either in series with the loop or in parallel with either resistor, or both. In the latter case changing a switch condition from open to closed in the parallel case and from closed to open in the series case can give rise to either a positive voltage or a negative voltage being applied to the diode bridge; the bridge is, of course, balanced initially. The diode bridge being a full wave rectifier will apply a negative  $V_{ba}$  to the following circuit in either case. The bridge resistors are large valued to minimize current drain from the battery but require that the following circuit have a large input resistance. Hence the buffer Tr<sub>1</sub> is employed.

When V<sub>ba</sub> goes negative, Tr<sub>1</sub> and with it  $Tr_2$ , conducts. Transistor Tr<sub>2</sub> then drives Tr<sub>a</sub> which is a higher power device capable of drawing a relay coil to produce the warning signal. At the same time when Tr. conducts, the collector of Tr<sub>3</sub> goes negative and hence via positive feedback through R<sub>5</sub> the base of Tr<sub>2</sub> remains negative, even if Vba is set back to zero. Hence, a latching action is obtained, which keeps the warning signal on. The warning signal will only be removed if the power supply is removed.

Capacitors  $C_1$  and  $C_2$  are required to prevent spurious pulses from triggering the alarm, in the case of  $C_1$ , and to prevent switching transients from triggering the alarm when the alarm is being reset, in the case of  $C_2$ .

Brake light monitor (Fig. 2.) Both of the identical counterwound coils are wound round the reed relay. Hence the relay switch will only close, giving a dashboard warning, if either of the brake lights fails either with an open circuit or short circuit.

### Water level alarm

Fig. 3 circuit is designed to produce a note from the loudspeaker when the sensor input terminals are shorted. As such it can be used for many applications apart from suggested water level/rain alarm. When the input terminals are shorted base drive to Tr<sub>1</sub> via  $R_1$  is obtained, and the supply voltage is switched to the unijunction relaxation oscillator comprising Tr<sub>2</sub>, R<sub>2</sub>, R<sub>3</sub> and C (card 4, series 3). A train of pulses of period mainly determined by the product  $R_2C$  is then presented to the base of  $Tr_3$ , thereby producing

Component values  $R_1, R_4: 150k\Omega$   $R_3: 200k\Omega$   $R_2: 250k\Omega$  variable  $R_5: 27k\Omega$   $R_6: 470\Omega$   $C_1, C_2: 0.33\mu$ F  $Tr_1, Tr_2: BC126$   $Tr_3: BFR41 \text{ or } BFY52$ Diodes 0A81  $V_1: 18V$  $V_2: 9V$ 

pulses of current through the loudspeaker. The loudspeaker alarm note can be altered by altering the product  $R_2C$ . Considerable effective output can be obtained by selecting the note to correspond to the resonant frequency of speaker. In practice the alarm will sound for any resistance between zero and five megohms. The quiescent current of the unit is of the order of nanoamps so that battery life is many months even if the unit is switched off. Provision to test the battery condition is made by switch position 2 which should cause  $Tr_1$  to switch on the oscillator provided the battery is in good condition. For water level sensing two conducting rods spaced an inch, or less, apart and positioned at the required level is all that is required.

For a rain alarm two rods separated by some blotting paper will suffice. When the blotting paper becomes wet contact between the rods is made, the alarm sounds and the washing is saved once more (provided the missus isn't away shopping).

#### **Component changes**

Resistor  $R_1$  may be any value up to 5M  $\Omega$  provided a true shorting of the sensor input terminals is obtainable. The  $R_2C$  product is dictated by the pitch of the note required. Resistor  $R_3$  should be much less than  $R_2$ , e.g.,  $R_2/10$ .

#### Further reading

Andrews, J. Security Alarm, Practical Electronics, 1973, p. 338. Moorshead, H. Rain & Water Level Alarm, Practical Electronics, 1971, p. 820. Morum, S. W. F., Brake Light Monitor, Practical Electronics, 1973, p. 588.

#### Component values

R<sub>1</sub>: 100kΩR<sub>2</sub>: 3.3kΩR<sub>3</sub>: 270ΩC: 0.5μFTr<sub>1</sub>, Tr<sub>3</sub>: 2N2926 (G) Tr<sub>2</sub>: 2N2646LS: 8-Ω loudspeaker Supply voltage: 9V



## Set 13: Alarm Circuits-12

### **Electromechanical alarms**

Electromechanical transducers are obtainable in a wide variety of types: they may be d.c. or a.c., resistive, reluctive or capacitive, contacting or noncontacting, analogue or digital, linear or angular, etc. Insofar as most alarm systems use a comparator (cross ref. 1) to compare the signal with a reference and as d.c. signals are easily compared we shall assume here that any a.c. systems are followed by signal conditioning equipment which includes a rectifier (cross ref. 2) of some sort so that the effective output is d.c.



#### **Displacement Alarms**

Circuit shows a reluctive displacement transducer, of the differential transformer type, followed by a demodulator to provide the d.c. output shown in graph. The core, which is shown in its zero output position, is attached to the member whose displacement is required. The core is generally made from high permeability ferromagnetic material so that flux linkages with and hence the e.m.fs of the secondary coils are highly dependent on the position of the core relative to the coils. Reluctive transducers generally have a displacement span of

rectilinear form, and between 0.05 and 90° in angular form. As the induced e.m.fs are proportional to frequency, very sensitive systems can be made at high frequencies. Capacitive transducers are used in situations where very small displacements have to be measured and/or non-contacting measurement has to be performed. Photoelectric/digital measurements (again noncontacting) are used when high accuracy is required, although fairly low cost versions can be constructed if accuracy is not essential.

between 0.01 and 120in, in

#### Acceleration alarm

Acceleration transducers all have one feature in common, viz. the seismic mass, M. The basic acceleration transducer is shown below. The case of the system is attached to the body whose acceleration is required. Due to a constant acceleration the seismic mass exerts a force  $M\alpha$  which in the steady state will stretch or compress the spring by an amount x where  $M\alpha = Kx$ , K being the spring constant. The dashpot simply provides damping whilst the mass is moving. If we know M and K then a measure of x gives a signal proportional to the acceleration. This can be done by any displacement transducer of suitable dimensions and sensitivity. Frequently, however, the spring arrangement is a leaf spring arrangement with strain gauges attached. The spring deflection gives rise to changes in resistance in the strain gauges which if connected in a Wheatstone bridge circuit gives a voltage proportional to the deflection and, hence, to the acceleration. As the Wheatstone bridge can be supplied from a d.c. source there is no need for rectifiers before feeding to a comparator. Strain gauge bridges usable up to 750Hz have been built. For higher frequencies piezoelectric crystals replace the spring. The crystal produces a



charge or voltage across its terminals when subjected to the stress of the seismic mass under acceleration. However, the output impedance of the crystal is large and amplifiers with an input impedance in excess of 500M  $\Omega$  typically have to be used. Furthermore, the cable between the crystal and the amplifier requires to have low capacitance and to be free from friction induced noise. On the other hand very large accelerations (>100g) can be measured and they can be used over a large temperature range (570°C for a lead metaniobate crystal).

#### Further reading

H. N. Norton, Handbook of Transducers for Electronic Measuring Systems, Prentice-Hall.

Considine, Encyclopedia of Instrumentation and Control, McGraw-Hill.

#### Cross references

Set 2, Comparators and Schmitts. Set 4, A.C. Measurements. Set 13, card 4.

#### Velocity alarm

Linear velocity transducers are most commonly used in the vibrations field where the displacement of the member whose velocity is required is small. Essentially, they consist of a coil moving in a permanent magnetic field, the coil e.m.f. being proportional to the speed. As a large proportion of the speed producing systems are driven by motors one can generally obtain information on linear speed from a knowledge of angular speed. This can be obtained by various types of a.c. or d.c. tachometers, but with the increasing use of digital instrumentation, toothed rotor,

photoelectric and similar systems are becoming increasingly common. Diagram shows basis of operation of the toothed rotor tachometer and the corresponding output when the rotor is rotated by the shaft of a motor. The output waveform is obtained because of the changing flux pattern caused by the changing magnetic circuit. If the output signal is fed to a zero crossing comparator (cross refs. 1, 3) or to a Schmitt trigger (cross ref. 1) one will then obtain a train of pulses, each pulse representing the pássage of a rotor tooth past the permanent magnet. Obviously the pulse frequency is



proportional to the shaft speed. If the train of pulses is then fed to a frequency-to-voltage converter a direct voltage proportional to shaft speed is obtained and this can be fed to a comparator to give an alarm if it exceeds a predetermined level. Because the number of teeth on the toothed rotor can easily be varied, the range of speeds measurable by this technique is extremely large. Further, the rotor can easily be constructed in any workshop, no great precision being required for many applications. Bolt heads on a coupling between two shafts often suffice as the toothed rotor.

1. The ability of the 555 timer to act as a monostable or astable and to deliver sufficient power to drive a loudspeaker directly is put to use in this burglar alarm. Switch  $S_1$  is closed arming the circuit, with  $S_2$  consisting of a number of normally-closed switches each detecting an entry point. If any of these switches is allowed to close by an opened window, etc. the relay is activated and power is applied to both monostable and astable. The audio output can be accompanied by a visual output via a second relay contact. The circuit

2. Positive feedback around an operational amplifier may be used to provide either astable or Schmitt characteristics. This novel circuit combines both while needing only one operational amplifier. When the op-amp output is low D<sub>1</sub> is reverse biased and R<sub>3</sub> provides the f.e.t. with a zero gate-source voltage. Hence the f.e.t. shorts out the capacitor grounding the inverting input. As the input voltage increases, it overcomes the effect of the negative voltage across the zeners, raising the non-inverting input above

3. The circuit is an astable gated by a single transistor that can be driven from a small positive current or from a low voltage (>0.6V), including the output of t.t.l., c.m.o.s. etc. Alternatively  $Tr_1$ can be replaced by an opencollector t.t.l. gate. The novel property of this astable that it drives an l.e.d. from a low voltage supply with the l.e.d. permanently on for input level high and flashing at, say, 1-10Hz depending on choice of





zero. The op-amp output swings positive, reverse-biasing the f.e.t. gate source and allowing the capacitor to charge positively. This continues until the inverting input rises to equal the now positive potential on the non inverting input. At this point the op-amp output



C for input level low. This allows a single l.e.d. to act as the on-off indicator for a

battery instrument as a supply voltage of 2V is sufficient to initiate conduction; the same

## Set 13: Alarm Circuits Up-date

been dissipated through  $D_2$ ,  $R_1$ , i.e. the alarm operates for only one cycle. No standby current flows as the contacts of  $S_2$  are normally open. Frequency of oscillation of the astable section is around 1kHz and transformer coupling to a low resistance loudspeaker is suggested in the article.

#### Reference

Long, J. D. Burglar alarm is effective, yet simple and inexpensive, *EDN*, 1974, Dec. 20, pp. 50/1.

switches back to its negative state and the capacitor discharges, with the cycle recommencing. The reference article gives equations representing the circuit performance and indicates that 0.1% stability of trip-point is achievable, though the frequency of oscillation is determined only to within about 30%.

#### Reference

Graeme, J. Voltage comparator circuit gives audio alarm when tripped, *Electronic Design*, 1975, May 10, p. 148.

l.e.d. acts as the visual indicator of a fault condition if that fault is converted into a low voltage/current to drive  $Tr_1$ . In the quiescent state the l.e.d. current may be typically 20mA. The three transistors may be any general purpose silicon devices including those from such multi-transistor packages as CA3046, CA3086, etc. The other two transistors in such a case are usable as part of a Schmitt trigger or amplifier stage.

## Set 14: Digital counters

This set was the first in the revised format for Circards. But for this book, we re-set the first three cards so that they would conform to the rest. The four-column presentation allows greater flexibility of layout and we hope you agree that the "unjustified" type-setting makes for greater readability, with its equal word-spacing and jagged right-hand edge. We took this opportunity to up-date the title area and to rename "series" with the more logical "set". Another excellent summary of the subject precedes this set of circuits. It covers all the essential points about the use of bistable circuits as counters, starting by setting out the different kinds (JK, RS, D and T, discussed in more detail on glossary card 12), defining ripple and parallel counters, design using Karnaugh maps, and concluding with sequence generators. In digital parlance notice that bistable circuits are referred to as flip-flops, whereas originally this term referred to monostable circuits, the words flip and flop indicating an unstable state and a stable state. (Anyone for flip-flip for astables and flop-flop for bistables?) Content of the cards calls for little comment. Card 12 is useful for newcomers, listing various binary codes against decimal number, and in describing the different kinds of "flip-flops". It also shows connections for RS and JK types to give D and T functions.

Basic binary counters 1 One out of n ring counter 2 Johnson counters 3 Reversible counters—I 4 Reversible counters—II 5 Divide by n counters 6 High-power counters 7 High-speed counters 8 Low-power counters 9 Decade counters 10 M-sequence generators 11 Glossary: flip-flops and b.c.d. codes 12

# An introduction to digital counters

A digital counter comprises an interconnection of bistable or two-state memory circuits or, colloquially, flip-flops. The counter embraces those circuits which accumulate pulses according to a specific code as they appear at the input, frequency dividers, sequence generators and pulse waveform generators. The application requirement will generally determine how the collection of flip-flops is identified, but in this article the generic name of counter will be used.

60

The basic flip-flop has one or two control inputs, and two outputs termed Q and  $\overline{Q}$  (not -Q), where  $\overline{Q}$  represents the opposite state of Q. The logic state of these outputs may be termed set or reset, high or low, 1 or 0, and the change of state may occur on 0 to 1 or 1 to 0 transitions at the input, depending on the type used. The varieties of flip-flops used in counters are normally described by the control inputs and are termed D-type, T-type, RS and JK. The triggering input, called the clock-pulse input, ensures that a change of state will only take place on the occurrence of a pulse at the clock-input. Other facilities that may be available are preset and clear inputs which allow a flip-flop to be set (Q = 1) or reset (Q = 0), independently of the control inputs. Typical symbols for these flip-flops are shown in Fig. 1. Other variations include operation by positive or negative logic, triggering on positive or negative pulse edges or a combination of these as in master-slave flip-flops.

A basic RS flip-flop using NAND gates is shown in Fig. 2. To represent the dependence of the Q output on the control inputs when a clock pulse occurs, a truthtable is used to demonstrate the state of Q at the *n*th clock pulse  $(Q_n)$ , and after the next clock pulse  $(Q_{n+1})$ —Fig. 3. For example, if S and R are both at logic zero when a clock pulse occurs, the output Q will not change state, but remain as it was before the clock pulse. However, if S = 1, R = 0, then Q becomes logic 1, i.e. if it was previously logic 0 a change of state occurs, and if it was logic 1, it remains so.

The indeterminate state of Q for the condition that R = S = 1 exists because of a race condition between gates and is one disadvantage of this flip-flop—such a condition must be avoided. The JK flip-flop, however, does not have this disadvantage—all output conditions are predictable, as shown in Fig. 4. The last combination of J = K = 1 permits a useful toggle action in which the output changes state on the occurrence of every clock pulse.

Counters are generally classified as asynchronous or synchronous. The basic asynchronous circuit is implemented with cascaded toggle flip-flops, where the output of a previous flip-flop is the clockinput for the next in sequence. Alternatively, the drive inputs may come from Boolean combinations of other outputs. In either case, a disadvantage is that each flip-flop changes state at a different time in a sequence. For each flip-flop a propagation delay exists between the occurrence of a trigger pulse and the next state of the output, and this delay "ripplesthrough" the counter. This restricts the maximum operational speed of the counter, since the maximum ripple-through delay must be less than the time between input pulses.

In integrated circuit technology, these counters have the advantage that each flip-flop operates at half the frequency of the preceding one. This allows a trade-off in high-speed (high-power dissipation) circuits to be used in the first stage, with lower speed (low-power) configurations being used in later stages. The maximum count (including zero) of a counter containing n flip-flops is  $2^n$ , feedforward or feedback techniques allowing counts less than this to be achieved. The number of distinguishable states through which the counter cycles is known as the modulus of the counter, and this may be fixed when implemented with individual flip-flops, but some m.s.i. packages are available that permit variation of the modulus by a simple connection change or simple gating. If the outputs of ripple-counters are to be



decoded, care must be taken to ensure that the decoder network is enabled only when it is certain that all intermediate state changes have occurred.

The disadvantage of the asynchronous counter is avoided by the synchronous or parallel counter, in which all flip-flops change state in synchronism with a common clock-pulse. The speed of operation is limited by one flip-flop delay and that of any gating necessary, and these will depend on the type of hardware being used, e.g. c.m.o.s., t.t.l., e.c.l. Recent Schottky synchronous counter packages have internal circuitry which eliminates all external gating, and counting speeds up to 70MHz are claimed, and e.c.l. packages are available for speeds up to 110MHz.

Any sequence may be generated using individual JK flip-flops and associated gating. The design is more complicated than the asynchronous types, but one technique simplifies the design problem using Karnaugh maps.<sup>1</sup>

The map is a two-dimensional representation of all possible combinations of a number of variables, where each square is one unique combination and adjacent squares are identical except for one variable. The rows and columns are arranged in accordance with the Gray code representation of decimal numbers, in which only one bit changes as we progress through adjacent numbers (Fig. 5). A Karnaugh map for four variables, A, B, C and D, is shown in Fig. 6, where a 1 in a square means the existence of logical "ANDed" variables, identified by the row and column common to that square.

The 1 squares are connected by the logical OR function, and the Boolean expression represented by Fig. 6 is  $F = \overline{A}.B.\overline{C}.D + A.B.\overline{C}.D + \overline{A}.B.C.D$ + A.B.C.D. A 0 in a square indicates that this particular combination does not exist. The advantage of the map is that minimization of the Boolean expression is simplified by being able to group adjacent squares in pairs, fours, etc. Two squares can be combined to eliminate one variable, and these two squares can be combined with another two adjacent squares to eliminate one more variable. The four squares may be looped as shown, because of their adjacency, thus A and C become redundant as both states of each are included in these squares reducing the function to F = B.D. This can be confirmed by a Boolean minimization. Note that adjacency of squares exists at the extreme ends of horizontal rows, and at the extreme ends of vertical columns.

A design of a modulo-6 Johnson counter is considered as an example of the technique. The maximum modulus of an *m*-stage Johnson counter is 2m, hence a minimum of three flip-flops is required. It is assumed that the outputs are taken from the Q output of the flip-flops designated A, B and C, and the map is used to minimize the gating necessary to obtain the prescribed sequence Fig. 8. As all possible combinations of the variables are not used, the "can't happen" or redundant states are denoted by a combination of one and zero (**()**) in the state table (Fig. 9) because the states are not specified and may be made 1 or 0 at will. In this case they will be considered as 1s. The state table shows the desired outputs at A, B and C on the occurrence of the numbered clock pulse, where 0 and 6 are equivalent, i.e. the 6th pulse resets the counter to zero. It will be assumed that the counter commences from the zero state.

The design technique requires the pre-

DECIMAL	GRAY-CODE			
	A	в	С	D
0	0	0	0	0
1	0	0	0	1
2	0	0	1	1
3	о	0	1	0
4	0	1	1	0
5	0	1	1	1
6	0	1	0	1
7	0	1	0	0
			1 · · · · · · · · · · · · · · · · · · ·	

Fig. 5. Gray code for decimal numbers.



Fig. 6. Karnaugh map for four variables.



paration of a Karnaugh map for the J and K input of each flip-flop to determine the control levels required at each input for every step of the sequence, by deriving a minimal Boolean expression for each map, though as these are derived independently the circuit may not necessarily be minimal. This then determines the internal gating required.

An excitation table for the JK flip-flop is derived from the JK truth-table shown earlier. This table (Fig. 7) shows the necessary J and K inputs to either hold the flip-flop in a 1 condition or a 0 con-

Qn	Q <sub>n+1</sub>	J	к
0	· 0	0	x
0	1	1	×
1	0	×	1
 1	1	×	0

Fig. 7. Excitation table for JK flip-flop.

с	В	А	input pulse no.1
0	0	0	0
0	0	1	1
0	1	1	2
1	1	1	8
1	1	0	4
1	0	0	5
0	0	0	6

Fig. 8. Johnson counter sequence.



Fig. 9. State table for Fig. 8.

dition, or to cause a 1 to 0 or 0 to 1 transition, all on the occurrence of an input clock-pulse. The X indicates that it does not matter what that particular J or K state is, provided the other control input is in the correct state.

For example, if it is assumed that Q = 1, and a transition to logic 0 is required on the occurrence of a clockpulse, then from the truth-table either J = 0, K = 1, or J = 1, K = 1 will cause this change, i.e. provided K = 1, J may be either 0 or 1. As each input pulse occurs, the flip-flops should change in accordance with the truth-table of Fig. 8. The steps involved in filling the JK maps are as follows.

The can't-happen conditions of the statetable are transferred to equivalent squares on the separate J and K maps. Consider the  $J_A$  and  $K_A$  maps. On the occurrence of the first pulse,  $Q_A$  should hold at logic 1, hence an X is put in the  $J_A$  map square representing pulse no. 1, and a 0 in the  $K_A$  map square. This is repeated for pulse no. 2. At pulse no. 3 a 1 to 0 transition is required, hence an X is put in  $J_A$  map square for pulse no. 3, and a 1 in  $K_A$  map. A 0 is maintained for pulse no. 4, hence a O in  $J_A$  and X in the  $K_A$  square for pulse no. 4 is necessary. This is continued until all squares for each map are filled.

As an example of the minimization, notice that symbols  $\mathbf{0}$  or X may be 1, hence a loop of four adjacent squares is available in the  $J_A$  map, i.e. the minimal solution for  $J_A$  is given by  $\overline{C}$ . Similar loops of four are obtained for each of the other maps, the circuit being implemented in Fig. 11. No external gates are required in this circuit, because the complemented outputs are already available from each flip-flop.

Common arrangements using this technique are b.c.d. counters, decade counters, up-down counters, though some are also available in m.s.i. packages.

The implementation of Fig. 11 has been described as a modulo-6 Johnson ring counter. However, examination of Fig. 8 shows that each output  $Q_A$ ,  $Q_B$  and  $Q_C$  has one pulse for every six of the input so that the device could be regarded as a divide-by-six frequency divider. Most frequency dividers, on the other hand, allow one to divide by an arbitrary number so it cannot be regarded as a very good frequency divider.

The device of Fig. 11 can also be regarded as a sequence generator, in that, given an input pulse sequence, one obtains a different output pulse sequence admittedly not a very interesting one.

There is an infinite number of sequence generators that one could build but of particular interest are those which produce so-called maximal length binary sequences (M-sequences). Used in many areas, such as data communication system identification and correlation methods, M-sequences are generated by synthronous shift registers with feedback from various stages being used to determine the next state to be fed in. Feedback complexity is not proportional to the register length and very long sequences can be generated by very simple feedback arrangements. Feedback is performed by modulo-2 addition, i.e. via exclusive-OR gates (Fig. 12).

The properties of these sequences depend on the clock rate,  $f_c$ , and on the number of stages in the shift register, n, but all of

the sequences possess to some degree properties close to those of band-limited white noise. (Hence the name pseudo-random binary sequences or p.r.b.s.) The signal bandwidth is given approximately by  $f_c/_3$  and the greater the value of *n* the more closely do the properties resemble those of random noise. The sequences are not in fact random because they are binary in nature and because they are cyclic, the cycle length being  $2^n - 1$  clock periods. The binary nature of the signals is easily removed by passing them through simple first- or second-order filters so that the signal becomes continuous and has a probability density function which is close to Gaussian. The cyclic nature of the signal is in fact one of the advantages of M-sequence and is one of the non-random features one would wish to retain. This is because experiments can be repeated for checking purposes over a cycle length without the statistical difficulties of genuine random noise. From this point of view it is therefore desirable to limit n.

Design is greatly facilitated by tables which indicate what feedback paths are necessary to produce an M-sequence of given length.<sup>2</sup> The problem comes down to one of choice of  $f_c$  and of *n* for the particular application in mind.



## Set 14: Digital counters-1

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# wireless world circard

### **Basic binary counters**



#### **Circuit** operation

The bistable circuit is a T-type "flip-flop" in which the output changes state for a negativegoing transition at the trigger-input. If the base-drive current is arranged so that Tr<sub>2</sub> is in saturation, its collector voltage will be about 0.2V. This is too low to forward-bias the base-emitter junction of Tr<sub>1</sub>, about 0.7V, and hence Tr<sub>1</sub> will be off. This means its collector-emitter voltage is high, depending on R<sub>1</sub> and R<sub>8</sub>, and the base-drive current for Tr<sub>2</sub> flows through R1 and R3. Hence the terminals identified (arbitrarily) as Q and  $\bar{Q}$  are low and high respectively (0 and 1 for binary coding). When the trigger input is high, the circuit is in a stable state. When the trigger input is driven near ground the negative-going pulse-edge is steered to Tr<sub>2</sub> base as D<sub>2</sub> is forward-biased. The anode of  $D_1$  is approximately at  $V_{CE(sat)}$  and because its cathode is connected to a high potential via R5 it is reversebiased. Therefore Tr<sub>2</sub> collector current is reduced, causing a rise in its collector voltage, increasing base-drive current to Tr<sub>1</sub>. This causes Tr<sub>1</sub> collector voltage to drop and Tr<sub>2</sub> base current decreases causing a further increase in Tr<sub>2</sub> collector voltage. The process continues until the other stable state, Tr, conducting and Tr<sub>2</sub> off, is sustained. The next negativegoing trigger pulse resets the

circuit to its previous state. It produces one output pulse for every two trigger pulses. The interconnection of these bistable circuits to give a binary ripple counter demands that the Q output of a previous flip-flop is connected to the trigger (or T-) input of the next flip-flop. This gives a natural count of  $2^n$  where *n* is the number of stages, and  $2^n$  is the number of states through which the counter progresses.

#### **Circuit modification**

Range of  $R_5$ ,  $R_6$ : 4.7k to  $47k\Omega$ Frequency variation: 150 to 30kHz Range of  $C_1$ ,  $C_2$ : 330 to 3300pF

Frequency variation: 140 to 90kHz Increase turn-on speed with capacitors across resistors R<sub>3</sub>

and  $R_4$  typically 5 to 20% of  $C_1$ ,  $C_2$ . Increased frequency of

operation possible with additional diodes connected across  $R_5$ ,  $R_6$  (anode to collector).

High-speed transistors BSX19, BSX20 permit counting speeds up to 10MHz.

#### IC binary counter

The ripple binary counter is commonly implemented with integrated circuits using J-K flip-flops e.g. the SN7493 is a 4-bit binary counter within one package allowing a typical count rate up to 18MHz for d.c. supply +5V, and a typical



Typical data Single bistable  $V_{CC}$ : +12V  $Tr_1$ ,  $Tr_2$ : BC108  $R_1$ ,  $R_2$ : 3.3k $\Omega \pm 10\%$  $R_3$ ,  $R_4$ : 8.2k $\Omega \pm 10\%$  $R_5$ ,  $R_6$ : 6.8k $\Omega \pm 10\%$  $C_1$ ,  $C_2$ : 800pF  $D_1$ ,  $D_2$ : PS101 Frequency 100kHz typically Trigger input  $\approx 4V$ Trigger input width > 1 $\mu$ s

load of  $400\Omega$  and 15pF. Synchronous binary counters using dual J-K master-slave flip-flops are shown above. In all cases decouple the power supply—typically  $0.01\mu\text{F}$  per package.

In Fig M1, since  $J_A = K_A = 1$ (high), the first flip-flop acts as a toggle. The second flip-flop is triggered by alternate clock pulses the third flip-flop is gated by the  $Q_A$  and  $Q_B$  outputs and only changes when  $Q_A = Q_B = 1$ . Similarly, the last flip-flop only changes state when  $Q_A = Q_B =$  $Q_C = 1$ . This counter has the disadvantage of long counter chains requiring AND gates with a large fan-in.

The situation is avoided with the counter of Fig. M2 where the fan-in is limited to two per gate. However this is a slower counter because the gatedpulses must propagate down the AND gates before the next clock-pulse arrives. For both these counters, use the SN7473 dual J-K flip-flop package.

Another example (Fig. M3) employs the SN7472 which has effectively 3-input AND gates for each J and K input, within the package, which eliminate the need for external gates.

#### Further reading

Electronic Counting, Mullard, 1967.

Designing with TTL Integrated Circuits (Texas), McGraw-Hill 1971.

Counter delay slashed in half with interconnection scheme, *Electronic Design* 13, 1972.

Cross references Series 14, cards 4, 6 & 12.



## Set 14: Digital counters-2

### One out of n ring counter



**Typical data** IC<sub>1</sub>: SN7495 IC<sub>2</sub>:  $\frac{1}{2}$ SN7474 V<sub>8</sub>: +5V R<sub>1</sub>: 1k $\Omega$ C<sub>1</sub>: 47pF

#### **Circuit description**

Component  $IC_1$  is a 4-bit shift left or right register, comprising master-slave R-S flip-flops, with a parallel-loading capability via the AND-OR-NOT gates at terminals U, X, Y and Z. This is conditioned by mode-centre terminal MC equal

Clock pulse No.	QA	Qв	Qc	QD	QE
1	0	1	1	1	1
2	1	0	1	1	1
3	1	1	0	1	1
4	1	1	1	0	1
5	1	1	1	1	0
6	0	1	1	1	1

to binary one. When MC=0, information transfers serially through the register, the clock-pulses being applied to the commoned right-shift and left-shift inputs (not shown). When used in conjunction with a positive-edge triggered flip-flop, IC<sub>1</sub>, this arrangement provides a self-starting, self-priming ring-counter for circulating a zero. When the supply  $V_s$  is switched on, the clear-input of IC<sub>1</sub> is pulled down to ground by CR network, setting  $Q_E = 0$ , and  $Q_E = 1$ . Hence MC = 1, and the counter is in the parallel mode. The left-hand AND gates are inhibited, and the voltage levels at inputs U, X,

Y and Z are transferred to the set inputs,  $S_A$  to  $S_D$ , of IC<sub>1</sub>. In master-slave flip-flops, the binary level at the set terminal is transferred to the Q terminal on the occurrence of the negative-going edge of the clock-pulse.

After the first clock-pulse,  $Q_A = 0$ ,  $Q_B$  to  $Q_D = 1$ . Also, Q<sub>A</sub> is connected to the preset input of  $IC_1$ , and hence  $Q_E$  is set to binary one. Hence  $\bar{\mathbf{Q}}_{\mathbf{B}} = \mathbf{0}$ , and the counter is switched to a serial-mode. The right-hand AND gates are now inhibited. Therefore  $S_A = 1$ , the low level of QA is gated to  $S_B$ ,  $Q_B$  to  $S_C$ , and  $Q_C$  to  $S_D$ . After the second clock-pulse, the Q outputs are as shown in the truth table, the sequence continuing with each clockpulse shifting the zero through the register. At the 5th pulse, Q<sub>D</sub> changes from 0 to 1, and this positive-edge triggers IC<sub>1</sub>.  $Q_E$  resets to zero,  $\bar{Q}_E = 1$ , and the parallel-mode is again entered. The 6th clock-pulse reloads the levels at U, X, Y and Z terminals and the cycle repeats.

#### **Circuit modifications**

• The number of bits can be extended by cascading  $IC_1$ packages as in Fig. M1.

• Circulate a 1 by inverting the counter outputs with t.t.l. NAND gates (SN7400) or c.m.o.s. hex buffers (CD4049). • A 4-bit self-starting and correcting counter (Fig. M2) for circulating 1 uses J-K flip-flops and feedback via an AND gate. The flip-flops are connected as a shift register, where the state of  $Q_A$  is passed to  $Q_B$ , and  $Q_B$  to  $Q_C$ , etc. with each clock pulse. In general, for self-correcting, and when using J-K flip-flops,  $J_A$ should be the Boolean product of the complements of all but the first and last stages.

#### Further reading

Self-correcting ring counter requires no external gates, *Electronic Design* 9, 1973, p.138. Malmstadt and Enke, Digital Electronics for Scientists, Benjamin, 1969. Texas Instruments application report CA102, Electronic Counting, Mullard, 1967.



## Set 14: Digital counters—3

### Johnson counters



#### **Circuit description**

This counter, also called a switch-tail ring counter, allows 2n counts where *n* is the number of cascaded flip-flops. It is a synchronous counter in that changes at the Q outputs only take place on the occurrence of a clock-pulse. If  $J = \overline{K}$ , the J input condition is transferred to the related Q output on the negative edge of the clock pulse, when the flip-flops are master-slave types. The above circuit has ten different states, the feedback via the AND gate causing self-correction after a few steps, should illegitimate states occur.

Consider all Q outputs to be in the 0 state. Hence  $J_A = 1$ because  $\bar{Q}_E = 1$ . On the occurrence of the first clock pulse, the counter will load according to the truth-table. At each subsequent pulse, "1"s will be fed through the counter from the left until  $Q_D = Q_E = 1$ . It follows then that since  $J_A = 0$ ,  $K_A = 1$ , that  $Q_A$ becomes 0 at the 6th clockpulse. Zeros are subsequently transferred through the register according to the truth-table, until  $Q_D = Q_E = 0$ , then  $J_A=1$ ,  $K_A=0$ , and the cycle repeats.

In general, when there are n stages in the counter, feedback via an AND gate from the last x stages, where x is the next larger integer to n/3 provides self-correction for n up to 25.

Decoding of each state is obtained using AND gates, as a unique pair exists for each state. These output pairs are indicated in the truth-table, and are applied to the gate inputs (these can be  $\frac{1}{2} \times SN7400$ in series).

#### Circuit modifications

The same output sequence is obtained from dual D-type flip-flops (SN7474) as Fig. M1. In general, assuming up to eight JK flip-flops are employed, A, B, C, D, E, F, G, H, then the

ciock puise	A	в	с	D	E	decoding outputs
	0	0	0	0	0	
1	1	0	0	0	0	АB
2	1	1	0	0	0	BĈ
3	1	1	1	0	0	СÐ
4	1	1	1	1	0	DĒ
5	1	1	1	1	1	AE
6	0	1	1	1	1	ĀВ
7	0	0	1	1	1	āc
8	0	0	0	1	1	ζD
9	0	0	0	0	1	δe
10	0	0	0	0	0	ĀĒ

feedback functions for an even cyclic pattern are

$J_A = \overline{C}, K_A = B.C$	(6)
$J_A = \overline{D}, K_A = C.D$	(8)
$J_A = \overline{E}, K_A = D.E$	(10)
$J_A = \overline{F}, K_A = E.F$	(12)
$J_A = \overline{G}, K_A = E.F.G$	(14)
$J_A = \overline{H}, K_A = F.G.H$	(16)

An odd sequence (2N-1)counter can be implemented by bypassing the all "1"s state of the normal Johnson sequence i.e. the last two bits of the 111 ... 10 state is detected. and the gating arranged to the first flip-flop so that the next state is 011 . . . 11. (Texas ref.). Other odd sequence counters can be implemented by J-K flip-flops without extra gating and are shown in Figs. M2 & 3. Fig. M2 uses feedforward gating and Fig. M3, feedback. In general, any 2n counter can be made 2n-1 by obtaining the K input of the first-flop from the second last Q output (cf. Fig. M3).

#### Further reading

**Typical** data

Supply: 4.75 to 5.25V  $IC_1$ ,  $IC_2$ :  $\frac{1}{2}$  SN7400  $IC_3$ ,  $IC_4$ :  $\frac{1}{2}$  SN7473 Min. clock width: 20ns (between 50% levels) Typical pulse height: 3.5 to 4.5V

Davies, A. C., Design of feedback shift registers and other synchronous counters, *Radio and Electronic Engineer*, April 1969.



Texas Instruments application report CA102. Kohonen, T. Digital circuits and devices, Prentice-Hall, 1972. Malvino, A. P. & Leach, D. P., Digital principles and applications, McGraw-Hill, 1969.

Cross references

Series 14, cards 9 & 12.

## Set 14: Digital counters—4

### **Reversible counters—1**

#### **Circuit description**

Up-down or reversible counters alter their mode of counting under electrical control. The circuits shown are binary up-down counters in which the count direction is controlled by steering logic and either the flip-flop Q outputs are used for toggling subsequent flip-flops (UP) or the Ö outputs are used when the Q outputs are inhibited. Fig. 1 is an asynchronous type and Fig. 2, a synchronous counter. The equation for the output of the selector gate is, for example,  $AX + \bar{A}\bar{X}$ , the gating usually being implemented by the NAND gate interconnection of Fig. 3. When a logic 1 is applied to the up-down control line, the gates connected to  $\bar{O}$ are inhibited, and the flip-flops change state when their clock inputs undergo a 1 to 0 transition i.e. the Q outputs will change according to the normal binary sequence. If the control line becomes 0, the Q outputs will reverse sequence. To avoid a false triggering condition, the count/inhibit line

must be 0 when the controlling function is altered from up to

down or vice-versa. In the synchronous counter of Fig. 2 such a condition is avoided. Each flip-flop will change state only if previous Q outputs are in the 1 state (when X = 1) i.e. counting up, and for a down count (X = 1), all the less significant Q outputs must be at 0 for a flip-flop to change on a 1 to 0 transition.

IC<sub>5</sub> (CD4029A) is a presettable synchronous up/down counter providing either a binary or b-c.d. decade sequence with appropriate mode control, i.e. binary: B/D = 1Decade: B/D = 0Up: U/D = 1Down: U/D = 0Preset: PE = 1J1 to J4 = 1 or 0 Parallel clocking allows cascading (Fig. 4) Where separate "clock up" and



"clock down" clock pulses are available the circuit of Fig. 5 provides a simple interface to the up/down "clock" inputs.

### Cross references

Series 14, cards 5 & 12.





## Set 14: Digital counters—5

### **Reversible counters—2**

The AND-OR-NOT gates provide gating control for the up/down mode. The Boolean expression for the P output is  $(UP)\overline{Q}_D + (DWN)Q_B$ . If the control line = 0, then  $P = Q_D = 0$ . Hence  $J_A = 1$ ,  $K_A = 0$ , and on the occurrence of the first clock pulse  $Q_A = 1$ . The counter would subsequently count up in Johnson code. However, if for example, the counter state is 1110 and control = 1, only the S output causes a change to make  $Q_c = 0$  on the next pulse, and thus count down. Circuit of Fig. 2 uses a 4-bit parallel adder and four D-type flip-flops. Consider the sum outputs  $\Sigma_4 - \Sigma_1$  show  $7_{10}$ i.e.  $0111_{s}$ , where  $\Sigma_1$  is the least significant bit, and that the next pulse should cause a decrement, hence control = 1. Each sum variable ( $\Sigma$ ) is transferred to the Q output on the occurrence of the clock pulse via each D-type flip-flop. Inputs  $(B_1A_1)$ ,  $(B_2A_2)$  etc. are added and each produce 0 carry 1 except for (B,A,) which makes  $\Sigma_4 = 1$ . When the "carry's" ripple through, the result is 0110 with  $C_{out} = 1$ , which can be ignored. (This is 2's complement arithmetic. where negative-one, represented by 1111, is added to achieve subtraction). IC1: SN7483 (full adders) IC2: SN7495 (or 2×SN7474)

Typical operating frequency  $\approx$  10MHz

An eight-bit shift register is used to provide a variable modulus counter in which maximum and minimum counts may be detected is shown in Fig. 3 Shorting any one pair of



external terminals decides the maximum counter loading at which the counter may then be considered to reverse. Assume a counter cleared and a link at Q<sub>C</sub>. Then, via gates X, Y, Z, S<sub>A</sub>=1, R<sub>A</sub>=0. The sequence of Q<sub>A</sub>, Q<sub>B</sub>, Q<sub>C</sub> for the first three clock pulses is 100, 110, 111 and then S<sub>A</sub>=0, R<sub>A</sub>=1 and "detect max" goes high. For the next three pulses, the sequence is 011, 001, 000, when "detect min" goes high. IC<sub>1</sub>: SN74164 (DM8570) IC<sub>a</sub>: SN7400

#### **Further reading**

Malmstadt & Enke, Electronics for Scientists, Benjamin, 1969. High-speed synchronous reversible binary and BCD counters. Texas Instruments application report B40. Baccolini, G., Benetazzo, L., & Clements, G., Variable dead-zone counter as a maximum value follower, *IEEE Trans.* vol. ECI-20, Aug. 1973.

Cross references Series 14, cards 4 & 12.

#### Integrated circuits IC<sub>1</sub>: $\frac{1}{2}$ SN7473 IC<sub>2</sub>: $\frac{1}{2}$ SN7451

IC<sub>1</sub>: <u>1</u>SN7451 IC<sub>3</sub>: <u>1</u>SN7400



## Set 14: Digital counters—6

### Divide by n counters

Counters with N states (modulus N) may comprise a sequence of counters with different moduli, integers  $n_1$ ,  $n_2$ ,  $n_3$  etc. and  $N = n_1 \times n_2 \times n_3 \dots$  Even numbers are achieved by cascading as in Fig. 2 (modulus  $2^m$ , where *m* is the number of flip-flops). Synchronous circuits for smaller odd numbers are shown below.  $Q_A$  is the least significant bit. the natural binary sequence is followed and positive logic assumed.

### Components (typical)

IC<sub>1</sub>:  $\frac{1}{2}$ SN7473. IC<sub>2</sub>:  $\frac{1}{3}$ SN7410. IC<sub>3</sub>:  $\frac{1}{3}$ SN7400. IC<sub>4</sub>: SN7472 (or  $\frac{1}{2}$ SN7473 plus 3 input gates). IC<sub>5</sub>: SN7493. IC<sub>6</sub>:  $\frac{1}{3}$ SN7408 (or  $\frac{1}{2}$ SN7400). IC<sub>7</sub>:  $\frac{2}{3}$ SN7410.

#### Description

For the small prime numbers, it is sufficient to consider Fig. 3 as an example. QA flip-flop acts as a toggle and will change state on every input pulse while  $\bar{Q}_{C} = 1$  ( $Q_{C} = 0$ ).  $Q_{B}$  changes state whenever QA goes from 1 to 0. Qc will only be set to 1 when both  $Q_A$  and  $Q_B$  are logic one i.e. on the 4th clock-pulse, because  $J_1 = J_2 = J_3 = 1$  and the K inputs are 0. Hence on the 5th pulse, since  $\bar{Q}_{C}$  is now 0,  $Q_{A}$ will remain at 0 hence QB is logic zero, and Qc will be reset to 0, because  $J_1$  to  $J_3 = 0$ and  $K_1$  to  $K_3 = 1$ . The sequence then repeats.

Fig. 6 demonstrates an alternative technique. The counter is reset to zero when a negative-going edge is simultaneously applied to the CLEAR inputs. This is obtained from the NAND gate, when the predetermined binary number, 1101 (13), is detected. The minimum duration between input pulses depends on propagation delays of flip-flops, the gate delay, and the reset delay. If the output is to be read, the lines should be gated to avoid transmitting spikes that will appear on some lines depending on the divisor N.



SN7493 connection				
Feedback	Extra gate			
A D				
BD	<del></del>			
ABD	2 I/P AND			
CD	·			
ACD	2 I/P AND			
BCD	2 I/P AND			
ABCD	3 I/P AND			
	Feedback A D B D A B D C D A C D B C D			

MSI integrated circuit packages e.g. SN7493 4-bit ripple counter contains a NAND gate for clearing, but additional gating is necessary for certain counts (see Table 1). Fig. 7 is the connection for  $N=11_{10}$ , the feedback being applied via reset terminals  $R_{0(1)}$  and  $R_{0(2)}$ . A useful arrangement for division by large numbers is shown in Fig. 8, where  $N_1$  and  $N_2$  must be prime numbers



with respect to each other. The arrangement  $(N_1=3, N_3=7)$  provides N = 21; input pulses are applied simultaneously to each counter, and each cycles through its own modulus until all outputs are l's together, thus enabling AND gates, and resetting to zero via reset terminals.



## Set 14: Digital counters-7

### High-power counters



#### **Circuit description-1**

Circuit shows a four-stage dynamic ring counter with power capabilities per stage of 20W at between 2 and 5A. Consider first the situation where all the s.c.rs are nonconducting. A set pulse applied at the gate of SCR<sub>1</sub> causes it to conduct freely via the load resistor R<sub>1</sub> and this continues in the absence of a set voltage until the supply is removed, which occurs when a trigger pulse is applied as base drive to Tr<sub>2</sub> is then shunted to ground, removing base drive to Tr<sub>3</sub>. While the supply is



#### Circuit description-2

A stepper motor drive is required to rotate a magnetic field pattern in either direction and at variable speed. This can be achieved by supplying the stator coils from an up-down counter with a variable pulse rate (see Card 4). Circuit shows  $\frac{1}{4}$  of the drive circuitry for an 8-phase stepper motor with power consumption of 11W which is dissipated in

applied to the counter and SCR<sub>1</sub> is conducting, point B is at ground voltage and C<sub>1</sub> charges via D<sub>3</sub>, so that  $V_{AB} = V_{CC}$ . When the supply is removed, C1 retains this charge as there is no discharge path (apart from leakage). When the supply is reapplied, SCR<sub>1</sub> remains non-conducting and point B rises to the supply. Consequently, point A will rise to  $2V_{CC}$ . If D<sub>4</sub> is chosen so that its breakdown voltage lies between  $V_{\rm CC}$  and  $2V_{\rm CC}$ , breakdown occurs and SCR<sub>2</sub> conducts so that current flows in the second load, R<sub>2</sub>. Further

## Component values $R_1, R_2: 33\Omega, 16W$

C<sub>1</sub>, C<sub>2</sub>:  $100\mu$ F, 40VW<sub>1</sub>, W<sub>2</sub>:  $9\Omega$  stator coils Tr<sub>3</sub>, Tr<sub>4</sub>: BSW 66 IC: SN75451P

#### Performance

Controlled by the torque characteristics of the motor. The coil current was 550mA and the maximum speed was 6000 steps per second, corresponding to a counter clock frequency of 6000 pulses per sec. A and Ä were normal t.t.l. voltage levels (5V and 0V).

only 4 of the phases at any one time. Two of these phases are shown in the diagram in the form of the  $9\Omega$  coils,  $W_1$  and  $W_2$ . The motor operation requires that current flows in one of these coils at a time and this is achieved as shown. A and  $\tilde{A}$  are obtained from one stage of a 4-stage Johnson, up-down counter. If A is "high" then  $T_1$  will not conduct and base drive is Component values  $R_1, R_2, R_3, R_4$ : 5 to 10 $\Omega$   $R_5, R_6, R_7, R_8$ : 100k $\Omega$   $C_1, C_2, C_3, C_4$ : 0.47 $\mu$ F  $D_2, D_4, D_6, D_8$ : 1N757  $D_1, D_3, D_5, D_7$ : 1N914 SCRs: 2N1595  $Tr_1, Tr_2$ : 2N1420  $Tr_3$ : 2N657A

trigger pulses will cause each

succeeding stage to conduct.

continue discharging below the

zener voltage of D<sub>4</sub>. Diode D<sub>3</sub>

arrests this discharge so that the final condition is zero

charge on C<sub>1</sub>-which is the

SCR<sub>1</sub> conducting.

initial condition of C1 prior to

Continual rotation of a single

described so far. Two adjacent

logic 1's cannot reliably be

rotated by this scheme but

there is no reason why any

pattern cannot be rotated

provided no two logic 1's are

logic 1 is what has been

Resistor  $\hat{R}_6$  allows  $C_1$  to

#### Performance

Vcc: 9V Trigger amplitude -6 to 9V width >200µs Set pulse: 3V max Maximum frequency: 1kHz Minimum frequency: depends on capacitor losses, s.c.r. leakage current, and diode leakage current

adjacent. Should more than one logic 1 be rotated then the supply section comprising  $Tr_1$ ,  $Tr_2$  and  $Tr_2$  will require redesign as the current drawn is n + (current for one stage), n being the number of on stages.

To prevent noise falsely triggering any s.c.r., resistors between gate and ground should be provided.

#### Further reading

Strangio, High power counter drives 20W loads, *Electronics*, March 1, 1973

presented to  $T_4$  and consequently current from the 23V supply will flow through  $W_1$ . Likewise no current flows through  $W_2$  in this condition. However, if Å is high the position is reversed.  $Tr_4$  and  $Tr_3$ are high current transistors capable of feeding inductive loads.  $R_1$ ,  $C_1$  and  $R_3$ ,  $C_3$  serve to reduce the circuit time constant so that higher speeds may be achieved.

#### **Further reading**

Shakaiba, M. A. Digital eontrol system for an 8-phase stepper motor, Project Report, Electrical Engineering Dept, Paisley College of Technology. King, D. S. Stepper motor for digital control systems, *Control* and Instrumentation, June 1971.

Cross reference Series 14, card 5.

## Set 14: Digital counters—8

### **High-speed counters**

#### **Circuit description**

In most counting arrangements the decoding of the end of the sequence and the resulting resetting of the counter occur in the same clock period. This can be avoided by using the philosophy of the circuit of Fig. 1 and thereby obtain increased counting speeds, irrespective of the logic family used. Circuit shows a frequency divider in which the second last number to appear on the output of the counters is decoded. This is done by the AND gates  $A_1$  and  $A_2$  and occurs on the falling edge of the clock pulse when A<sub>0</sub> goes to logic 1, corresponding to the number 97. Inputs J' and K' are then in the logic 1 state and the next clock pulse triggers the flip-flop and also the strobe pulse. The counter remains in this state until the clock pulse at the end of the 99th state toggles the flip-flop back again. The counter is then ready to start up-counting again from the initial state, which is, of course, dictated by the b.c.d. data inputs.

This arrangement allows the reset pulse to be a full clock period wide and unaffected by the counter states. Further, the decoding time and the reset time occur in different clock periods, rather than in the same period as in other methods. Hence the time taken for N to be fed in can be almost a full clock period and as a result higher clock frequencies can be handled.

With the devices quoted

counting speeds of over 40MHz have been achieved. The second stage of the counter need not be such a high speed device (and, consequently, high power consuming device) as clock





#### Components

Counters: 8290, decade up-counter for which the input data is on output when S is logic 0. Flip-flop: 74H102 (high speed)  $A_1, A_2, A_3: \frac{1}{3}$  MC3006

frequency to it is 1/10th of that to the first stage. Higher frequencies are achievable if one uses faster logic families. Fig. 2 shows an e.c.l. device being used in a fashion similar to that of Fig. 1. In this case the device is a down counter, counting down from the preset state N and giving an output frequency  $f_{\rm CL/N}$ . Clock frequencies in excess of 110MHz can be handled. Again the second last number in the sequence is decoded, in this case 0100, and this causes the flip-flop to change state on the next clock pulse, thereby resetting the counter to N.

#### **Further reading**

Clifford, D. Reset dividers faster with a single flip-flop, *Electronic Design*, Aug. 5, 1971. Balph & Granden, Boost counting speeds to 110MHz, *Electronic Design*, April 1, 1973. Tan, Z. C. New tunnel diode ring counter, *Proc. IEEE*, April 1973.

### Low-power counters



#### **Circuit description**

The counter of Fig. 1 uses a Johnson configuration and obtains speeds in the region of 5 to 7MHz, each flip-flop operating at only one tenth of the input frequency. The counter is disabled if enable is high, reset to zero being achieved by applying a logic 1 level to reset. The counter sequences on the leading edge of each clock pulse, the Johnson code being maintained by ensuring that the D<sub>c</sub> input is only high, when either  $Q_A$ and QB or QC and QB are high, via NOR gates 1 and 2. Additional gating (not shown) provides ten decoded outputs, which sequentially are high for one full clock period. For count <ten, use quad-NOR i.c. CD4001. In Fig. 2 the cross-coupled pair is a reset latch, to ensure reset when flip-flops have different reset propagation delays. The decimal zero output is low except when counter is cleared, hence when the N output goes high, point  $X \rightarrow 0$ , and a reset

Fig 2

clock

CD4017

ړ

pulse is generated while the clock is high. When counter resets, zero terminal goes high. This can drive another counter.

#### Programmable counter

N = 1 to 10 IC: CD4018 This is a 5-stage Johnson counter but with the Q outputs buffered with inverters to provide a Johnson BCD output. Counts less than ten achieved by feedback to the DATA input terminal, see Fig. 3. The odd count is obtaining by ANDing the two  $\bar{Q}$  outputs (table) with 1 CD4011. This ensures that the all '1's state of the counter is avoided. The counter may be preset to any combination fed to the J inputs, by pulsing the presetenable input. Voltage levels and speeds are similar to CD4017AE. As a preset counter, counter will advance from preset state to 11110, where the right-most bit is  $\bar{Q}_{5}$ . The presence of  $\bar{Q}_4$  and  $\bar{Q}_5$ should be detected as shown in Fig. 4 to reset the counter.



14 stages but outputs available from stage 1 and stages 4 to 14 inclusive. Typical speed: 7MHz at 10V 2.5MHz at 5V

Power dissipation: typical 1mW at 1MHz at 5V 10mW at 1MHz at 15V

### **CD 4040AE**

12 stages All 12 buffered outputs available

Typical speed: 8MHz at 10V. Up-Down counter CD 4029AE

4 stage: Either BCD decade or



our court	~~~~	14
	$\bar{\mathbf{Q}}_{1}  \bar{\mathbf{Q}}_{2}$	3
	<b>Q₂ Q</b> ₃	5
	Q, Q,	7
	Q₄Q₅	9



#### Package CD 4017AE Temperature: -40 to $+85^{\circ}C$ Typical speeds:

Set 14: Digital counters—9

5MHz at 10V d.c. 1MHz at 3.5V 100kHz at 3.5V Approximate power dissipation for the above values are 30mW, 1mW and 100 $\mu$ W respectively, for 15pF loading. Minimum pulse width 100 nanoseconds at 10V.

binary by input control Typical speed: 5MHz at 10V Dissipation: 30mW Load: 15pF Typical speed: 100kHz at 3.5V Dissipation: 1mW Load: 15pF

#### Further reading

**RCA Solid State Databook** Series SSD-203A 1973

**Cross references** Series 14, cards 4 and 10. Series 11, card 6.


#### **Decade counters**





Fig 4



Fig 5



#### **Circuit description**

The four master-slave flip-flops of Figs. 1 and 3 are contained within one i.c. package, SN7490, and provides separate  $\div$ 5 and  $\div$ 2 facilities. A symmetrical decade counter where the period of the output pulse at Q<sub>A</sub> is ten times the input pulse period with equal mark-to-space ratio is shown in Fig. 1, with the associated waveforms of Fig. 2. The J-K terminals with no inputs are internally connected to be logic 1. Flip-flop C toggles for all 1 to 0 transitions of Q<sub>B</sub>, and  $Q_D$  is set on the 4th pulse, but reset on the 5th pulse (S = 0, R = 1), thus setting  $Q_A$ . QB cannot change because  $J_B = 0$ ,  $K_B = 1$ . For the next five pulses, the sequence of flip-flops B, C and D is similar, when  $Q_A$  is reset on the tenth pulse.

The connections of Fig. 3 allow the counter to sequence in an 8 4 2 1 b.c.d. code, where flip-flop D has the maximum weighting: gated direct reset lines (not shown) are provided to inhibit count inputs and return outputs to zero. Typical frequency and power dissipation is up to 18MHz, and around 160mW.

Another asynchronous 8 4 2 1 b.c.d. counter uses J-K flip-flops in a toggle mode (Fig. 4), has no logical hazards, and may be implemented with two SN7473 and one SN7400.

Fig. 5 is an 8421 b.c.d. synchronous counter, which requires 3-input AND gates or triple input J-K flip-flops. Q<sub>A</sub> changes state for every clock-pulse (unused J-K inputs may be connected to logic '1'). Q<sub>B</sub> changes on the 2nd, 4th, 6th, 8th clock pulses, but is inhibited from a 0 to 1 transition on the 10th pulse, since  $\dot{Q}_D = 0$ . Flip-flop C is toggled whenever  $Q_A = Q_B = 1$ , and  $Q_D$ undergoes a 0 to 1 transition when  $Q_A = Q_B = Q_C = 1$ , on the occurrence of the 8th pulse. Q<sub>D</sub> resets to zero on the 10th pulse, because  $J_1 = 0$  and the K inputs are high.

Implemented with either two SN7473 and three SN7410, or four off SN7472.

Figs. 6 & 7 are two other forms of b.c.d. synchronous counters. Fig. 6 counts in 8421 code, Fig. 7 in excess-3 code. In each case the least significant bit is flip-flop A.

#### Further reading

T.T.L. Integrated Circuits, Counters and Shift Registers, application report CA102. Texas Instruments. Malmstadt and Enke, Digital Electronics for Scientists, Benjamin, 1969. Kohonen, T. Digital Circuits and Devices, Prentice-Hall, 1972.

Cross references Series 14, cards 3 & 9.



# Set 14: Digital counters—11

#### M-sequence generators



#### Description

Maximum length sequences (M-sequences) have many uses in data communications system identification and correlation methods. Some of them have properties very similar to that of band limited white noise, particularly if passed through simple first or second order R-C filters-hence the name pseudo-random-binary sequence (p.r.b.s.). They are produced by a simple synchronous shift register or counter with feedback from various stages determining the state of the first stage on receipt of each clock pulse. The feedback is basically by means of exclusive-OR gates (modulo-two gates). The basic diagram is as shown in Fig. 1. the output being taken from any stage. The sequence produced is cyclic and repeats itself after  $2^{n}-1$  pulses, the all-zero state being avoided. The maximum number of states for an *n*-stage register is 2" but to prevent the all-zero state becoming a permanent

state requires considerable extra logic and, hence, this state is avoided and the length  $2^{n}-1$  is described as maximal. Long sequences can be generated by simple feedback arrangements. The Table indicates the simplest feedback arrangements for all those registers up to length 18, and all those beyond 18 and up to 33 which require only one exclusive-OR gate. Fig. 2 shows how the characteristic polynomials of the Table are interpreted in terms of hardware for the particular case of n = 8. If JK flip-flops are used as the register stages some simplification is possible if the polynomial contains D to the power one. Column 3 or the table indicates the logic necessary for the J input and column 4 indicates the K input. If the output of stage 1 is regarded as the output, versions of this sequence delayed by  $L\Delta T$ , where  $\Delta T$  is the clock period and L = 0 to n-1, are clearly available from the

remaining stages. Delays of up to NAT, where  $N = 2^n - 1$  can also be generated by modulotwo addition of several of the output stages (ref. 1). When the sequence is being used as a noise source the output is arranged so that logic 1 = +a volts and logic 0 = -a volts. The r.m.s. value of the waveform is than  $a^{2}$  and the mean value is a/N (this would be zero if  $N = 2^n$ rather than  $2^{n}-1$ ). The power spectrum, which is discrete, is where G(k) is the power in (volt)<sup>2</sup> of the kth harmonic. G(k) is shown in Fig. 3. The small d.c. term is often ignored.

Spacing between the lines of Fig. 3 is  $1/N\Delta T$  and hence the power density spectrum (power per unit bandwidth) is  $G(k)N\Delta T$ . The 3-dB point for the power density spectrum occurs at approximately  $1/3\Delta T$ so that for systems with bandwidth less than  $1/3\Delta T$  the signal appears as white noise. Further, the autocorrelation function for the signal is very similar to that for white noise (ref. 1). The probability distribution of the signal is not at all Gaussian because it consists of two lines at  $\pm a$ 

respectively. However, when passed through a suitable R-C filter with a break point less than  $1/3\Delta T$  the distribution does become close to Gaussian i.e. that for band-limited white noise (ref. 2).

#### References

1. Davies, W. D. T., Generation and properties of M-sequences, *Control*, June, July and August 1966.

2. Roberts, P. D. & Davis, R. H., Some statistical properties of smoothed M-sequences, *Proc. I.E.E.* 1966 vol. 113, p.190.

#### Further reading

Davies, A. C., Design of feedback shift registers and other synchronous counters, *Radio and Electronic Engineer*, April 1969. Peterson, W. W., Error Correcting Codes, MIT Technical Press 1961. Golomb, S. W., Shift Register Sequences, Holden Day 1967.





 $G(k) = \frac{a^3}{N^3} + \sum 2a^3 \frac{(N+1)}{N^2} \left( \frac{\sin \frac{\pi \kappa}{N}}{\frac{\pi k}{N}} \right)^2$ 

2

# Set 14: Digital counters—12

### Glossary: flip-flops and b.c.d. codes

Integrated circuit flip-flops are usually clocked i.e. the change of state is initiated by a timing pulse called the clock pulse, The outputs are commonly termed Q and  $\overline{Q}$ . If Q = 1,  $\bar{\mathbf{Q}} = \mathbf{0}$  (and vice versa). These states are dependent on the logical states of the flip-flop (or bistable) inputs and this dependence is shown in each associated TRUTH table, where  $Q_n$  is the state of the O output after the nth pulse, and  $Q_{n+1}$  is the new state after the next pulse.

When Q is made to be logical '1', the flip-flop is said to have been SET, and when Q is made '0', the flip-flop has been **RESET or CLEARED.** Edge-triggered and master-slave types are available. The time for which data must be present before the clock pulse threshold (set-up time) and the time for which data must be maintained after the clock edge (hold time) are normally specified. The transfer of information in the master-slave flip-flop is according to the numbers marked on the pulse shown, and it may be considered that the master and slave flip-flops are distinct, but isolated or connected by gates.

- (1) slave isolated from master
- (2) data entered into master
- (3) master is isolated from input terminals
- (4) data is transferred from master-to-slave



R-S flip-flop



R = S = 1 must be avoided because the logical value of the Q output is uncertain.





This has two data inputs termed J and K (and may be considered to be similar to S and R of the R-S flip-flop), but no indeterminate state exists for any combination of

#### Further reading

Walters, D. J. Integrated Circuit Systems, Iliffe Books, 1971.

Malmstadt & Elke, Digital Electronics for Scientists, W. A. Benjamin, 1969. Texas Instruments Inc. Designing with TTL Integrated Circuits.

Cross references Series 14, card 5. Series 14, card 3. the inputs. Multi-input J and K terminals are achieved on some i.cs with internal gating, where the J and K inputs are for example the ANDED inputs  $J_1 J_2 J_3$  and  $K_1 K_2 K_3$ . CP is



the clock-pulse terminal, Preset and Clear terminals may also be available. These inputs are maintained normally at logical '1'. A negative edge applied to PRESET, sets Q = 1, and if applied to CLEAR, makes  $\ddot{Q} = 1$ . The negative edge triggering can be indicated by the small circle at the terminal as in diagram.

#### **D-type flip-flop**



This has one data input, and may also have both outputs available. In the table, D is the input before clocking and  $Q_{n+1}$  is the Q output after clocking.

Note—CMOS flip-flops are cleared by a positive-going edge at the clear input.

#### **T-type flip-flop**



This may be considered as the basic binary or toggle flip-flop. When T = 0, the Q output

will not change state on the occurrence of a clock pulse. If T = 1, Q takes up the opposite state when the flip-flop is clocked.



#### **Decimal codes**

These are listed with the least significant bit (LSB) rightmost for the weighted codes, i.e. those in which a decimal numerical weighting is assigned to each bit position.

JOHNSON	NATURAL BCD 8-4-2-1	4-2-2-1	GRAY	EXCESS-3	NUMBER
00000	0000	0000	0000	0011	0
00001	0001	0001	0001	0100	1
00011	0010	0010	0011	0101	2
00111	0011	0011	0010	0110	3
01111	0100	1000	0110	0111	4 5
1 1 1 1 1	0101	0111	0111	1000	
11110	0110	1100	0101	1001	6
11100	0111	1101	0100	1010	7
11000	1000	1110	1100	1011	8
10000	1001	1111	1101	1100	9

# Set 14: Digital counters Up-date

1. Control primarily depends on the logic simulation of the CD4048 gate where, if  $K_{\rm B} = K_{\rm C} = 0$ ,  $K_{\rm A} = 1$ , the gate acts like an eight-input AND gate i.e. output is 1 if all inputs are high. Also if  $K_{\rm A}=0$ , the gate acts like an eight-input NOR gate, and the output is 1 if all inputs are logic 0. Using exclusive-OR gates between the counter and the CD4048, the counter can then be arranged to cycle between any two symmetrical binary states by programming inputs B to H inclusive, e.g. if BCDEFGH is programmed

2. The odd synchronous counter provides symmetrical high and low output levels although driven from an unsymmetrical clock pulse source. Flip-flop A is driven from an inverted waveform and hence will change state, depending on the D input, at a clock pulse rising edge, one clock pulse width earlier than any changes that occur in flip-flops B and C. Observe at clock pulse marked tw, flip-flop C changes at the leading edge of the clock

COUNTER ISH UP/DOWN MODE CONTROL 1 = UP 0 - DOWN

to be 0001111, then when the counter reaches 11110000. the CD4048 will see all logic Is at the inputs and provide an output which can be used to clock the D-type flip-flop, i.e. the maximum output is 240<sub>10</sub>. When the counter then starts to count down, an output is again obtained when the counter reaches 00001111, where the exclusive-OR gates will provide all 0s to the gate-inputs and as KA is then 0. The effective NOR gate senses the 0s to provide a changeover signal.



pulse and flip-flop edge changes at the trailing, since it is driven via the inverter. The minimum clock pulse allowable depends on the propagation delay of flip-flop C and the data set up time of flip-flop 1. Figures quoted are 45 and 20ns respectively giving a minimum pulse width of 65ns.

#### Reference

Greenberg, M. Ideas for Design, *Electronic Design 20*, Sept. 27, 1974.

3. This arrangement of a 4-bit comparator and a 4-bit synchronous counter provides a programmable variable modulus counter in the range decimal 2 to 15. A binary number N is set up at the preset inputs B<sub>0</sub> to B<sub>3</sub> (where  $B_0$  is the least significant bit), i.e. over the range 0001 to 0111. The clock input causes the counter to reach the number (N+1), where the counter outputs compared with the programmed comparator value will show equality by generating an output pulse at the carry terminal. This is used to reset

the counter to zero, and the sequence repeats. For

initialising the counter, gate  $G_2$  ensures that the



exponential rise of voltage that will occur across C when the load terminal is driven low at commencement, will provide a fast logic level change at gate  $G_3$  when the capacitance voltage reaches the threshold level of  $G_2$ . Cascade connexions of comparators allow an extension of the modulus to 256 for two stages, the carry from the first counter being ANDed with the driving clock pulse.

#### Reference

Panzitta, J. Ideas for Design, Electronic Design 11, May 24, 1974.



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# Set 15: Pulse modulators

The following article is intended to introduce techniques of conveying analogue information using pulse-train modulation. It describes all the important methods of coding pulse trains and discusses their properties. If you have no previous experience of pulse modulation, we recommend reading this concise summary.

Various ways of producing amplitude and durationmodulated trains are given on cards 1-4, 6, 7 & 10, including two cards whose circuits use the 555 timers. Card 8 is not really a modulator—it gives d.c. motor control schemes using pulse modulated trains. The results are especially interesting in that the motor is operated well outside its normal operating conditions. Card 11 shows how position modulated pulses are derived. Two techniques, relying on a monostable to convert p.d.m. to p.p.m., are described in the article (Figs. 7 & 8), the card giving appropriate waveforms and a circuit.

In pulse-code modulation, card 12, p.a.m. is first produced and quantized, linearly or otherwise, and then coded in a binary way. The attraction of the scheme is improved signal-to-noise ratio over the other methods (the noise hierarchy goes from p.a.m. to p.d.m. to p.p.m. to p.c.m.). Card 9 gives two modulator circuits that produce a train of binary pulses relating to the rate of change of modulating signal between sampling instants, as opposed to instantaneous values, for differential p.c.m. or delta modulation.

Two signals can be modulated onto a single pulse train in the variable slope modulator of card 5, by modifying the slope of both leading and trailing edges. A modification shows that by restricting the dynamic range of the two signals, a third may be modulated as amplitude.

Pulse amplitude modulator with precision limiter 1 IC pulse duration modulator 2 Pulse amplitude modulator with shunt gate 3 Pulse duration/position modulator 4 Variable slope modulator 5 Pulse modulator using 555 timer 6 CMOS pulse amplitude/duration modulator 7 DC motor control using p.d.m. 8 Delta modulators 9 DC amplifier/pulse duration modulator 10 Pulse position modulator 11 Pulse code modulator 12 Systems using amplitude modulation, frequency modulation and phase modulation use a sinusoidal carrier which has one of its characteristics—amplitude, frequency or phase—varied under the control of an analogue signal of message x(t). In pulse modulation systems the carrier may be considered to be a periodic, rectangular pulse train c(t) as shown in Fig. 1. The pulse train has an amplitude  $A_c$ , a pulse width  $\tau_c$  and a fundamental frequency  $f_c = 1/T_c$ .

Amplitude, width or position (in time) of the carrier pulses may be varied as a function of the lower-frequency analogue signal waveform to produce pulse modulation, p.a.m., pulse width (duration) modulation, p.d.m., or pulse position modulation, p.p.m. For all practical applications of pulse modulation, the analogue modulating signal should be band-limited with a suitable filter so that it contains a reasonably well-defined upper frequency component  $f_m$  which is the highest significant frequency in the modulating waveform.

Each pulse in a modulated carrier pulse train y(t) contains one sample of information regarding the "instantaneous" value of the modulating signal x(t). The sampling theorem states that all the information in x(t) will be preserved if the samples are evenly spaced and occur at a rate that is not less than  $2f_m$ . In practice the sampling rate normally exceeds this minimum value by a factor of at least about 1.2.

Pulse amplitude modulation may be considered to be the process of changing the amplitude of a periodic rectangular carrier pulse train in synchronism with, and in proportion to, the instantaneous variations of the modulating analogue signal. Using a sinewave as an example of the modulating signal the p.a.m. waveform shown in Fig. 2 is obtained. The basic form of a pulse amplitude modulator is shown in Fig. 3.

The unmodulated carrier pulse train shown in Fig. 1 has a Fourier series representation

$$c(t) = \frac{A_c \tau_c}{T_c} \left[ 1 + 2 \sum_{n=1}^{\infty} \left\{ \frac{\sin(\omega_n \tau_c/2)}{\omega_n \tau_c/2} \right\} \cos \omega_n t \right]$$

where  $\omega_n = 2_n / \tau_c$  and  $n = 1, 2, 3 \dots \infty$ .

With a sinusoidal modulating signal of  $A \sin \omega t$ , the amplitude of the pulses become  $A_c(1 + m_A \sin \omega t)$  where  $m_A$  is the modulation index  $A/A_c$  which must be  $\leq 1$  to avoid overmodulation of the pulse train.

The p.a.m. wave shown in Fig. 2 may be considered either as a sinewave amplitude modulating the pulse train, or as the unmodulated carrier sampling the amplitude function  $A_c(1 + m_A \sin \omega t)$  over an interval of  $\tau$ , at a rate equal to the p.r.f. of the pulse train,  $f_c$ . The frequency spectrum of the p.a.m. signal may be seen conceptually by considering initially that of the unmodulated carrier which will contain a d.c. component and a theoretically infinite series of sinusoids at  $f_c$ ,  $2f_c$ ,  $3f_c$ , etc, which have diminishing amplitude with increasing frequency.



Fig. 1. Carrier pulse nomenclature.

Fig. 2. A p.a.m. waveform—a modulated pulse or a sampled input signal.

Fig. 3. The basic pulseamplitude modulator.





Fig. 4 (top). In this example of p.d.m., both edges are shifted, modulating the pulse symmetrically.

Fig. 5. The input signal x(t) varies the threshold of a triangular pulse from the integrator to produce p.d.m.

The amplitude modulation will produce a lower sideband and an upper sideband to each of the carrier components  $f_c$ ,  $2f_c$ , etc and a component at the original modulating frequency or band of frequencies. This latter component will have an amplitude of  $A_c m_A \tau / T_c$  and may be recovered by passing the p.a.m. wave through a low-pass filter, which also passes the d.c. component. P.a.m. are required to preserve the signals shape of the modulated pulses and therefore suffer from the same signal-to-noise ratio restrictions as a.m. but with a transmission bandwidth requirement of approximately  $3/\tau_c$  Hz. Consequently p.a.m. is more commonly used as a part of the signal processing in other systems than as a system in its own right.

Pulse duration modulation may be considered as the process of changing the width of the pulses in a periodic, rectangular carrier pulse train in synchronism with, and in proportion to, the instantaneous variations of the modulating analogue signal. In p.d.m., either the leading edges or the trailing edges or both edges of the carrier pulse train may be shifted in time due to the modulation. Fig. 4 shows an example of a p.d.m. wave where both edges have been shifted and

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the time interval between the centres of successive pulses remains fixed at  $T_c$ . With a sinusoidal modulating signal the unmodulated pulse width  $\tau_c$  becomes  $\tau_c (1+m_w \sin \omega t)$  where  $m_w$  is the modulation index  $\tau_m/\tau_c$  which must be  $\leq 1$  as the pulses cannot have a negative width,  $\tau_m$  being the maximum deviation of the pulse width from its unmodulated value  $\tau_c$ .

The p.d.m. wave contains a d.c. component, the original modulating frequencies and the harmonic series of carrier frequencies each of which has an infinite number of sideband pairs associated with it. In practice, not more than about three of these sideband pairs has significant amplitude so the original analogue signal may be recovered by passing the p.d.m. wave through a low-pass filter provided the carrier pulse train has a sufficiently high p.r.f. compared with  $f_m$ . The signal-to-noise ratio obtainable with p.d.m. is greater than that obtainable with p.a.m. due to the use of a wider transbandwidth, the improvement mission being similar to that of phase modulation compared with amplitude modulation. One form of pulse duration modulator is shown in Fig. 5.

Pulse position modulation may be considered to be the process of varying the position in time of the pulses in a periodic, rectangular, carrier pulse train in synchronism with, and in proportion to, the instantaneous variations of the modulating analogue signal. These modulated pulses cannot be advanced in time so that they may be considered as being displaced continuously in time with respect to the positions in time where the unmodulated





pulses would have occurred as indicated in Fig. 6. Pulse position modulation can be produced by generating a p.d.m. signal and feeding it into a monostable multivibrator. Figs 7 & 8 show two methods of generating a p.p.m. signal by using a pulse train p(t) having a normal leading or trailing edge with a negative-or-positive slope ramp. The method of Fig. 8 avoids the need to generate a p.a.m. signal as part of the process.

With a sinusoidal modulating signal the "instantaneous" position of the pulses in time may be represented by  $z(t) = f_c t + m_p \sin\omega t$  where  $m_p$  is the modulation index  $T_m/T_c$ ,  $T_m$  being the peak deviation of pulses from their unmodulated position. The p.p.m. signal may be converted back to p.d.m. or p.a.m. to recover the modulating signal by means of a low-pass filter. Because noise has less effect on the position of pulses compared with its effect on the amplitude or edges of pulses, p.p.m. can provide a better



Fig. 7 (top). A pulse-position modulator. Fig. 8. The p.a.m. is dispensed with in this system.



Fig. 9. Combined amplitude and width

modulation for two information channels.



Fig. 10. Three-channel operation.



Fig. 11. A pulse-code modulator.

signal-to-noise ratio than either p.a.m. or p.d.m.

Sometimes it is useful to convey two modulating signals on the same pulse train without resorting to more complex time-division multiplexing techniques. For example, it is possible to first pulse-width modulate a pulse train with one signal and then pulse-amplitude modulate the p.d.m. wave with a second signal, the resulting p.d.m.-p.a.m. wave appearing like that shown in Fig. 9. Three signals may be made to modulate the same pulse train by using one source to vary the slope of the leading edge, one source to modulate the slope of the trailing edge and a third source to modulate the amplitude of the resulting pulses, as indicated in Fig. 10.1

In pulse code modulation the analogue signal to be encoded is band-limited and passed to a sampling gate to produce p.a.m. The p.a.m. signal can have any instantaneous value within its allowed dynamic range and these variations are converted to a finite number of allowed levels by a quantizer. This process introduces an error into the signal, producing quantization noise which places a fundamental limit on the achievable signal-tonoise ratio. Each quantized p.a.m. sample is then encoded into a group of binary pulses to produce the p.c.m. signal. The number of pulses in each code group is determined by the number of allowed levels in the quantization scheme. For example, speech that has been bandlimited to 0.3 to 3.4kHz and sampled at a rate of 8kHz is commonly represented by a 128-level quantization scheme so that 7-bit code group is required since a  $128 = 2^7$ .

To avoid transmission of the d.c. component in the binary coded pulses the p.c.m. signal is often converted to a bipolar form. Because only the presence or absence of a pulse, rather than its shape, needs to be determined at the decoder a transmission bandwidth of only about  $1/\tau_c$  is sufficient, where  $\tau_c$  is the width of a pulse within a code group. The binary signals can be identically regenerated during transmission, so the overall signal-to-noise ratio obtainable with p.c.m. is much greater than that with the modulation methods already discussed.

With speech signals a further improvement is obtained by compressing its dynamic range at the encoder and expanding it at the decoder, a process known as companding. The basic processing in a p.c.m. transmitter is shown in Fig. 11 and that for a time-division multiplexed p.c.m. transmitter in Fig. 12.

Delta modulation  $(\Delta m)$ , or differential p.c.m., does not transmit pulses related to the instantaneous value of the modulating signal at a sampling instant, but uses a one-bit code to convey information about the rate of change of the modulating signal between successive samples. The greater the rate of change of the analogue modulating signal the greater is the repetition rate of the output pulses.

The basic form of a delta modulator is shown in Fig. 13, where the pulse modulator transmits pulses from the pulse source to the integrator with one polarity if the comparator output is negative and with opposite polarity if the comparator output is positive. Thus the output from the integrator is a stepped waveform that "oscillates" about the continuous input signal waveform x(t), and always attempts to keep this difference at a small value.

As a delta modulator transmits information about the rate of change of the analogue signal, an overload condition can be reached if the analogue signal changes too rapidly for the successive pulses, of fixed amplitude, to follow the change. The modulating signal can be recovered by feeding the delta modulated signal into an integrator followed by a lowpass filter. Although only a 1-bit code is used in delta modulation, the sampling or pulse rate has a minimum value that is higher than that required by the sampling theorem for p.c.m. For similar performance delta modulation generally needs a wider bandwidth than p.c.m. and the signal-to-noise ratio decreases with increasing frequency. However, the circuitry required in the encoder and the decoder is much simpler than for p.c.m.



Fig. 12. Multi-channel operation in p.c.m. using time-division multiplex.



Fig. 13. The basic delta modulator.

### Set 15: Pulse modulators—1

### Pulse amplitude modulator with precision limiter



#### **Circuit description**

In most pulse amplitude modulator realizations the unmodulated carrier is in the form of a low duty-cycle, unidirectional pulse train. So with a symmetrical modulating signal the amplitude of the carrier pulses may be varied over the maximum range of zero to twice their unmodulated value with 100% modulation. The narrower the carrier pulses the greater the conservation of power but this is achieved at the expense of a wider transmission bandwidth for defined performance, as the shape of the pulse tops must be preserved.

A common method of producing p.a.m. is by the use of a diode bridge that allows and prevents transmission of the modulating signal under the control of a carrier switching pulse train. For many applications this technique is acceptable although its accuracy is determined by the characteristics of the diodes. When more precise p.a.m. is required the effects of the diodes can be greatly reduced by using a precision full-wave rectifier, as shown above which effectively reduces the forward diode p.d. by the open loop gain of the operational amplifier A1.

When  $V_{in1}$  is a unidirectional positive pulse train,  $V_{in2}$  is a symmetrical modulating signal and  $V_m$  is negative, the output  $(V_x)$  from  $A_1$  is zero when  $(V_{in2}+V_m)$  is less than  $V_{in1}$ and is  $-(V_{in1}+V_{in2}+V_m)$  when  $(V_{in2} + V_m)$  exceeds  $V_{in1}$ . The p.a.m. output from the summing amplifier  $A_2$  is  $-V_{in1}$ when  $V_x = 0$  and is  $(V_{in2} + V_m)$ when  $V_x$  is less than zero.

#### **Component changes**

Useful range of supply about  $\pm 4$  to  $\pm 18V$  with suitable adjustment of  $V_{1n_1}$ ,  $V_{1n_2}$  and  $V_M$  levels.

 $V_{in1}$  (max): 7.5-V pulses  $V_{in1}$  (min): 3.5-Vpulses  $|V_M$  (max)|: 3.6V  $|V_M$  (min)|: 2.8V Max. p.r.f. of  $V_{in1}$ : 70kHz (mark-space ratio of approximately unity required to retain output p.a.m. waveshape).

#### **Circuit modifications**

To produce a wholly positive pulse amplitude modulated output waveform instead of a purely negative output, the following changes should be made:  $V_{1n_1}$  is changed to a unidirectional negative pulse train, diodes  $D_1$  and  $D_2$  are connected with reverse polarity and  $R_7$  is connected to the positive supply rail. With a negative p.a.m. output



Typical performance Supply:  $\pm 15V$ , +3.4mA, -17mAA<sub>1</sub>, A<sub>2</sub>: 741 R<sub>1</sub> to R<sub>6</sub>, R<sub>8</sub>: 10k $\Omega$ ; R<sub>7</sub>: 1k $\Omega$ linear D<sub>1</sub>, D<sub>2</sub>: PS101 V<sub>in1</sub>: 4-V positive pulse train, p.r.f. 10kHz, m-s ratio 1: 10 V<sub>in2</sub>: 1-V pk-pk sinewave, f = 1kHzV<sub>m</sub>: -3.2V V<sub>out</sub>: p.a.m. output—see diagrams

signal the alternative biasing arrangement shown above may be used, i.e. a d.c. bias source can be connected in series with the modulation source,  $V_{in2}$ , and  $R_7$  and  $R_8$ removed. To produce a purely positive p.a.m. output waveform with this arrangement, the polarity of the  $V_m$  bias source and of  $D_1$ and  $D_2$  is reversed as well as that of the carrier pulse train  $V_{in1}$ .

#### Further reading

Graeme, J. G., Tobey, G. E. and Huelsman, L. P. Operational Amplifiers----Design and Applications, McGraw-Hill, 1971, pp.400/1. Applications Manual for Operational Amplifiers, Philbrick/Nexus, 1965. Cattermole, K. W., Transistor Circuits, 2nd edition, Heywood 1964.

Cross references Series 15, cards 3 & 7. Series 4, card 3.



# Set 15: Pulse modulators—2

### IC pulse duration modulator



#### **Typical performance** Supply: $\pm 15V$ , $\pm 12mA$ IC<sub>1</sub>: 748, IC<sub>2</sub>: 311 R<sub>1</sub>, R<sub>2</sub>, R<sub>4</sub>, R<sub>5</sub>: 1k $\Omega$

R<sub>1</sub>, R<sub>2</sub>, R<sub>4</sub>, R<sub>5</sub>: 1k $\Omega$ R<sub>5</sub>: 100k $\Omega$ , R<sub>6</sub>: 2.2k $\Omega$ C<sub>1</sub>: 33nF, C<sub>2</sub>: 1nF, C<sub>5</sub>: 30pF V<sub>in1</sub>: 400mV pk-pk square wave at 100kHz V<sub>out</sub>: 28V pk-pk

#### **Circuit description**

In pulse width modulation, the unmodulated pulse train is often in the form of a square wave having a constant pulse repetition rate. The duty cycle of this square wave is then varied under the control of a d.c. or low-frequency modulating signal. The source waveform of the carrier signal may take various forms, e.g. a sinewave which is converted to a square wave for application to an integrator and comparator. The integrator converts the square wave to a triangular wave which is applied to one input of the comparator. In the absence of modulation, which is applied to the other input of the comparator, the output is a periodic pulse-train or square wave having a fixed duty cycle.

With a modulating signal applied to the second comparator input, the duty cycle of the output pulse train varies in sympathy with changes in the modulating waveform's instantaneous value, as the comparator's output state changes when the modulation input level exceeds or falls below that of the square wave applied to its other input. The above diagram shows an integrated circuit version using a voltage comparator and a general-purpose, externallycompensated operational amplifier for the integrator. Capacitor  $C_1$  removes any d.c. component from the input square wave and R<sub>3</sub> provides d.c. negative feedback to define the mean output voltage of the

triangular wave. The overall linearity obtainable is a function of the linearity of the triangular waveform applied to the comparator.

#### **Component changes**

Supply variation:  $\pm 2.2$  to  $\pm 18V$  $f_{max} \approx 200$ kHz with components shown. Change integrator time constant for different carrier p.r.fs. Unidirectional pulse width modulation can be produced by feeding the integrator with a train of narrow pulses and hence a sawtooth is applied to the comparator. The comparator may be fed directly from a triangular or sawtooth wave source.

#### **Circuit modifications**

If source voltage is large, the input to the integrator may be clamped using a pair of back-to-back diodes (PS101,

# 1N914, etc.) as shown left.

When the available carrier source waveform is in the form of a sine wave it may be converted to a suitable form for application to the integrator by the circuit shown centre which amplifies and clips the output signal applied to the integrator. The input coupling capacitor of the original integrator can be dispensed with. Suitable components could be IC<sub>3</sub>: 741; D<sub>3</sub>: small silicon diode;  $R_7$ : 100k $\Omega$ ;  $R_8, R_9: 1.5k\Omega; R_{10}, R_{11}: 1k\Omega.$ Resistors  $R_8$  and  $R_9$  can be trimmed to provide a symmetrical triangular wave at the output of the integrator. Instead of allowing the comparator's output to swing between approximately  $\pm V$  its excursions may be limited by including a pair of back-to-back zener diodes as shown right.

#### Further reading

Eimbinder, J. (Ed.), Linear



IC's: Theory and Applications, Wiley, 1968, pp.15-7. Graeme, J. D. and Tobey, G. E., Operational Amplifiers. McGraw-Hill, 1971, pp.412/3.

#### Cross references

Series 15, cards 4, 6, 7, 8, 10 & 11. Series 2, card 1. Series 3, card 1.



## Set 15: Pulse modulators—5

### Variable slope modulator



Typical performance Supplies:  $\pm 25V$ ,  $\pm 2mA$ , -5mA $\pm V_{1}, \pm 6V, \pm 14mA, -13mA$  $+V_{2}=+3V, 14mA$ Tr<sub>1</sub>, Tr<sub>4</sub>, Tr<sub>5</sub>, Tr<sub>7</sub>: BC125 Tr<sub>2</sub>, Tr<sub>3</sub>, Tr<sub>6</sub>: BC126 Diodes: PS101; R1, R4, R8,  $10k\Omega$  $R_2, R_7: 2.7k\Omega; R_3, R_6: 1.5k\Omega$  $R_5: 27k\Omega$ R<sub>10</sub>; 100Ω; R<sub>10</sub>: 820Ω  $C_1$  to  $C_4$ : 50 $\mu$ F  $C_5: \ln F; V_x + 8.2V$  $V_y - 8.8V$ vins 1V pk-pk square wave, p.r.f.: 40kHz See vout waveform opposite for vin1 and vin2 available range



#### **Circuit description**

When a capacitor is charged or discharged with a constant current, the p.d. across it changes linearly with time. In the circuit shown the unmodulated output pulse train is in the form of a trapezoidal wave having the same p.r.f. as the input square wave ( $V_{1n_3}$ ) and having leading and trailing edge slopes determined by  $V_x$  and  $V_y$ respectively.

When  $V_{in3}$  goes positive,  $Tr_3$ is switched on and passes the constant current from  $Tr_1$  to charge  $C_8$ . The magnitude of this current, and hence the rate of rise of  $V_{C_5}$ , is set by  $R_1$ and is linearly related to  $V_x$ . When  $V_{in3}$  goes negative,  $Tr_5$ discharges  $C_5$  with a constant current through  $Tr_6$ . The magnitude of this current is set by  $R_8$  and is linearly related to  $V_y$ .

Thus the slopes of the leading and trailing edges of the pulsetrain output waveform from the emitter follower Tr, are controlled by  $V_x$  and  $V_y$ respectively, which can be made to vary independently by sources of modulation Vini and  $V_{in2}$ . The p.r.f. of the carrier pulse train Vins is much greater than the modulating frequencies, so the square-wave half-cycles may be considered to be alternately sampling the signals Vin1 and Vin2. During the sampling intervals, the instantaneous values of these signals therefore vary the

slopes of the output waveform about their mean positions set by  $R_1$  and  $R_8$ . Positive peaks of  $V_{out}$  are clamped by  $D_4$ and  $+V_2$ .

#### **Component changes**

 $v_{in_1}(\max) = v_{in_2}(\max) \approx$ 760mV (1kHz) Minimum frequency of  $v_{in1}$  and  $v_{1n_2} < 10Hz$  $v_{ins}$  (min)  $\approx$  700mV pk-pk Max. p.r.f. of  $v_{in_3} \approx 200 \text{kHz}$ with  $\hat{C}_{s} = 47 \text{pF}$ Min. duty cycle of  $v_{1n_3}$  30% Min. load resistance  $\approx 220\Omega$ Vary  $+V_2$  to set vout positive peak in range 0 to +3.5VVary vout leading edge slope with  $+V_1$  in range +3 to +6.6V Vary Vout trailing edge slope with  $-V_1$  in range -6 to -7.5V

Reduce  $R_3$  to reduce rise time of  $v_{out}$  Reduce  $R_6$  to reduce fall time of  $v_{out}$ 

#### **Circuit modifications**

If the dynamic range of the modulating signals is restricted, the variations in the slopes of the leading and trailing edges of the output waveform will never be sufficient to cause the latter to assume a "triangular" pulse shape. With this restriction, Vout is a constantamplitude trapezoidal pulse train with its positive peaks clamped by  $D_4$  at  $+V_2$ . Hence, a third signal  $(V_{in4})$  may be superposed on the  $+V_2$  supply to provide trapezoidal-pulse amplitude modulation as shown left with Vout assuming the variations indicated below the circuit. One method of modulating the clamp voltage is shown right which uses an emitter-follower. All channels

are no longer independent of each other since the maximum value of  $V_{C_5}$  is determined by the charging current to  $C_5$ which in turn depends on  $V_{1n_1}$ . Transistor  $Tr_8$  may be a general-purpose type, typical component values being:  $V_3 + 9$ ;  $R_{11}$  100k $\Omega$ ;  $R_{18}$  47k $\Omega$ ;  $R_{13}$ ,  $R_{14}$  470k $\Omega$ ;  $C_6$ ,  $C_7$ 50 $\mu$ F.

#### Further reading

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Kruse, E. K. and Dobbs, D. Triple-channel modulation of a single pulse train. *Electronic Engineering*, April 1971, pp.36/7.

Cross reference Series 15, card 7.



# Set 15: Pulse modulators-

### Pulse modulation using 555 timer



#### **Circuit description**

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Modulation of pulse period or repetition-rate may be carried out independently by control of the threshold voltages or the charging rate of the timing capacitor in astable circuits. In the 555 timer a constantcurrent charging C gives a linear ramp. Upper threshold is equal to the potential at pin 5, vmod, and the lower threshold to  $v_{mod}/2$  by virtue of the internal potential divider. The capacitor is thus charged through  $v_{mod}/2$  at a constant rate, giving a period that is a linear function of the modulation potential provided that the discharge time is very short. This indicates a low value for R<sub>2</sub>, which may be reduced to zero (except where large values of C would result in the 200mA current limit of the discharge transistor being exceeded for long periods). The constant current is critical to the linearity, since for the usual astable with a resistor from pin 7 to  $V_s$ , the charging rate varies throughout the cycle. For example, as v<sub>mod</sub> approaches V<sub>s</sub> the fractional increase in period is greater than that of the modulation voltage. One possible constant-current circuit is the enhanced current mirror. This requires a low terminal p.d. for the constantcurrent transistor Tr<sub>2</sub>, i.e. the constant-current stage does not impose a limit on the upper threshold/modulation voltage that can be used. Since the charging current is proportional to the supply voltage (ignoring V<sub>be</sub> effects) the frequency

stability is supply-dependent, in contrast with the basic mode where increased threshold voltages and charging currents neutralize each others' effects on frequency.

#### **Component changes**

V<sub>s</sub>: +4.5 to +18V C: 100p to 100 $\mu$ F. Taken in conjunction with suitable charging currents, the frequency range extends from  $\ll$  1Hz to > 100kHz.

I: May be as low as  $1\mu$ A for very long periods, but should exceed  $10\mu$ A for reasonable stability of period. Up to 10mA permissible for high frequency generation.

 $R_2$ : Should limit discharge current at pin 7 to 200mA. May be reduced to zero for low values of C in most cases, though peak currents exceed this rating for very short times. This reduces duration of lowTypical performance IC: NE555V Supply: +12VC: 4.7nF $R_2$ :  $1k\Omega$ ,  $I=100\mu A^*$  $v_{mod}$ : 5VPeriod:  $150\mu s$ Output: Duration of low-state  $6\mu s$ , for load resistance  $1k\Omega$ \*Current may be provided by any constant-current circuit; that shown is one example for which Tr<sub>1</sub> to Tr<sub>3</sub> are elements of IC type CA3084;  $R_1=47k\Omega$ .

state of output towards zero.  $V_{mod}$ : For Vs of 12V,  $v_{mod}$ may range from 1.5 to 9V. The minimum value is of the same order at other supply voltages, while the upper is about 70% of Vs or greater for  $V_s > 10V$ . At low supply voltages internal Vbe drops restrict the range further.

#### **Circuit modifications**

In circuits based on capacitor charging to define the period of the waveform, then constant current charging linearizes the waveform and, for defined switching points, the period will be inverse to the current, i.e. the frequency will be a linear function of the current. A simple adaptation is to apply the modulation to the current generator—a simpler current mirror is shown as an example with the diode p.d. introducing



an offset to the vmod/frequency graph together with some drift. Again,  $R_2 \rightarrow 0$  minimizes flyback-time errors. If a clock generator is applied to the monostable in Fig. 2 then the output is at the clock frequency but with a pulse width controlled by the modulation voltage, i.e. p.d.m. Again, for linear modulation the waveform at C has to be linearized by the addition of a constant-current stage. The clock interconnection from a previous 555 may be as shown. The circuit may be conveniently implemented with a dual 555, but any other astable giving a negative-going edge approaching supplyvoltage magnitude may be used.

#### **Cross references**

Series 15, cards 2, 4, 7, 8, 10 & 11. Series 3, card 9.



Set 15: Pulse modulators-7

### CMOS pulse amplitude/duration modulator



#### **Circuit description**

In the circuit shown a c.m.o.s. Schmitt-trigger circuit is formed by using  $A_1$  and  $A_2$  as a cascaded pair of inverters with positive feedback. In the absence of modulation,  $(V_{in_2}=0)$  the Schmitt switching action is determined by the ratio  $R_3/R_1$  and  $V_{in_1}$ . Provided  $R_3/R_1$  is less than the forward gain in the linear region of the inverters, the switching action of the Schmitt follows the threshold crossings of the triangular wave input, Vin1. With a 10-V supply the Schmitt switches to  $+V_{DD}$ when Vini exceeds about +2.4V and switches back to  $V_{SS}$  (0V) when  $V_{in1}$  falls below about +1.85V. For the purposes of pulsewidth modulation it may be required to produce an unmodulated square-wave output (Vout<sub>1</sub>) and this may be obtained by superposing Vini on a suitable d.c. bias (+2.1V)to produce unity mark-to-space ratio. This ratio may then be varied by causing the switching times of the Schmitt to be controlled by the p.d.m. signal  $(V_{in_2})$  which is conveniently fed to the  $A_1$  input through  $R_2$ . The p.d.m. signal is fed to  $Tr_1$  which provides a similar output waveform but with its positive peak amplitude determined by the voltage to which  $R_4$  is returned. Hence by returning  $R_4$  to a second source of modulation (Vina) a waveform (Vout2) results which is simultaneously modulated in both amplitude and duration.

### Component changes

Useful range of  $V_{DD}$  + 3 to + 15V

Typical performance

 $A_1, A_2: \frac{1}{3} \times CD4007$ 

R<sub>3</sub>: 10M $\Omega$ ; R<sub>4</sub>: 10k $\Omega$ v<sub>in1</sub>: 11V pk-pk 50kHz

vina: Pulse-amplitude

a square wave

of +5V

 $V_{DD}$ : +10V, 830 $\mu$ A; V<sub>SS</sub>: 0V

 $Tr_1: 1/6 \times CD4007; R_1, R_2: 1M\Omega$ 

triangular wave superposed on +2.1V d.c. bias to make  $v_{out_1}$ 

 $v_{1n_2}$ : Pulse-width modulation source is 500mV pk-pk at 1kHz

modulation source 2V pk-pk at 2kHz superposed on a d.c. bias

Maximum useful  $v_{1n_1}$  frequency 150kHz Maximum pulse width modulation is achieved with  $v_{1n_2} \approx 900$ mV pk-pk Maximum pulse amplitude modulation is obtained with  $V_{1n_3} \approx 12$ V pk-pk superposed on a d.c. bias of +9V

#### Circuit modifications

The d.c. bias on which  $V_{in1}$  is superposed to provide a square-wave output may be dispensed with if a unity mark-to-space ratio is not required. If a low-duty-cycle pulse train output is required, the d.c. bias may be removed and the amplitude of the triangular wave (Vin1) reduced so that it is only slightly in excess of the upper threshold level of the Schmitt but sufficient to allow modulation. An approximately square-wave output can be obtained without a d.c. bias if the input triangular wave has a much larger amplitude. When a d.c. bias is used to control the unmodulated mark-to-space ratio of Vout, a constant-current source may be used for this purpose as shown above. One method is to use an integrated circuit current mirror to provide the constant-current bias to set the unmodulated duty cycle as Shown right. Negative feedback obtained by the inclusion of the emitter resistor  $R_{\delta}$  raises the output

impedance of the current mirror above that of a common-emitter stage. Pulse duration modulation may be obtained by controlling  $I_R$  with the modulation signal.

#### Further reading

Schmidt, B. Schmitt trigger design uses CMOS logic, *Electronic Design*, vol. 20, April 1972, p.72. Hart, B. L. Current generators, *Wireless World*, vol. 76, 1970, pp.511-4.



### Cross references

Series 15, cards 1 to 6, 8 & 10. Series 2, card 3. Series 6, card 4.

### DC motor control using p.d.m.



#### Performance data Supply: +40V $D_1$ : 1N4004 $Tr_1$ : BFR41 $Tr_2$ : TIP3055 R: $1k\Omega$ $V_p$ : 6VPulse frequency: 40 per sec Motor: 240V, 0.1-h.p. 6500 rev/min universal motor



#### **Circuit description**

Circuit shows a pulse driven high-current switch Tr1 and Tr<sub>2</sub> controlling the voltage applied to a motor. Basic principle involved is identical to that for thyristor driven motors viz that the average applied voltage controls the motor speed. During the pulse mark time Tr<sub>2</sub> conducts and the supply is able to supply current to the motor and during the pulse space time Tr<sub>2</sub> does not conduct and the supply is unable to deliver current. Clearly the greater the mark to space ratio of the pulse train the greater is the average applied voltage and with it the average motor current, Im. Graphs show the results obtained. The linearity of these results, despite the fact that the motor was being used well outside its specifications, indicates the potential usefulness of the scheme.

Values of V<sub>p</sub> and R are not critical so long as they provide sufficient base drive to Tr<sub>1</sub> to effect satisfactory switching and at the same time do not destroy Tr<sub>1</sub>. In this case since we are switching currents less than 200mA and the current gain of Tr<sub>1</sub> and Tr<sub>2</sub> is greater than 1000 then the base drive to Tr<sub>1</sub> should be of the order of 0.2mA. Considerably less may suffice. The pulse frequency is not critical either. Maintaining a constant mark-space ratio, i.e. maintaining a constant average voltage, the motor ran at the same speed when the frequency was varied from 40Hz up to at least 4kHz. At

higher frequencies lack of switching speed in the transistor caused the motor speed to change.

Diode  $D_1$  is the diode normally necessary with motors to prevent damage due to Ldi/dteffects.

Motor speed control by means of a voltage can be obtained by using the p.d.m. section of the p.p.m. shown in card 4; alternatively, the discrete p.d.m. shown in card 10 can be used with alterations to allow for the use of n-p-n rather than p-n-p transistors. A possible closed-loop speed control scheme is shown left. This will, of course, reduce the effects of nonlinearities, disturbances, etc. If the motor and p.d.m. are known in advance the maximum value of e is fixed and this effectively dictates  $v_{in}$ ,  $k_T$  and the differencing amplifier.

Diagram right shows a position control system; the output transducer need not, of course, be a potentiometer. This scheme has a considerable advantage in performance terms over conventional continuous control systems, because the steady state error in response to a step input in the presence of coulomb friction is eliminated. This is important in small motors in which brush friction is frequently a large effect when compared with the maximum torque developed by the motor. In continuous systems the torque produced is proportional to the actuating signal 'a' and when this torque is less than the coulomb friction torque the motor shaft stops. Hence, 'a' can be non zero. However, if 'a' is feeding a p.d.m. as shown right, the motor develops maximum torque so long as 'a' is non zero, albeit for shorter and shorter intervals as 'a' reduces. Since the maximum torque is greater than the coulomb friction torque the motor can only come to rest when 'a' is zero.

#### Further reading

Ghonaimy, M. A. R. and Aly, G. M. Phase-plane method for analysis of pulse-width modulated control systems. *International Journal of Control*, vol. 16, no. 4, 1972, pp.737-50. Pulse-width modulation for d.c. motor speed control, *Semiconductors* (Motorola) vol. 2, no. 2, 1971, pp.36-8.

#### Cross references

Series 15, cards 2, 4, 6, 7 & 10.



# Set 15: Pulse modulators-8

# Set 15: Pulse modulators-9

### **Delta modulators**



#### Circuit description

A delta modulator encodes an analogue signal into a train of binary pulses that represent the difference between the levels of the input signal at successive sampling times. The encoded pulse train has a repetition rate governed by the rate of change of the analogue signal; the greater this gradient the higher the density of the output pulses produced. The delta modulator shown above employs a monostable multivibrator with C5 controlling the monostable period and C4 acting as a 'speed-up" capacitor. The voltage follower  $A_1$  is used as a low-output-impedance buffer between the modulation source and the junction of C<sub>2</sub> and R<sub>2</sub>, R<sub>8</sub> being included to reduce ringing on the pulses at this junction.

In the stable state  $Tr_1$  is on and Tr<sub>2</sub> is off so that a complementary Vout waveform is fed to  $C_3$  via  $R_2$ . This signal is added to the modulating signal at the junction of  $R_2$  and  $C_2$ . This composite voltage is compared with a threshold voltage to determine whether the monostable will be triggered to its quasi-stable state. If the composite voltage across C<sub>3</sub> is less than the threshold, the differential clock pulses successfully trigger the monostable via  $D_1$  until the voltage across C<sub>3</sub> is raised sufficiently to exceed the threshold. When this occurs the clock pulses fail to trigger the monostable circuit and C<sub>3</sub> discharges until the modulating signal input exceeds the voltage on C<sub>3</sub>.



#### **Component changes**

+V<sub>CC</sub> (min): +7.4V -V<sub>BB</sub> is non-critical: ensures Tr<sub>2</sub> off-state Max. p.r.f. with above components: 130kHz v<sub>1n1</sub> (min): 8.8V pk-pk with 1 $\mu$ s pulse width and v<sub>1n2</sub> 2V pk-pk Min. pulse width of v<sub>1n1</sub>  $\approx$ 900ns, with 10V pk-pk amplitude and v<sub>1n2</sub> 2V pk-pk Adjust C<sub>5</sub> for required v<sub>out</sub> period

#### **Circuit modification**

Another monostable form of delta modulator using an integrated circuit is shown above where the width of the output binary pulses is controlled by  $C_3$ . The analogue signal to be encoded  $(V_{1n_2})$  and the clock pulses  $(V_{1n_1})$  are fed to the junction of  $R_1$  and  $D_1$  via capacitors  $C_1$  and  $C_2$  respectively.



As  $V_{1n_3}$  would normally be a low-output-impedance source the complementary output voltage waveform (pin 6) is integrated by  $R_1$  and  $C_1$ . Clock pulses are fed to the monostable via  $D_1$  after being differentiated by  $C_2$  and  $R_1$ .

If the junction of  $C_1$  and  $R_1$  is at a voltage below the threshold set by the amplitude of the clock pulses, their differentiated edges will trigger the monostable, producing a positive pulse into  $R_1$ . If this junction voltage is above the threshold the clock pulses are prevented from triggering the monostable. Junction threshold voltage is correctly adjusted by varying the amplitude of the clock pulses to cause the output pulse rate to be half the clock pulse rate. Typical components are: monostable—DTµL9951  $R_1 \ 10k\Omega; C_1 \ 50nF; C_2 \ 50pF.$ 

#### Further reading

Steele, R. and Thomas, M. W. S. Two-transistor delta modulator, *Electronic Engineering*, Sept. 1968, pp.513-6.

#### Cross reference

Series 15, card 11.



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50kHz, duty cycle 10%  $v_{1n_2}$ : 2V pk-pk sinewave at 1kHz  $v_{out}$  (unmodulated) 0 to +12V pulses, p.r.f. 50kHz, m-s ratio 1:3 See overload characteristic

# Set 15: Pulse modulators-10

### DC amplifier/pulse duration modulator



#### **Circuit description**

In the above circuit  $C_1$ ,  $R_1$  and  $D_1$  act as a d.c. restorer where the peaks of the sawtooth wave (Vin1) are clamped to a level determined by Ving. In the usual application of such a circuit,  $V_{ing}$  is a fixed d.c. level and provided that the time constant  $C_1R_1$  is very much greater than the periodic time of  $V_{in1}$  the latter will be clamped to the desired level. If a modulating signal, which varies much more slowly than V<sub>1n1</sub>, is used in place of a fixed value of Ving the level to which the peaks of the sawtooth wave is clamped will be controlled by the variations in Ving. Thus, the voltage appearing at the junction of C<sub>1</sub> and R<sub>1</sub> is effectively due to the addition of Vin1 and Vin2. This combined signal is applied to the input of a two-stage, high-gain, d.c. amplifier containing Tr1 and Tr<sub>2</sub> which has the same form as a Schmitt trigger but with the hysteresis removed. When the input to this amplifier is large enough to change the state of  $Tr_1$  an output pulse is obtained at Tr<sub>2</sub> collector having an amplitude almost equal to that of the  $-V_{CC}$  supply. As the switching threshold is controlled by the level of the modulating signal the duration of the Vout pulses is linearly related to  $V_{1n_2}$  over a wide range of the latter--superposed on a suitable negative d.c. bias. Hence, Vout is a pulse duration modulated pulse train.

#### **Component changes**

With a restricted range of pulse width variation  $-V_{CC}$  and

 $+V_{BB}$  have useful minimum values of about -2 and +2V respectively.

#### **Circuit modifications**

If required, the modulating signal Ving may be superposed on a positive d.c. bias if the polarity of  $D_1$  is reversed. If n-p-n transistors are used in the d.c. amplifier, polarity of the supplies and that of  $D_1$ should be reversed. A different approach to linear pulse-duration modulation is shown here. This circuit uses a monostable multivibrator to set the width of the unmodulated output pulses which have a repetition rate determined by that of the input positive pulse train, V<sub>in1</sub>. Output pulse width is a linear

function of the modulating signal (Ving) applied to the base of Tr<sub>2</sub> which serves as a constant-current generator. With  $Tr_1$  off and  $Tr_4$  on and saturated  $D_1$  is reverse-biased, if  $V_{in_2} < V_{CC}$ , and  $C_1$  charges. When  $Tr_1$  is switched on by a  $V_{in1}$  pulse the charge on  $C_1$ changes at a rate determined by the constant-current transistor Tr<sub>2</sub> until the base-emitter voltage of Tr<sub>4</sub> rises sufficiently to switch Tr. on and hence Tr<sub>1</sub> off. Capacitor  $C_1$  is now isolated from  $Tr_1$ collector which therefore switches off rapidly. Typical performance is indicated left, with  $+V_{cc}$ : +24V;  $R_1, R_9: 750\Omega; R_2: 7.2k\Omega;$  $R_3: 1k\Omega; R_4: 30\Omega; R_5, R_7:$ 9.1k $\Omega$ ; R<sub>8</sub>: 100k $\Omega$ ; R<sub>8</sub>: 3k $\Omega$ ; C<sub>1</sub>: 1nF.,

Typical performance

 $+V_{BB}$ : +12V 1.8mA Tr<sub>1</sub>, Tr<sub>2</sub>: BC126

D<sub>1</sub>: PS101; R<sub>1</sub>: 100kΩ R<sub>2</sub>: 22kΩ; R<sub>3</sub>: 2.7kΩ

C<sub>3</sub>: 470pF

50kHz

Supplies: -Vcc: -12V 9.7mA

 $R_4$ : 470Ω;  $R_5$ : 6.8kΩ;  $R_6$ : 1kΩ  $C_1$ : 100nF;  $C_2$ : 100pF

vin1: 3V pk-pk sawtooth at



#### Further reading

Hart, E., Generator permits infinite pulse-width variation, in "100 Ideas for Design", no. 5, Hayden, 1965, p.5. Hemingway, T. K., Electronic Designer's Handbook, Business Publications, 1967, pp.17/9. Hughes, R. S. Pulse width vs. control voltage made linear by generator in "100 Ideas for Design", no. 5, Hayden, 1965, p.76.

#### **Cross references**

Series 15, cards 2, 4, 6, 7, 8 & 11.



# Set 15: Pulse modulators—11

### **Pulse position modulator**



#### **Circuit description**

The pulse-position modulator employs a pulse-duration modulator feeding a t.t.l. integrated-circuit monostable package. Many different pulse-duration modulators could be used to drive the monostable provided that the output pulses are t.t.l.compatible. The one used was the d.c. amplifier type described in card 10. To provide the required t.t.l. compatibility, the circuit shown in card 10 was modified to use BC125 (n-p-n) transistors with supplies of  $+V_{CC} + 5V$  and  $-V_{BB} - 5V$  the polarity of  $D_1$ also being reversed. See card 10 for circuit description. The t.t.l. monostable package provides complementary output pulses which can be initiated in several ways. With the connections shown, input B (pin 5) is held high and input  $A_2$  (pin 4) which is unused is taken to  $+V_{\rm CC}$  through a 1-k $\Omega$  resistor. In this form the package acts as a monostable circuit providing an output pulse of defined width whenever input A<sub>1</sub> (pin 3) receives a logic-level negative-going trigger pulse. Width of the output pulses is determined by the value of C<sub>1</sub> with R<sub>2</sub> ensuring that this width is obtained accurately and repeatedly. Larger R<sub>2</sub> values for a given  $C_1$  will widen the output pulses. As the negative-going edges fed to the monostable circuit are

produced from a pulse duration modulator, times of occurrence will vary in

sympathy with the instantaneous values of V<sub>1n2</sub>. Hence the shift in time of the monostable output pulses is determined by Ving giving pulse-position modulation,

#### **Component changes**

Useful range of supply  $\approx +3.5$ to +5.25V (minimum value not guaranteed) Max. range of  $V_{1n_2} \approx 7V$  pk-pk superposed on a bias of +3.75Vgives pulse shift of  $\approx 14\mu s$  at p.p.m. output Change  $R_2$  and  $C_1$  for different output pulse widths

#### **Circuit modifications**

Pulse-position modulated signals may be produced by a variety of electronic circuits which normally perform the signal processing indicated left. In the upper diagram the modulating signal x(t) is added in a summing amplifier to a pulse train c(t) having a negative-slope ramp. The composite signal x(t) + c(t) is fed to a comparator having a fixed reference level which produces a p.d.m. output w(t) having the modulation on its trailing edge only. Applying w(t) to a monostable gives time-shifted constant-width pulses (p.p.m.) at its output, y(t).

In the lower diagram the modulating signal x(t) has been sampled at regular intervals to produce the flat-topped p.a.m. wave p(t). The same sampling pulses are used to trigger a generator producing a synchronous train of pulses c(t) having a negativeSupply: +5V, +23.5mA IC1: SN74121N  $R_1, R_2: 1k\Omega; C_1: 6.8nF$ Pulse duration modulator: see card 10 with Tr<sub>1</sub>, Tr<sub>2</sub> BC125 (n-p-n) transistors,  $+V_{CC}$ : +5V;  $-V_{BB}$ : -5V and  $D_1$ polarity reversed. V<sub>in1</sub>: 3V pk-pk sawtooth at 50kHz V<sub>in2</sub>: Sinewave modulation superposed on a direct bias of +3.75V to give square wave output from modulator

slope ramp. The c(t) and p(t) signals are added as before and fed to a comparator producing a p.d.m. output w(t) which in turn feeds a monostable to produce the p.p.m. output y(t). Reversing the slope of the ramp in c(t) produces leading-edge p.d.m. to feed to the monostable.

#### Further reading

TTL data book for design engineers, Texas, 1973, pp.82 & 134-7.





#### Cross references

Series 15, cards 2, 4, 6, 9, 10 & 12. Series 3, cards 2, 4, 6.

# **Pulse code modulator** to-D converte

Diagram shows in block form the basic processes used in a single-channel pulse-code modulator. The analogue signal to be encoded, x(t), is passed to a sampling gate via a low-pass filter which defines the bandwidth of the modulation. The sample pulses are of low duty cycle and have a constant p.r.f.  $(f_8)$  that is at least twice that of the highest modulating signal component (f<sub>m</sub>). In practice  $f_s > 2f_m$ , e.g. for speech that has been bandlimited to 0.3 to 3.4kHz,  $f_8 = 8$ kHz. The output of the sampler is a p.a.m. signal which has an infinite possible number of amplitudes that are quantized, uniformly or nonuniformly, into a finite number of allowed levels. Although each sample is converted to the nearest allowed level, quantization inherently introduces errors or quantization noise. Nonuniform quantization of speech produces an improvement in the signal-to-quantization noise ratio.

Each quantized sample of the p.a.m. wave is then encoded into a group of pulses according to a binary code, the number of pulses in each code group being determined by the number of allowed levels in the quantization scheme. For speech transmission 128 levels are normally used, hence a 7-bit code is used  $(2^7 = 128)$ . For transmission, the coded signal is normally converted to a bipolar form to avoid wasting transmitter power by sending a d.c. component containing no information. One such code is alternate-mark-inversion (a.m.i.) which is a pseudo-ternary code

with binary significance. See the works, and their bibliographies, listed under in further reading for detailed system and circuitry techniques. Diagram over shows an adaptive pulse code modulator for encoding speech. In this technique the coding signals change to track the changes in the "envelope" of the speech input x(t) after it has been bandlimited by the filter to 0.25 to 2.4kHz. The output is fed simultaneously to the voltage comparators A1, A1 and  $A_3$ ;  $A_1$  and  $A_3$  together provide updated amplitude information by comparing x(t) with a feedback voltage  $V_{f}(t)$ and its inverse respectively. Comparator  $A_1$  produces an output logic 1 when x(t) > $+ V_{f}(t)$  and  $A_{s}$  produces a logic 1 output when x(t) is more negative than  $-V_t(t)$ . The  $A_1$  and  $A_3$  outputs are fed to bistable circuit FF2 via an OR gate, the output of which is sampled at 4.8kHz. Thus, each amplitude information

bit at the FF2 output is a logic 0 when the x(t) sample is in the range  $-V_{\rm f}(t) < x(t) <$  $+ V_{f}(t)$  and is a logic 1 when x(t) is outside this range. The A<sub>2</sub> comparator provides x(t)-polarity information, producing a logic 1 at its output when x(t) is positive and a logic 0 when x(t) is negative. The A<sub>2</sub> output feeds bistable circuit FF1 which is sampled at 4.8kHz to produce polarity bits that are combined with the amplitude information bits in the multiplexer (M), which simply transmits its 2-channel inputs alternately at 9.6kbit/s.  $V_f(t)$  is obtained by feeding the FF2 output to a 10ms RC integrator giving a positive output  $V_{a}(t)$  to which is added a small d.c. bias  $(V_B)$  to ensure that  $V_f(t)$ never falls to zero.

#### Further reading

Cattermole, K. W., Principles of Pulse Code Modulation, Iliffe 1969. Sheingold, D. H., Analog-



**Cross reference** Series 15, card 11.



Set 15: Pulse modulators-12

# Set 15: Pulse modulators Up-date

1. Pulse modulation is not restricted in its application to communication systems and data processing. Amongst the other areas of application is that of switching power amplifiers or class D systems. Recently, sub-divisions of class D have been identified which simplify the filtering of the output (class BD) and a reduction in the switching losses in the power elements (class ABD). In the former version, a set of comparators are fed with a triangular reference waveform and the signal, with a unity gain inverter giving anti-phase drive to one pair of



comparator inputs. A pair of power output stages drives the load such that on large positive inputs comparator 1 output is high except at the positive peaks of the triangular reference wave; comparator 2 goes low at the negative triangular wave peaks. Hence  $S_A$  would be gated on except for small time intervals. Throughout the positive half-cycle of the input signal  $S_B$  would be held off. The

mean load voltage is thus proportional to the fraction of the time for which  $S_A$ conducts during positive inputs. Non-linearity of the triangular wave leads to distortion; push-pull operation reduces even harmonics. For inductive loads  $S_C$  must be closed when both  $S_A$  and  $S_B$  are off.

#### References

Martin, J. D. Class BD Amplifier Circuit, *Electronics Letters*, 1970, vol. 6, pp. 839–41. See also Daniels, A. R. and Slattery, D. Class ABD Amplifier, *Electronics Letters*, 1974, vol. 10, p. 364.

2. This circuit is aimed at producing positive halfsinusoidal envelopes with respect to zero voltage reference, with modulation at 100%.

For absolute rectification, the diodes are included in the feedback loop of the inverting operational amplifier. The gain constraint on the input pulses is then limited by the power supply levels of the op-amp. During the negative part of the modulating signal, the amplifier output swings positively by an amount depending on the ratio of



 $R_3/R_2$  plus the drop across D<sub>1</sub>. The input pulses are therefore forced to swing within this limit, but note that the pulse amplitude must be sufficiently great relative to the modulation amplitude to obtain 100%modulation. Overdriving the pulse level by 20% more than required is suggested. Full sine wave modulation can be obtained by adding another resistor to the summing input X, and applying an appropriate negative bias level.

#### Reference

Quick, D. Improve amplitude modulation of fast digital signals, *EDN*, Sept. 20, 1975.

3. D.c. potentiometers normally depend on the precise ratio of a pair of resistances; that ratio has to be variable over a wide range to give the high resolution needed. This circuit replaces the precise ratio of resistances by the ratio of time intervals. These intervals are controlled by a digital clock, and the clock frequency is not critical as it is only the ratio of the times for which the input switch is connected to inputs 1 and 3 respectively. As with conventional chopper-circuits

the switch should combine minimal input offset with maximum switching speed. The original paper uses a mechanical switch at the input, with a series-shunt switch at a later stage (not shown) to inhibit response to the switching transient. The final switch activates a sample-andhold circuit that stores the integrator output. The final



output thus has minimal ripple except during an input step to which it responds rapidly (99.9% within two sample periods is claimed). Additional applications noted include current comparison and cascaded voltage division from values as high as 1kV. The overall accuracy is claimed to be  $\pm (0.1 \text{ p.p.m.} + 5\mu\text{V})$ .

#### Reference

Sugiyama, T. & Yamaguchi, K. Pulsewidth modulation d.c. potentiometer, *IEEE Trans. Instrum. and Meas.*, 1970, pp. 286–90.



### Set 16: c.d.as — signal processing

Current differencing amplifiers, or Norton amplifiers, are an ingenious solution to the need for single-supply op-amps. As there is superficial similarity in c.d.a. circuits to normal op-amp circuits, the following article warns against the transposing op-amp circuits into c.d.a. form, drawing a parallel with the days when the transistor was first introduced. Those new to circuits of the LM3900 kind should therefore make a point of reading the articles preceding sets 16 and 17, which introduce the ideas behind c.d.as in a gradual and easily assimilable way.

The first article provides a simple look at the c.d.a. circuit, the amplifier being considered as a d.c. feedback pair, fed by a current mirror. It is the current mirror that is responsible for the amplifier's name, in that it is used in such a way that the input to the feedback pair is the difference between two input *currents*. It is, in effect, the current dual to the voltagedifferencing action of the long-tailed pair used in op-amps. For the next stage of circuit elaboration see card 1, which shows incorporation of an improved emitter follower output stage and an additional transistor which only operates when the input is over-driven. A third stage of elaboration is described in the article preceding set 17. Some circuits can nevertheless be "borrowed" from

conventional op-amp practice, for example, the bandpass and band stop filters shown on cards 7 & 8. But notice that the Sallen & Key filters cannot, because it isn't possible to use series applied feedback (but see the final circuit modification on card 9). Card 10 of set 5 also gives c.d.a. circuits.

Current differencing amplifiers 1 Basic amplifiers—I 2 Basic amplifiers—II 3 Logic gates 4 High voltage amplifiers 5 Power amplifiers 6 Bandpass filters 7 Notch filters 8 Low pass/high pass filters 9 Gain-controlled amplifiers 10

### **Current differencing amplifiers**



# Set 16: c.d.as-signal processing-1

Three sets of cards deal with current differencing amplifiers of the LM3900 kind. This set covers signal processing applications, set 17 covers signal generation and a third set deals with various other circuits including test, measurement, detection, logic and driving circuits.

Set 16: c.d.as—signal processing—2

Typical performance Supply: 15V R<sub>L</sub>: 5kΩ Voltage gain: 2,800 (69dB) Output swing: 0.1 to 14.2V Output current: source 10mA, sink 1.3mA (overdriving inverting input increases sink current up to >30mA) Input current: 30nA Unity-gain bandwidth: 2.5MHz Slew rate: +0.5V/μs, -20V/μs

N.B. Data is for National

Circuit description Transistors  $Tr_1$ ,  $Tr_2$  are a current mirror with the Semiconductor LM3900. A similar amplifier is available from Motorola and other manufacturers are expected to "second-source" such devices. Refer to manufacturers data sheets particularly for maximum ratings. While other current-differencing amplifiers may be expected to have similar performance in the circuits to be described it is important that the ratings of particular devices are not exceeded.

collector current of Tr<sub>s</sub>, approximately equal to non-

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### **Basic amplifiers**—1



Circuit description-1 No direct current flows in R<sub>1</sub> and hence R<sub>2</sub>, R<sub>2</sub> determine the d.c. operating conditions. For perfect balance between the input circuit transistors they will carry equal current and for  $V_{be} \ll + V$ , the direct output voltage is given by  $R_{s}/R_{s} \times (+V)$ , i.e.  $R_{s} = 2R_{s}$  is the usual condition for maximum available voltage swing with the output biased at supply mid point. As there is no significant alternating current in R<sub>2</sub>, it is the alternating currents in R<sub>2</sub> and R<sub>1</sub> that are equal in magnitude

Typical performance Supply: +15VR<sub>1</sub>:  $100k\Omega$ R<sub>2</sub>:  $2.2M\Omega$ R<sub>3</sub>:  $1M\Omega$ C<sub>1</sub>:  $0.1\mu$ F Direct output voltage  $\approx 7V$ Voltage gain  $\approx -10$ 

while the virtual earth at the inverting input (though  $\approx 0.6V$ d.c.) gives a voltage gain of  $-R_s/R_1$ . The addition of reactive components modify the gain, so that a highfrequency roll-off is readily achieved by placing a capacitor across  $R_s$  (corner frequency  $1/2\pi R_s C$ ).

Capacitive coupling may be required to the load, while the reactance of  $C_1 \ll R_1$  at lowest frequency. Maximum resistance values of up to 10M may be used, but roll-off due to stray capacitances is likely.



#### Circuit description-2

The base-emitter voltage of the input transistor is  $\approx 0.55V$  at room temperature, falling by ≈2.5mV for every 1% rise in temperature and by less than this (typically 0.5 to 1mV) for each 1-V increase in the supply voltage. This voltage is thus sufficiently stable to be used as the reference voltage for setting the d.c. output conditions, and the technique may be called the "nVbe" biasing method. It is identical in principle to that used in the d.c. feedback pair and the "amplified-diode". If the source has an internal resistance to ground  $< R_1$  then direct coupling may be used

Typical performance Supply: +15VR<sub>1</sub>:  $470k\Omega$ R<sub>3</sub>:  $4.7M\Omega$ R<sub>2</sub>:  $470k\Omega$ Direct output voltage  $\approx 6V$ Voltage gain  $\approx -10$ 

with R<sub>1</sub> chosen to provide the required input resistance for the circuit and R<sub>2</sub> determining both the voltage gain and the direct output voltage. Resistor R<sub>2</sub> is omitted in this mode as is the input coupling capacitor. Direct output voltage  $\approx (R_3/R_1 + 1)V_{be}$ . Voltage gain  $\approx -R_3/R_1$ . The method requires modification both for high and low gains as the direct output voltage may not be convenient. By capacitive coupling to R<sub>1</sub> the d.c. and gain conditions can be made independent, with direct output voltage  $\approx (R_s/R_2 + 1)V_{be}$  and voltage gain  $\approx -R_{\rm s}/R_{\rm 1}$ .

inverting input current, subtracting from inverting input current at base of Tr<sub>a</sub>. The net input current to Tr<sub>3</sub> is (I-) - (I+) and this is amplified by Tr<sub>3</sub> with Tr<sub>5</sub>, Tr<sub>6</sub> forming an improved emitter follower output stage. Constant-current generators define the operating conditions while Tr, comes into action on over-driving the input to maximize the sink-current. Output depends on the difference between two positive input currents with negative feedback taken to the inverting input when the gain is to be defined. The noninverting input is outside the feedback loop, and behaves as a forward-biased p-n junction. With resistive negative feedback applied between output and inverting input, the direct currents in the two inputs will be equalized to within the accuracy of the current mirror. If the noninverting input current is defined by a resistor to +V. the direct output voltage is then a fixed fraction of +V. Transistor Tr<sub>3</sub> base current is  $\approx 30nA$ , allowing very low bias/signal currents, and like the voltage gain and output current capabilities is controlled over wide temperature and supply variations. An internal regulator (not shown) ensures this by providing the constant currents while also biasing a set of transistors that clamp each input to  $\approx -0.3V$  on negative input swings.

#### Further reading

Frederiksen, T. M., Howard, W. M., Sleeth, R. S., The LM3900—A New Current-Differencing Quad of  $\pm$ Input Amplifiers, National Semiconductor application note AN72. Frederiksen, T. M., Norton quad amplifier subtracts from costs, adds to design options, *Electronics*, Dec. 6, 1973, pp.116-20. Motorola Linear Integrated Circuit Data Book, pp.7-446, 7-453, 7-456 and 7-463; data sheets on MC3301P and MC3401P amplifiers. National Semiconductor. Linear Integrated Circuits, pp.226-33, data sheets on LM3900. Frederiksen, T. M., Howard, W. M., Sleeth, R. S., Use Current-mode IC amplifiers, Electronic Design, vol. 21, no. 2, Jan. 18, 1973, pp.48-55, Mortensen, H., Use a quad amplifier to handle transducer bridge signals, Electronic Design, vol. 21, no. 3, Feb. 1, 1973, pp.74-6.

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Circuit description-3 A third variation on the biasing methods available, is to take the bias resistor RL to a separate reference voltage Vret which can be decoupled to make the output voltage much less dependent on supply ripple. A single reference voltage (here equal to + V/2) may be used for a number of separate amplifiers separate control of the quiescent output conditions is by variation of R<sub>4</sub> for each amplifier while adjustment of R1, R2 varies all of them simultaneously. The amplifier is shown with the signal applied to the

Typical performance Supply: +15VR<sub>1</sub>:  $47k\Omega$ R<sub>2</sub>:  $47k\Omega$ R<sub>3</sub>:  $100k\Omega$ R<sub>4</sub>:  $1M\Omega$ R<sub>5</sub>:  $1M\Omega$ C<sub>1</sub>:  $0.1\mu$ F C<sub>2</sub>:  $1\mu$ F Direct output voltage  $\approx 7.5V$ Voltage gain  $\approx +9.5$ 

non-inverting input. No feedback is available at this input and so the impedance of the input transistor affects the input current. At room temperatures,  $r_1 \approx 0.026/I_{R4}$ , giving a value >3k $\Omega$  for the values shown. This reduces the gain to about 3% below the simple theoretical relationship  $R_5/R_3$ . This biasing method is equally applicable to the inverting amplifiers. **Circuit description**—4 The value of the feedback resistor is limited to a few megohms for several reasons (bias instability, effect of stray capacitance, noise and hum). If it is required to have a high input resistance and high voltage gain than the a.c. and d.c. feedback must be different. As shown, R4 and R3 constitute a potential divider for the output signal while only R<sub>3</sub> is involved in the d.c. feedback. Output voltage has a quiescent value of  $+ V(R_2 + R_3)/R_5$ . Voltage gain is  $\approx (-R_2/R_1)$  $(R_3/R_4 + 1)$ . Where  $R_2 = R_1$ , a convenient condition, the

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Typical performance Supply: +15VR<sub>1</sub>:  $1M\Omega$ R<sub>2</sub>:  $1M\Omega$ R<sub>3</sub>:  $1M\Omega$ R<sub>4</sub>:  $10k\Omega$ R<sub>5</sub>:  $2.2M\Omega$ C<sub>1</sub>:  $0.1\mu$ F C<sub>2</sub>:  $4.7\mu$ F Direct output voltage  $\approx$  7V Voltage gain  $\approx -95$ 

voltage gain simplifies to  $-(R_3/R_4 + 1)$ . However when the ratio  $R_3/R_4$  is large the feedback theory demands that the limited open-loop gain be taken into account. In practice, a ratio that should set the gain to -20 will do so to within about 1%, while a nominal gain of -100 would be nearer to -95.

Cross references Set 5, cards 5, 8, 9, 10. Set 7, cards 4, 10. Set 10, cards 1, 9. Set 16, cards 2, 5, 6, 10.

### **Basic amplifiers**—2



Circuit description—1 Common-mode signals are a problem when transmission over lines has to take place in a noisy environment. By coupling the signals through a transformer such common mode signals are minimized,

ile the anti-phase inputs of me current differencing amplifier offer a further improvement. Any commonmode voltage at the



transformer secondary produces equal currents at the two inputs largely cancelling each other because the gain at the two inputs is equal and opposite. R<sub>line</sub> is inserted to achieve the correct loading on the source with R<sub>1</sub>, R<sub>L</sub> sufficiently larger not to affect that loading. Quiescent output voltage  $\approx V(R_8/R_4)$ ; voltage gain  $\approx -R_4/R_1$ .





Typical performance Supply: +15VR<sub>1</sub>:  $10k\Omega$ R<sub>2</sub>:  $100k\Omega$ Direct output voltage  $\approx 6V$ Voltage gain  $\approx 11$ (transformer secondary to output)

**Circuit description**-2 Where the source is inductive or is to be transformercoupled, a variant of the "nV<sub>be</sub>" biasing method provides a simple solution. Again there is the restriction that the direct output voltage and the gain of the amplifier itself are controlled by the same resistor ratio but decoupling part of R<sub>1</sub> to ground can make the ratio for signal frequencies  $\gg$  the ratio at d.c. if required. In the extreme case, R<sub>1</sub> can be completely decoupled giving the full open-loop gain of the amplifier.

This coupled with the step-up turns ratio of the transformer gives a very high overall gain. As shown, the overall voltage gain is  $\approx n(R_2/R_1+1)$  and the quiescent output voltage is  $\approx (R_2/R_1+1)V_{\rm be}$ Because the input current required by the amplifier is very small ( $\ll 1\mu A$ ) the effective input impedance remains high regardless of the gain, and high step-up ratios are possible. This yields a very sensitive microphone amplifier though the noise performance is unlikely to allow use in audio applications.

# wireless world circard

**Typical performance** 

Output logic 0: 150mV

For inputs commoned, output

changes state for input voltage

Output logic 1: 19.2V

Supply: +20V

IC: 1 LM3900

 $\approx 20\%$  of +V.

R1: 150kΩ

R: 82kΩ

### Logic gates



#### **Circuit description**

Availability of two current inputs simplifies the design of basic logic gates with these amplifiers. For example, a low current at one input can hold the output in one desired state while the other input receives the sum of the currents from two or more inputs. This sum can be set to overcome the bias when only one input is high or only if all are simultaneously high, leading to OR and AND-type circuits respectively. (The amplifier is working as a high-gain comparator and can also Set 16: c.d.as—signal processing—4

provide a majority-gate in which any two out of three inputs are enough to provide the required output. By extension some of the simpler forms of threshold logic are possible by scaling the values of resistors to assign a different weight to their importance in decision making.) In the first configuration, if any input is high the current driven into the non-inverting input exceeds the inverting input current and the output is driven high, i.e. an OR gate. The remaining input resistors connected to logic 0 bypass a small portion of that current (<0.5V/R for each resistor) but unless the number of inputs is large and/or the supply voltage is low, this is not a problem. Speed of response is limited to  $\approx 0.5 V/\mu s$  for positive swings and up to  $20V/\mu s$  for negative swings though the fall in voltage is

slower as logic 0 is approached. By interchanging the inverting and non-inverting inputs with no change in component values, a NOR gate is produced. This flexibility of being able to produce different logic functions from the same package is very attractive. In addition, one or more of the amplifiers can be used to provide astable, Schmitt trigger functions, etc. for obtaining the appropriate waveforms with which to drive the gates.

#### **Component** changes

+V Normal voltage range is +4 to +36V, but some devices will operate to <3V without difficulty.

R<sub>1</sub>, R Ratio of these resistances is chosen to ensure that with the lowest expected value of logic 1 to any one input that, the resulting current flow into the non-inverting input is sufficient to overcome



Circuit description-3 The biasing techniques described above are easily adapted for dual supply operation, making the amplifier compatible with standard op.amp. ±15V systems. One method is to bias the non-inverting input from the centre rail, leaving the same p.d. across R<sub>2</sub> and R<sub>3</sub> if  $R_2 = R_3$ , i.e. the output is substantially at OV if the amplifier has well-matched inputs. This remains true for all values of positive and negative supply voltages including those cases where they are markedly unequal in value. R<sub>1</sub> plays no part in the d.c. conditions and is chosen



for the desired gain/input resistance. Voltage gain  $\approx -(R_2/R_1)$ . Non-inverting gain is obtained as before simply by transferring  $R_1$  to the non-inverting input. The gain is less well-defined but the bandwidth is increased, because there is then no resistor to attenuate the negative feedback, i.e. the bandwidth can approach that of the amplifier with 100% feedback while the gain can approach the value defined by  $(R_2/R_1)$ .



Circuit description-4 The present form of currentdifferencing amplifier is not suited to low-level d.c. amplification and cannot replace standard operational amplifiers directly. The adaptation shown has a reasonable performance for d.c. signals of >100mV provided the offset adjustment is made and subsequent changes in temperature and supply are restricted. Effective input of the amplifier is point A and for  $R_1 = R_2$  its potential is a OV in the absence of feedback (the

Typical performance Supply:  $\pm 15V$ R<sub>1</sub>, R<sub>2</sub>: 1M $\Omega$ R<sub>3</sub>: 820k $\Omega$ R<sub>4</sub>: 500k $\Omega$ R<sub>5</sub>: 220k $\Omega$ R<sub>6</sub>: 2.2M $\Omega$ Direct output voltage  $\approx 0V$ Voltage gain  $\approx -10$ 

> diode compensates for the input Vbe). Adjusting R4 sets the output voltage to 0V prior to the application of feedback with  $(R_3 + R_4) \approx (R_1 + R_2)/2$ . The resistors Rs, Rs then set the input resistance and gain as for previous amplifiers. Point A is no longer such a good equivalent to a virtual earth from the signal standpoint since the presence of R<sub>2</sub> lowers the amplifier gain, while  $R_2//R_1$  is the equivalent resistance to ground at this point.

Cross references Set 5, cards 5, 8, 9, 10. Set 7, cards 4, 10. Set 10, cards 1, 9. Set 16, cards 1, 5, 6, 10.

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inputs gives a NAND gate. By scaling the resistors in some of these and/or feeding signals to both inputs circuits which are intermediate between true analogue and true circuits become possible. (Fig. M2) • If it is required to have multiple inputs to an AND gate, simple diode gating may be used, Fig. M3, with the amplifier serving a similar function to the transistors in d.t.l., but with the advantage that the ratio of output to input current, which determines the fan-in fan-out capability, is of the order 10<sup>6</sup> (25nA-25mA). If any input is at logic 0 the current in  $R_1$  flows through that diode. All inputs have to be at logic 1 for the current to be diverted into the amplifier non-inverting input to give a logic 1 out. The threshold voltage and hence the noise margin is low in this circuit. If all inputs but one are at logic 1 then that input need only rise to  $\approx 0.5V$  to be recognized as logic 1. A NAND function is obtained by reversing the amplifier inputs.

Cross references Set 11, cards 1, 2, 11.





Fig. M1

the fixed current in the positive input. (Allow for 0.5V/R lost in each logic 0 input.) Typically  $R = R_1/2$  is a suitable starting point and R may vary from 10k to 10M $\Omega$ . Fan-out To maximize this

R, R<sub>1</sub> should be large but there is no great advantage to raising them above  $1M\Omega$ .

#### **Circuit modifications**

 The AND/NAND functions are slightly more difficult to construct for multiple inputs. This is because for an *n*-input gate the sum of the currents produced by (n-1) inputs at logic 1 differs little from that produced by n inputs if n is large. Up to three inputs can be used with a reasonable margin of safety as in Fig. M1, where the resistor R<sub>2</sub> reduces the current flow when two inputs are on. For a two-input AND gate R<sub>2</sub> may be omitted and the values of the input resistors adjusted. For example, with  $R_1 82k\Omega$ , R 100 or  $120k\Omega$ . • Again interchanging the inverting and non-inverting

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# High voltage amplifiers



#### **Circuit description**

Where a high output voltage swing is required the amplifier must be fed from a separate low-voltage supply, and a suitable high-voltage transistor employed to withstand the main supply voltage. The

nfiguration depends on aether it is the output voltage or current that is to be defined. If the former, then the feedback is taken from in shunt with the

load. For an inverting-gain amplifier and the transistor in the common-emitter mode, the inverting gain of the transistor necessitates that the feedback be applied to what is normally considered as the non-inverting input. For an input of 0V d.c., the current flow in R1 is small and that in R<sub>2</sub> is forced by the feedback to equal that in R4. If in this condition the output is desired to be +HT then



Typical performance Supply: +15V, +300V (HT) R1: 330kΩ R<sub>3</sub>: 10MΩ R<sub>8</sub>, R<sub>4</sub>: 470kΩ R5: 1kΩ C1: 100pF Voltage gain: -31.2 (d.c. to 1kHz) Cut-off frequency: 3kHz Output impedance:  $<10k\Omega$ (d.c. to 1kHz) Input impedance: 330kΩ (d.c. to 1kHz)

 $+HT/R_2 = +V/R_4$  gives the required value of R4. Overall voltage gain is given by  $(-R_2/R_1)$  as the circuit is effectively a "see-saw" amplifier while the input resistance is approximately R1. Resistance R<sub>5</sub> introduces a small amount of negative feedback into the output stage and raises the amplifier quiescent voltage well into its linear region. Capacitor C1 modifies the gain/frequency



characteristic to maintain stability at the higher loop gain. Output voltage swing can be up to 95% of the supply voltage if lightly loaded and the negative feedback keeps the output impedance reasonably low.

#### **Component** changes

R<sub>1</sub>, R<sub>2</sub> These set the input resistance and the voltage gain  $R_2 \gg R_3$  to minimize loading on

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### **Power amplifiers**



#### **Circuit description**

Addition of a Darlingtonconnected pair of transistors increases the output current capability from 10mA to 1-5A depending on the ratings of the transistors used. One restriction is that the Vbe's of the transistors limit the output voltage to around 2.5V below the supply level allowing for the amplifier internal saturation. The additional phase shifts that may occur even in an emitter

follower make external compensation desirable. For supply, input and output to be all positive, the configuration shown is adequate, where with  $R_2 = R_3$ , V<sub>o</sub> varies linearly with V<sub>1</sub> provided V<sub>1</sub> is above the amplifier internal Vbe. The relationship is then  $V_0 \approx V_{be}$  $+(R_4/R_1)(V_1 - V_{be})$  as the currents in R1 and R4 have to be equal. This means that as a d.c. amplifier it is of relatively low accuracy but is quite

suitable for supplying small d.c. motors under the control of a phase-locked loop. A combination of the techniques for increasing the current ratings and voltage could allow the production of high-power amplifiers. Replacing the emitter follower stages by common emitter amplifier increases the available positive output swing to within a hundred millivolts of the positive supply.

Tr<sub>2</sub>: TIP3055

IL of 0-1A

V1: 1.03V

Vo: 5.0V

Set 16: c.d.as-signal processing-



**Component** changes

R1 This sets the voltage gain in conjunction with R4. Because of the Vbe offset, the output voltage becomes temperature dependent particularly for V<sub>1</sub> comparable to Vbe, i.e. high gains are not compatible with good stability in this configuration. R1 100k to 10MΩ.

R<sub>2</sub>, R<sub>3</sub> These provide forward bias for each of the inputs allowing the output to be

the output.  $R_2 1M$  to  $22M\Omega$ . R<sub>3</sub> Load resistance, dictated by user requirements. For use as an r.c.-coupled amplifier, R<sub>s</sub> has to carry a quiescent current greater than peak load current required.

R4 Sets d.c. conditions at output. If  $V_0 = +HT$  required for  $V_{in} = 0$ , then  $+ V/R_4 =$  $+HT/R_2$ . If output quiescent to be +HT/2 as when used for amplifying a.c. input signal, then  $+ V/R_4 = + HT/2R_2$ . R<sub>5</sub> Not critical. Raises amplifier output quiescent voltage to 1 to 3V range, i.e. into linear region. Value

dependent on output quiescent current but might be  $50\Omega$  to 5kΩ. Chosen to suit amplifier.

+Vand available supplies (+4 to +36V).

+HT Dictated by load requirements. Tr<sub>1</sub> Must have voltage rating in excess of +HT particularly

if inductive loading possible.

**Circuit modifications** 

A non-inverting amplifier

controlled for inputs down to zero. This is best achieved for  $R_1 = R_4$  when  $V_0 = V_1$  is the first-order approximation, the Vbe effects at the two inputs cancelling.

R4 As suggested, R4 may equal R<sub>1</sub> for optimum stability but with gain restricted to +1.  $R_4 \ge 2.0R_1$ except for a.c. amplifiers where some drift in quiescent conditions is acceptable. **R**<sub>5</sub> 1k $\Omega$ . Together with C<sub>1</sub> is part of the suggested network for maintaining stability. Limits amplifier current under load short circuit conditions providing protection for amplifier and output stage. Not adequate unless proper heat-sinking used since limit of output current depends on transistors' current-gains and is ill defined.

C1, C2 Control high frequency performance. C1: 330p to 2.2nF, C2: 5.6 to 22pF. Choose lowest values giving stability under operating conditions.

R. Load resistance. This may



again using a common emitter output stage has the feedback applied to the amplifier non-inverting input, while the signal is applied to the inverting input (the inversion in the transistor ensuring the correct overall sign for the gain). For a transfer function that passes through the origin a pre-biasing network is required. Again the gain is set by the ratio of the feedback to the input resistor. For a high output-impedance, the feedback is taken from the transistor emitter. The method can be employed where the load is to be transformer coupled and the direct voltage

+300V

21M

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#### **Circuit modifications**

• A simpler circuit based on the same principles is Fig. M1. The limiting and compensation components have been eliminated together with the high-power transistor. The bias network shown is suitable for a source with resistive path to ground  $\ll 470 k\Omega$ , with the non-inverting input grounding. The bias method is then basically the "nVbe" method

as in the "amplified diode". Alternatively a capacitively coupled source may be used with no direct current in R<sub>2</sub> and  $R_1 \approx 2R_3$  to set  $V_0 \approx + V/_2$ . • For higher efficiency the usual Class B technique can be

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drop across the primary is too small to allow of the d.c. feedback. The two  $10-M\Omega$ resistors define the potential at the transistor emitter as about 2 Vbe and allows the output current to be fixed reasonably accurately, falling with rise in temperature. By decoupling the emitter and taking overall a.c. feedback from the load, the output impedance can still be made low if required. Increased output resistance is possible by using compound output stages of various kinds that can also increase the current capability, subject to device power limitations. An f.e.t. draws no current from the amplifier ensuring that the load and emitter resistor currents change together if the current Ibias is made constant, either because of the high supply voltage and correspondingly high resistance R, or by a separate low-voltage constant-current stage.

**Cross references** Set 7, cards 5, 7. Set 16, cards 1, 2, 6.

applied with a complementarysymmetry output (Fig. M2). Not recommended for low-distortion applications, since additional bias networks to overcome crossover problems would lose the advantage of simplicity normally offered by this amplifier.

• Although the supply voltage capability is good (up to 36V) the output swing can be extended to  $\approx$  70V (Fig. M3) using a bridge configuration. The simplest arrangement for a.c. signals has the  $1M\Omega$  and  $470k\Omega$  resistors setting the output quiescent voltages to + V/2 for maximum undistorted voltage swing. Replacing the two input resistors by a potentiometer and applying the signal via the pot to opposite-phase inputs gives anti-phase outputs that can be set for equal magnitude with an overall voltage gain of about 18.

**Cross references** Set 7, cards 2, 4, 7, 8, 10, 11. Set 16, card 5.

# Set 16: c.d.as-signal processing-7

### **Bandpass** filters

**Circuit description** 

applicable to current-

Active filter techniques based

on operational amplifiers are

shown is a direct adaptation of

differencing circuits. That

Circard 1, Set 1. A Wien

stwork is placed between

source and output with the

monitored by the amplifier

input terminal. Resistor R1 is a

potential at its junction



Typical performance R:  $47k\Omega$ C: 10nFR<sub>1</sub>, R<sub>2</sub>:  $1M\Omega$ R<sub>3</sub>:  $6.8M\Omega$ R<sub>4</sub>:  $100k\Omega$ Supply: +15Vf<sub>0</sub>: 320HzFor Q = 15 k = 0.640N.B. Onset of oscillation at k = 0.648 compared with theoretical value of 0.66.

high-value resistance to minimize loading on the network. Resistor  $R_2$  provides a variable amount of positive feedback to increase the Q of the circuit with minimal effect on the centre-frequency while  $R_3$  sets the quiescent output voltage to allow for maximum signal swing. Because  $R_2 > R_1$ would be required for stability, with  $R_3$  taken directly to the output, and variable high-value resistors are inconvenient, the alternative is to tap  $R_2$  onto a variable portion of the output voltage. The result is a band-pass filter with centre frequency given by  $f = 1/2\pi CR$ and with Q controlled by  $R_4$ . The limited gain and bandwidth capabilities of the amplifier does not allow the circuit to provide large stable Os nor



to operate successfully at frequencies  $\ge 10$ kHz. The input impedance falls as the Q is increased because of the increased output swing (gain  $\propto$  Q to a first order). It is also true that the sensitivity of such circuits to variations in component values is proportional to the Q, i.e. for Q = 10 a 1% change in a critical resistor might

# wireless world circard

### **Notch filters**



Typical performance Supply: +20VR<sub>1</sub>:  $100k\Omega$ R<sub>2</sub>, R<sub>3</sub>: 1MR<sub>4</sub>:  $1k\Omega$ R:  $47k\Omega$ C: 10nFNotch frequency: 319HzAchieved for  $m \approx 0.35$ for k = 0 and k = 1

Set 16: c.d.as—signal processing—8



#### **Circuit description**

The circuit is again derived from a Wien Bridge to demonstrate the principles by which known circuits can be adapted to current-differencing amplifiers. It provides a notch or null in the response at a frequency set by the RC values ( $f = 1/2\pi RC$ ) with R<sub>1</sub> providing a trimming action to get as true a null as possible. If  $R_4$  is a low resistance potentiometer, then the depth of the null is unaffected as the tapping point is varied, since both ends of the potentiometer are at zero for this condition. However, the positive feedback introduced has the effect off-null of sharpening up the response so that the gain remains close to unity for frequencies close to the null. This makes the circuit useful for nulling out the fundamental (or particular harmonic) of a complex waveform with minimal effect on the other harmonics. A weakness of this particular circuit is that it relies on the matching of the inverting and non inverting input sensitivities making it prone to variation with supply/

#### temperature changes, etc.

#### **Component changes**

R, C The difficulty with this particular circuit is that these should have a low impedance compared with  $R_3$  so that the latter does not load them, while not in turn being disturbed by the varying source resistance of  $R_4$  as k is varied. Typically R 10 to 100k $\Omega$  produce >10% variation in the Q itself.

#### **Component** changes

**R**, **C** The load on the output should not fall much below  $10k\Omega$  at any frequency to avoid loss of loop gain, distortion, etc. **R** 10k to  $1M\Omega$ , **C** 220p to  $10\mu$ F **R**<sub>1</sub> Used to set a convenient level of current at amplifier inputs; too low and loading on Wien network disturbs Q, fo; too high and stray capacitive coupling disturbs response. 220k to  $10M\Omega$ .

R<sub>2</sub> Convenient to set  $R_2 = R_1$ using R<sub>4</sub> to control feedback. Alternatively for fixed Q particularly if low, eliminate R<sub>4</sub> taking R<sub>2</sub> directly to output. Then  $R_2 > 1.5R_1$ .

**R**<sub>4</sub> Not critical, but  $\ge 10k\Omega$ and <**R**<sub>2</sub>.

 $R_3$  Sets quiescent output voltage for maximum swing. Should be chosen after other components have been selected for desired response. Typically 2 to  $10 \times R_1$ , but for a given value, small variations

with C to give frequency as  $f = 1/2\pi RC C$ , 220p to  $10\mu F$ . R<sub>2</sub>, R<sub>3</sub> 1 to  $10M\Omega$ . At higher frequencies stray capacitance effects bring phase shifts preventing good null. R<sub>1</sub> 10 to 250k $\Omega$ . R<sub>4</sub> 1 to  $10k\Omega$ .

This circuit experiences shift in the null because of supply temperature changes and requires careful adjustment before use if sharp notch required.

#### **Circuit modifications**

• T-networks normally have the disadvantage that to control the frequency while retaining a sharp notch requires adjustment of a number of interacting components. Interconnected with an amplifier (top circuit), using positive feedback to steepen the sides of the notch (equivalent to varying the Q but not the centre frequency), it has the advantage that variations in the amplifier gain only affect the steepness leaving the depth unaffected. A further advantage



in k and hence Q can be accommodated without serious bias changes.

#### Circuit modifications

• Multiple feedback circuits (Fig. M1) are the equivalent of the virtual earth circuits used with conventional operational

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amplifiers. With correct scaling

of resistors, capacitors Q > 1

with centre-frequency gain of

unity. Again both are strongly

is achieved simultaneously

dependent on component

stability at high-Q values.

• A better approach to

filter design with current-



accrues using the currentdifferencing amplifier in that the gain can be adjusted, while retaining a fixed input impedance, by varying  $R_r$ . This also affects the gain at frequencies well away from the notch, but the approach is often used where the gain is just less than unity the sides of the notch very steep and unity gain can be assumed off resonance. Independent adjustment of C and R are required to obtain a complete null and the system is easier to

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differencing amplifiers is to select passive networks whose transfer-function gives a defined output current. It is possible to adapt the Wien and similar networks, Fig. M2. Normally R, would be grounded and the p.d. across it fed back to a high impedance point on the amplifier. By feeding back the current in R<sub>2</sub> into what is almost equivalent to an a.c. virtual earth point, the overall transfer function can be varied more easily. Components  $R_{1,2} C_{1,2}$  determine  $f_0 = 1/2\pi (R_1 R_2 C_1 C_2)^{\frac{1}{2}}$ , while  $R_5$ sets the Q and R4 the quiescent output value.

• Any of the other RC networks used in bandpass filter designs are applicable and Fig. M3 gives almost identical performance for  $R_1 = R_2 = R$ ,  $C_1 = C_2 = C$ .

#### **Cross references**

Set 1, cards 1, 3, 6, 7, 8, 12. Set 5, cards 6, 10. Set 10, card 8. Set 16, cards 8, 9.

use if fine frequency control is possible at the oscillator, as may be possible when used as part of a distortion measuring system.

• Other passive networks have been described which give a good null at a specified frequency and though normally used as shown with a separate feedback attenuator and buffer, the network can also be used as above. By deriving the feedback directly from the output of the amplifier variable, resistance of the attenuator is avoided and the need for the buffer eliminated.

#### Further reading

Rowe, N. B. Designing a low frequency active notch filter, *Electronic Engineering*, April 1972.

Dance, J. L. & Edwards, K. H. Simple null filter with variable notch frequency, *Electronic Engineering*. vol. 36 1964, pp.478/9.

Cross references Set 1, cards 9, 10. Set 16, cards 7, 9.

### **Gain-controlled** amplifiers



# Set 16: c.d.as-signal processing-10



Ci

Vo



#### **Circuit description**

The non-inverting input has a diode-connected transistor as part of a current-mirror. For a given feedback resistor to the inverting input, the current flowing in the non-inverting input gives a proportional output voltage. The non-

earity of the input impedance ads to output waveform distortion if any low-resistance parallel path is used to attenuate the signal. If the

path consists of a suitablybiased silicon diode then its non-linearity is comparable with that of the input stage and the input signal division between the two paths has little variation with signal amplitude, i.e. little distortion results. This remains true while the signal current is well below the bias level.

As the direct current in the diode is increased, the corresponding fall in slope-

resistance by-passes more signal current reducing the gain. Resistor R4 provides sufficient bias current to set the output potential to a minimum of  $V_8/3$ . This applies when the bias voltage is low with the direct current in R<sub>3</sub> as well as the a.c. signal current via R1 are both shunted away through R2, D2. The gain is virtually zero in this condition. As VB approaches one diode voltage. the direct currents in D<sub>1</sub>, D<sub>2</sub>



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### Low pass/high pass filters



#### **Circuit description**

The most convenient configuration for most feedback circuits with these amplifiers is the virtual earth configuration; it is not possible to use series applied feedback as with standard Sallen & Key type filters. The network is not easy to analyse component-bycomponent because of the interactions between them but the overall transfer function is Typical performance Supply: 5V R<sub>1</sub>, R<sub>3</sub>: 150kΩ  $R_2: 47k\Omega$ C1: 470pF C2: 120pF -3dB point  $\approx$  8kHz (low frequency gain about -1) Max. input/output swing  $\approx 1V \text{ pk-pk}.$ 

well defined particularly if the gain in the pass-band is low, e.g. unity. Then it is sufficient if the amplifier voltage gain is >100 to have a transfer function that is very close to the theoretical value. A second-order low-pass filter results where the cut-off frequency may range from 1Hz to >100kHz. A convenient means of adjustment of the filter properties is via  $C_1$ ,  $C_2$ . It is their product that



ratio determines the shape of the transfer function (i.e. Butterworth, etc.). The output impedance is low and the input impedance may be up to  $1M\Omega$  if required allowing such filters to be cascaded with negligible loading. As shown the output voltage is defined as  $(R_s/R_1 + 1)V_{be}$ provided the input has a quiescent value of zero, as

would be achieved by capacitive coupling from the previous stage. Since  $R_1 = R_3$ is a convenient value for design of the filter characteristics this restricts the output to a quiescent value of about 1.2V regardless of the supply. This may be inconvenient where larger voltage swings are desired and alternative biasing schemes may be needed (see Circuit modifications).

# Set 16: c.d.as—signal processing—9



The bias range from 0 to 1V may be raised to any higher level by adding a potential divider.

#### **Component** changes

**R<sub>1</sub>** Sets input impedance **1k** to **1M** $\Omega$ . **R<sub>2</sub>** Not critical but  $\ll R_1$ . **R<sub>3</sub>**, **R<sub>4</sub>** Normally equal, setting limits to output quiescent voltage. **R<sub>3</sub>** = **R<sub>4</sub>** = 3**R<sub>5</sub>** for output levels ranging from V<sub>8</sub>/3 to 2V<sub>8</sub>/3.

**R**<sub>5</sub> Ratio  $R_5/R_1$  sets maximum value of voltage gain. **C**<sub>1</sub> Not critical, but reactance  $\ll R_1$  at lowest signal frequency. **D**<sub>1</sub>, **D**<sub>2</sub> Small signal silicon diodes.

Vs Whole voltage range of amplifier, 4 to 36V. Vn Centre of gain-control

region occurs for 0.6V but this is temperature dependent.

#### **Circuit variations**

• In some applications it may be more convenient to make the control voltage compatible with the normal output of

Component changes Transfer function (voltage gain) is

$$\frac{\overline{R_1 R_2 C_2 C_1}}{s^2 + \left(\frac{1}{R_2} + \frac{2}{R_1}\right)\frac{1}{C_1}s + \frac{1}{R_1 R_2 C_1 C_2}}$$

where  $s = j\omega$ 

provided  $R_1 = R_3$ . This is the basic form of a second-order low-pass filter giving  $\omega_n = 1/\sqrt{R_1R_2C_1C_2}$  and equivalent  $Q = 1/2\xi$  where  $\xi$  is the damping factor given by

$$Q = \frac{\sqrt{R_1 R_2}}{R_1 + R_2} \cdot \sqrt{\frac{C_1}{C_2}}$$

Setting R<sub>1</sub>, R<sub>2</sub> to determine input impedance, C<sub>1</sub> and C<sub>2</sub> may be varied to control cut-off frequency and shape of response. Note that the equations proposed in National Semiconductor applications note AN72 pp.14-15 show a circuit as having Q = 1, for which the above equations suggest  $Q \approx 0.4$ . Measurements



other amplifiers. The potential divider raises the range of bias voltages to 0 to 10V. The component values, Fig. M1, give comparable gain figures at much higher impedance levels (taken from manufacturers application notes).

• Automatic control of output voltage against input signal variations (a.g.c.) is possible by detecting the output

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Fig. M2 Fig. M3 Fig. M3 Fig. M3 Fig. M3 Fig. M3 Fig. M3

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support the equations quoted above.  $R_1, R_2$  10k to 1M $\Omega$ .  $C_1, C_2$  47p to 10 $\mu$ F.

Circuit modifications • If the filter is part of the system where it is convenient to bias input and output to +V/2 then the addition of resistor R<sub>4</sub> to the positive supply rail as shown permits this for  $R_2 + (R_1 + R_3/2) = R_4$ . The resistors R<sub>1</sub>, R<sub>2</sub>, R<sub>3</sub> are first selected to set the transfer function and the nearest amplitude and feeding a smoothed control signal back to the amplifier bias input. For best control the rectified output can be compared against a direct reference voltage with the amplified difference being fed back, Fig. M2.

• A simpler circuit, Fig. M3. uses a.c. coupling between the diode being controlled by R. and the amplifier non-inverting input. An alternative d.c. biasing technique sets the output voltage to  $(n + 1)V_{be}$  provided the value of R<sub>1</sub> is sufficiently low that the current in it  $(\approx 0.6 V/R_1)$  is  $\geq 30 nA$ , the typical input current of the amplifier. Increasing the bias voltage lowers the slope resistance of D<sub>1</sub>, diverting signal current from the amplifier input and reducing the gain (maximum gain  $\approx nR_1/R_2$ ). The gain cannot be reduced to zero with this circuit.

Cross references Set 16, cards 2, 3.

preferred value for  $R_4$  chosen using the given equation (Fig. M1.)

• The corresponding high-pass filter has the locations of the capacitors and resistors interchanged. Resistor  $R_2$  is the only one contributing bias current to the inverting input leaving  $R_4 = 2R_2$  as the condition for a quiescent output voltage of + V/2. (Fig. M2.)

• A current differencing amplifier can be adapted to behave as a voltage amplifier of defined gain—unity gain in the example, Fig. M3. R can be high in value and the loading effect on a separate passive network is light enough that filters similar to the Sallen & Key filters using voltage followers are possible.

Cross references Set 1, cards 4, 5, 6, 7, 11 Set 5, cards 2, 3, 7, 10. Set 16, cards 7, 8.

# **Current-differencing amplifiers**

signal processing circuits

In the process of improving on previous designs a stage is reached where further advances are won only with great difficulty. At that point the problem has to be changed if it is to be solved. Conventional operational amplifiers have reached a very high level of performance but the demands from industry were for still lower cost and for more functions in a given space. This could be met by packing four amplifiers into a standard 14-pin dual in-line package, using two pins for the power supplies and two inputs/one output for each amplifier. Unfortunately the chip size then increases to a level where yields fall off and the costsaving in the packaging is partially neutralized.

When this conclusion was set alongside the continuing demand for amplifiers capable of operating from a single supply such as car batteries, a discontinuity appeared in the design process—an integrated-circuit amplifier was designed that depended on the difference between two input currents for its performance rather than the voltage differencing action of the conventional operational amplifier with its input long-tailed pair. The resulting design is brilliant, simple in concept but very subtly implemented.

To take advantage of its novel characteristic the user must re-think his philosophy. Most functions that can be carried out with operational amplifiers can be carried out with current-differencing amplifiers; the configurations may look similar; the component values may be comparable. This surface conformity can obscure the underlying differences, and the user should be chary of attempting a direct transposition of familiar circuits. The problems are similar to those faced by valve designers when transistors first appearedtransistors could be forced to operate in the known valve circuits but alternative arrangements were soon found to exploit the unique properties of these cold and fragile new devices that lacked the warming glow of the familiar friends.

It is to be assumed that what we are now seeing is but the first generation of current-

differencing amplifiers, devices that seek to capture the largest possible market by accepting compromises in many areas. Before this article is published there may be high frequency/low noise/high power current-differencing amplifiers on sale, but the basic design techniques should be applicable to these just as users of the 709 can transfer their skills to the super- $\beta$ , infinite bandwidth, zero drift op-amps that every self-respecting manufacturer now seems able to offer. In this article two subcircuits are discussed which when combined give the prototypical form of the current-differencing amplifier now available as a quad amplifier in 14-pin dual in-line form.

One of these sub-circuits is a very old friend. Fig. 1 shows a common-emitter amplifier driving a common-collector or emitter follower stage. The current in the input stage is low enough, while a high current gain minimizes the output impedance. If overall feedback is applied as in Fig. 2, the transimpedance of the circuit is reasonably well-defined by  $R_{3}$ , i.e., for an input current i, the output voltage swing is close to  $-iR_1$ . This is assuming high enough gains that the input behaves as a virtual earth to signals. The quiescent  $V_{be}$  of  $\approx 0.6V$  prevents accurate low-level d.c. amplification, and this is a limitation shared by the monolithic i.c. based on this concept.





Voltage transfer function, Fig. 3, is non-linear with a steep slope corresponding to a fairly high voltage gain, from a few tens to a few hundreds depending on supply voltage, resistance values, etc. An input resistor can be added to produce an elementary see-saw amplifier, Fig. 4; the resulting voltage transfer function as shown in Fig. 5 is linearized by the negative feedback. Use of a voltage source short-circuits the negative feedback in the circuit of Fig. 2 which is intended as a current in/voltage out or transimpedance stage.

The circuit is the d.c. feedback pair and is already one of the most versatile problemsolvers for the hard-pressed designer. In monolithic i.c. form the ready availability of constant-current sources allows them to replace  $R_1$  and  $R_2$ , as in Fig. 6. Two benefits accrue. First, the increased dynamic resistance boosts the gain, and secondly the transistor currents can be tightly controlled against wide variations in supply voltage and temperature. This latter effect ensures that the voltage gain is stabilized as well as being increased.

A second sub-circuit which is also well known is now added. Fig. 7 shows the current-mirror which plays the dual role in these amplifiers, to that played by the longtailed pair in operational amplifiers. For well-matched transistors operated at equal  $V_{be}$ , the collector currents are comparable. The differences arise partly from inevitable imbalance in the transistors, partly because as shown the base currents contribute an error and partly because the collector-emitter voltages may differ. When a current-mirror is added to the d.c. feedback pair from Fig. 6 then a complete, albeit limited performance, currentdifferencing amplifier results (Fig. 8).

With negative feedback added the third source of error listed above is removed, because the collector potential of  $Tr_2$  is constrained to equal that of  $Tr_3$  base, i.e.,  $Tr_1$  and  $Tr_2$  have comparable  $V_{r_2}$  values as well as identical  $V_{be}$  values. With resistive feedback the output voltage is controlled by the difference between the two input currents. That at the inverting input would normally flow on through the feedback path (provided  $Tr_i$  base current is small) while that at the non-inverting input is mirrored in the current drawn by Tr, collector, reducing the net input. Both currents are normally positive. Transistors  $Tr_1$ ,  $Tr_2$  are inoperative for negative currents while that in the inverting input may have either polarity. As a main advantage of the amplifier is that it can perform complex functions while requiring only a single-polarity supply, negative currents are not normally present except for some a.c./pulse circuits. Additional networks are provided in the practical amplifiers to clamp the inputs on negative voltage swings.

A more detailed description of the practical circuit will be given in a following article. This set of Circards (no. 16) is concerned with signal amplifying and processing, which will be followed by a series on the generation of signals and waveforms. In each of these areas novel solutions can be found, and we can only hope as users to match the ingenuity of the designers of these new amplifiers. 97

Three sets of cards deal with current differencing amplifiers

of the LM3900 kind. This set covers signal processing applications, set 17 covers signal generation and a third set deals with various other circuits including test, measurement,

detection, logic and driving circuits.



### Current differencing amplifiers

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#### Typical performance Supply: 15V

R<sub>L</sub>: 5kΩ Voltage gain: 2,800 (69dB) Output swing: 0.1 to 14.2V Output current: source 10mA, sink 1.3mA (overdriving inverting input

increases sink current up to >30mA) Input current: 30nA

Unity-gain bandwidth: 2.5MHz Slew rate:  $+0.5V/\mu s$ ,  $-20V/\mu s$ 

N.B. Data is for National Semiconductor LM3900. A similar amplifier is available from Motorola and other manufacturers are expected to "second-source" such devices. Refer to manufacturers data sheets particularly for maximum ratings. While other current-differencing amplifiers may be expected to have similar performance in the circuits to be described it is important that the ratings of particular devices are not exceeded.

#### Circuit description

Transistors Tr<sub>1</sub>, Tr<sub>2</sub> are a current mirror with the collector current of Tr<sub>2</sub>, approximately equal to noninverting input current, subtracting from inverting input current at base of Tr<sub>a</sub>. The net input current to Tr<sub>3</sub> is (I-) - (I+) and this is amplified by Tr<sub>3</sub> with Tr<sub>5</sub>, Tr<sub>6</sub> forming an improved emitter follower output stage. Constant-current generators define the operating conditions while  $Tr_4$  comes into action on over-driving the input to maximize the sink-current. Output depends on the difference between two positive input currents with negative feedback taken to the inverting input when the gain is to be defined. The noninverting input is outside the feedback loop, and behaves as a forward-biased p-n junction. With resistive negative feedback applied between output and inverting input, the direct currents in the two inputs will be equalized to within the accuracy of the current mirror. If the noninverting input current is defined by a resistor to +V, the direct output voltage is then a fixed fraction of +V. Transistor Tr<sub>s</sub> base current is  $\approx$  30nA, allowing very low bias/signal currents, and like the voltage gain and output current capabilities is controlled over wide temperature and supply variations. An internal regulator (not shown) ensures this by providing the constant currents while also biasing a set of transistors that clamp each input to  $\approx -0.3V$  on negative input swings.

#### Further reading

Frederiksen, T. M., Howard, W. M., Sleeth, R. S., The LM3900-A New Current-Differencing Quad of  $\pm$ Input Amplifiers, National Semiconductor application note AN72. Frederiksen, T. M., Norton quad amplifier subtracts from costs, adds to design options, Electronics, Dec. 6, 1973, pp.116-20. Motorola Linear Integrated Circuit Data Book, pp.7-446, 7-453, 7-456 and 7-463; data sheets on MC3301P and

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# Set 16: c.d.as—signal processing—2

### **Basic amplifiers**—1



**Circuit description**—1 No direct current flows in R1 and hence R<sub>2</sub>, R<sub>3</sub> determine the d.c. operating conditions. For perfect balance between the input circuit transistors they will carry equal current and for  $V_{be} \ll + V$ , the direct output voltage is given by  $R_3/R_2 \times (+V)$ , i.e.  $R_2 = 2R_3$  is the usual condition for maximum available voltage swing with the output biased at supply mid point. As there is no significant alternating current in R<sub>2</sub>, it is the alternating currents in R2 and  $R_1$  that are equal in magnitude

Typical performance Supply: +15VR<sub>1</sub>:  $100k\Omega$ R<sub>2</sub>:  $2.2M\Omega$ R<sub>3</sub>:  $1M\Omega$ C<sub>1</sub>:  $0.1\mu$ F Direct output voltage  $\approx 7V$ Voltage gain  $\approx -10$ 

while the virtual earth at the inverting input (though  $\approx 0.6V$ d.c.) gives a voltage gain of  $-R_3/R_1$ . The addition of reactive components modify the gain, so that a highfrequency roll-off is readily achieved by placing a capacitor across  $R_3$  (corner frequency  $1/2\pi R_3 C$ ).

Capacitive coupling may be required to the load, while the reactance of  $C_1 \ll R_1$  at lowest frequency. Maximum resistance values of up to 10M may be used, but roll-off due to stray capacitances is likely.



#### Circuit description—2

The base-emitter voltage of the input transistor is  $\approx 0.55V$  at room temperature, falling by  $\approx 2.5 \text{mV}$  for every 1 deg C rise in temperature and by less than this (typically 0.5 to 1mV) for each 1-V increase in the supply voltage. This voltage is thus sufficiently stable to be used as the reference voltage for setting the d.c. output conditions, and the technique may be called the " $nV_{be}$ " biasing method. It is identical in principle to that used in the d.c. feedback pair and the "amplified-diode". If the source has an internal resistance to ground  $< R_1$  then direct coupling may be used

Typical performance Supply: +15VR<sub>1</sub>:  $470k\Omega$ R<sub>3</sub>:  $4.7M\Omega$ R<sub>2</sub>:  $470k\Omega$ Direct output voltage  $\approx 6V$ Voltage gain  $\approx -10$ 

with  $R_1$  chosen to provide the required input resistance for the circuit and R<sub>3</sub> determining both the voltage gain and the direct output voltage. Resistor  $R_2$  is omitted in this mode as is the input coupling capacitor. Direct output voltage  $\approx (R_3/R_1 + 1)V_{be}$ . Voltage gain  $\approx -R_3/R_1$ . The method requires modification both for high and low gains as the direct output voltage may not be convenient. By capacitive coupling to R<sub>1</sub> the d.c. and gain conditions can be made independent, with direct output voltage  $\approx (R_3/R_2 + 1)V_{be}$  and voltage gain  $\approx -R_s/R_1$ .



#### Circuit description-3

A third variation on the biasing methods available, is to take the bias resistor  $R_4$  to a separate reference voltage V<sub>ref</sub> which can be decoupled to make the output voltage much less dependent on supply ripple. A single reference voltage (here equal to + V/2) may be used for a number of separate amplifiers. Separate control of the quiescent output conditions is by variation of  $R_4$  for each amplifier while adjustment of R1, R2 varies all of them simultaneously. The amplifier is shown with the signal applied to the

Typical performance Supply: +15VR<sub>1</sub>:  $47k\Omega$ R<sub>2</sub>:  $47k\Omega$ R<sub>3</sub>:  $100k\Omega$ R<sub>4</sub>:  $1M\Omega$ R<sub>5</sub>:  $1M\Omega$ C<sub>1</sub>:  $0.1\mu$ F C<sub>2</sub>:  $1\mu$ F Direct output voltage  $\approx 7.5V$ Voltage gain  $\approx +9.5$ 

non-inverting input. No feedback is available at this input and so the impedance of the input transistor affects the input current. At room temperatures,  $r_1 \approx 0.026/I_{R_g}$ , giving a value >3k $\Omega$  for the values shown. This reduces the gain to about 3% below the simple theoretical relationship  $R_5/R_3$ . This biasing method is equally applicable to the inverting amplifiers.



Circuit description-4 The value of the feedback resistor is limited to a few megohms for several reasons (bias instability, effect of stray capacitance, noise and hum). If it is required to have a high input resistance and high voltage gain than the a.c. and d.c. feedback must be different. As shown, R<sub>4</sub> and R<sub>3</sub> constitute a potential divider for the output signal while only R<sub>3</sub> is involved in the d.c. feedback. Output voltage has a quiescent value of  $+ V(R_2 + R_3)/R_5$ . Voltage gain is  $\approx (-R_2/R_1)$  $(R_3/R_4 + 1)$ . Where  $R_2 = R_1$ , a convenient condition, the

### Typical performance

Supply: +15VR<sub>1</sub>:  $1M\Omega$ R<sub>2</sub>:  $1M\Omega$ R<sub>3</sub>:  $1M\Omega$ R<sub>4</sub>:  $10k\Omega$ R<sub>5</sub>:  $2.2M\Omega$ C<sub>1</sub>:  $0.1\mu$ F C<sub>2</sub>:  $4.7\mu$ F Direct output voltage  $\approx 7V$ Voltage gain  $\approx -95$ 

voltage gain simplifies to  $-(R_3/R_4 + 1)$ . However when the ratio  $R_3/R_4$  is large the feedback theory demands that the limited open-loop gain be taken into account. In practice, a ratio that should set the gain to -20 will do so to within about 1%, while a nominal gain of -100 would be nearer to -95.

**Cross references** Set 5, cards 5, 8, 9, 10. Set 7, cards 4, 10. Set 10, cards 1, 9. Set 16, cards 2, 5, 6, 10.
Typical performance

## **Basic amplifiers**—2



Circuit description—1 Common-mode signals are a problem when transmission over lines has to take place in a noisy environment. By coupling the signals through a transformer such common mode signals are minimized, while the anti-phase inputs of the current differencing amplifier offer a further improvement. Any commonmode voltage at the



transformer secondary produces equal currents at the two inputs largely cancelling each other because the gain at the two inputs is equal and opposite. R<sub>line</sub> is inserted to achieve the correct loading on the source with R<sub>1</sub>, R<sub>2</sub> sufficiently larger not to affect that loading. Quiescent output voltage  $\approx V(R_4/R_3)$ ; voltage gain  $\approx -R_4/R_1$ .



Supply: +15V  $R_1: 10k\Omega$   $R_2: 100k\Omega$ Direct output voltage  $\approx 6V$ Voltage gain  $\approx 11$ (transformer secondary to output)

Circuit description-2 Where the source is inductive or is to be transformercoupled, a variant of the "nVbe" biasing method provides a simple solution. Again there is the restriction that the direct output voltage and the gain of the amplifier itself are controlled by the same resistor ratio but decoupling part of R1 to ground can make the ratio for signal frequencies  $\gg$  the ratio at d.c. if required. In the extreme case,  $R_1$  can be completely decoupled giving the full open-loop gain of the amplifier.

This coupled with the step-up turns ratio of the transformer gives a very high overall gain. As shown, the overall voltage gain is  $\approx n(R_2/R_1+1)$  and the quiescent output voltage is  $\approx (R_2/R_1+1)V_{\rm be}$ . Because the input current required by the amplifier is very small ( $\ll 1\mu A$ ) the effective input impedance remains high regardless of the gain, and high step-up ratios are possible. This yields a very sensitive microphone amplifier though the noise performance is unlikely to allow use in audio applications.



Circuit description—3 The biasing techniques described above are easily adapted for dual supply operation, making the amplifier compatible with standard op.amp. ±15V systems. One method is to bias the non-inverting input from the centre rail, leaving the same p.d. across R<sub>2</sub> and R<sub>3</sub> if  $R_2 = R_3$ , i.e. the output is substantially at 0V if the amplifier has well-matched inputs. This remains true for all values of positive and negative supply voltages including those cases where they are markedly unequal in value. R<sub>1</sub> plays no part in the d.c. conditions and is chosen

Typical performance Supply:  $\pm 15V$ R<sub>1</sub>:  $100k\Omega$ R<sub>2</sub>, R<sub>3</sub>:  $1M\Omega$ R<sub>4</sub>:  $10k\Omega$ C<sub>1</sub>:  $0.1\mu$ F C<sub>2</sub>:  $1\mu$ F Direct output voltage  $\approx 0V$ Voltage gain  $\approx -10$ 

for the desired gain/input resistance. Voltage gain  $\approx -(R_2/R_1)$ . Non-inverting gain is obtained as before simply by transferring  $R_1$  to the non-inverting input. The gain is less well-defined but the bandwidth is increased, because there is then no resistor to attenuate the negative feedback, i.e. the bandwidth can approach that of the amplifier with 100% feedback while the gain can approach the value defined by  $(R_2/R_1)$ .



Circuit description—4 The present form of currentdifferencing amplifier is not suited to low-level d.c. amplification and cannot replace standard operational amplifiers directly. The adaptation shown has a reasonable performance for d.c. signals of >100mV provided the offset adjustment is made and subsequent changes in temperature and supply are restricted. Effective input of the amplifier is point A and for  $R_1 = R_2$  its potential is a 0V in the absence of feedback (the

Typical performance

Supply:  $\pm 15V$ R<sub>1</sub>, R<sub>2</sub>:  $1M\Omega$ R<sub>3</sub>:  $820k\Omega$ R<sub>4</sub>:  $500k\Omega$ R<sub>5</sub>:  $220k\Omega$ R<sub>6</sub>:  $2.2M\Omega$ Direct output voltage  $\approx 0V$ Voltage gain  $\approx -10$ 

> diode compensates for the input Vbe). Adjusting R4 sets the output voltage to 0V prior to the application of feedback with  $(R_3 + R_4) \approx (R_1 + R_2)/2$ . The resistors  $R_5$ ,  $R_6$  then set the input resistance and gain as for previous amplifiers. Point A is no longer such a good equivalent to a virtual earth from the signal standpoint since the presence of R<sub>2</sub> lowers the amplifier gain, while  $R_2//R_1$  is the equivalent resistance to ground at this point.

**Cross references** Set 5, cards 5, 8, 9, 10. Set 7, cards 4, 10. Set 10, cards 1, 9. Set 16, cards 1, 5, 6, 10.

## Set 16: c.d.as—signal processing—4

## Logic gates



## Supply: +20VIC: $\frac{1}{4}$ LM3900 R<sub>1</sub>: $150k\Omega$ R: $82k\Omega$ Output logic 0: 150mVOutput logic 1: 19.2VFor inputs commoned, output changes state for input voltage $\approx 20\%$ of +V.

Typical performance

### **Circuit description**

Availability of two current inputs simplifies the design of basic logic gates with these amplifiers. For example, a low current at one input can hold the output in one desired state while the other input receives the sum of the currents from two or more inputs. This sum can be set to overcome the bias when only one input is high or only if all are simultaneously high, leading to OR and AND-type circuits respectively. (The amplifier is working as a high-gain comparator and can also provide a majority-gate in which any two out of three inputs are enough to provide the required output. By extension some of the simpler forms of threshold logic are possible by scaling the values of resistors to assign a different weight to their importance in decision making.) In the first configuration, if any input is high the current driven into the non-inverting input exceeds the inverting input current and the output is driven high, i.e. an OR gate. The remaining input resistors connected to logic 0 bypass a small portion of that current (<0.5V/R for each resistor) but unless the number of inputs is large and/or the supply voltage is low, this is not a problem. Speed of response is limited to  $\approx 0.5 V/\mu s$  for positive swings and up to  $20V/\mu s$  for negative swings though the fall in voltage is

slower as logic 0 is approached. By interchanging the inverting and non-inverting inputs with no change in component values, a NOR gate is produced. This flexibility of being able to produce different logic functions from the same package is very attractive. In addition, one or more of the amplifiers can be used to provide astable, Schmitt trigger functions, etc. for obtaining the appropriate waveforms with which to drive the gates.

### **Component changes**

+V Normal voltage range is +4 to +36V, but some devices will operate to <3V without difficulty.

R<sub>1</sub>, R Ratio of these resistances is chosen to ensure that with the lowest expected value of logic 1 to any one input that, the resulting current flow into the non-inverting input is sufficient to overcome the fixed current in the positive input. (Allow for 0.5V/R lost in each logic 0 input.) Typically  $R = R_1/2$  is a suitable starting point and R may vary from 10k to  $10M\Omega$ . Fan-out To maximize this

R, R<sub>1</sub> should be large but there is no great advantage to raising them above  $1M\Omega$ .

### **Circuit modifications**

• The AND/NAND functions are slightly more difficult to construct for multiple inputs. This is because for an *n*-input gate the sum of the currents produced by (*n*-1) inputs at logic 1 differs little from that produced by *n* inputs if *n* is large. Up to three inputs can be used with a reasonable margin of safety as in Fig. M1, where the resistor R<sub>2</sub> reduces the current flow when two inputs are on. For a two-input AND gate R<sub>2</sub> may be omitted and the values of the input resistors adjusted. For example, with  $R_1 82k\Omega$ , R 100 or  $120k\Omega$ . • Again interchanging the inverting and non-inverting inputs gives a NAND gate. By scaling the resistors in some of these and/or feeding signals to both inputs, circuits which are intermediate between true analogue and true digital circuits become possible.

(Fig. M2)
If it is required to have multiple inputs to an AND gate, simple diode gating may be used, Fig. M3, with the amplifier serving a similar function to the transistors in d.t.l., but with the advantage that the ratio of output to input current, which determines the fan-in fan-out capability, is of



the order 10<sup>6</sup> (25nA-25mA). If any input is at logic 0 the current in  $R_1$  flows through that diode. All inputs have to be at logic 1 for the current to be diverted into the amplifier non-inverting input to give a logic 1 out. The threshold voltage and hence the noise margin is low in this circuit. If all inputs but one are at logic 1 then that input need only rise to  $\approx 0.5V$  to be recognized as logic 1. A NAND function is obtained by reversing the amplifier inputs.

Cross references Set 11, cards 1, 2, 11.



## High voltage amplifiers



### **Circuit description**

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Where a high output voltage swing is required the amplifier must be fed from a separate low-voltage supply, and a suitable high-voltage transistor employed to withstand the main supply voltage. The configuration depends on whether it is the output voltage or current that is to be defined. If the former, then the feedback is taken from in shunt with the load. For an inverting-gain amplifier and the transistor in the common-emitter mode, the inverting gain of the transistor necessitates that the feedback be applied to what is normally considered as the non-inverting input. For an input of 0V d.c., the current flow in R1 is small and that in  $R_2$  is forced by the feedback to equal that in  $R_4$ . If in this condition the output is desired to be +HT then  $+HT/R_2 = +V/R_4$  gives the required value of R4. Overall voltage gain is given by  $(-R_2/R_1)$  as the circuit is effectively a "see-saw" amplifier while the input resistance is approximately R<sub>1</sub>. Resistance R<sub>5</sub> introduces a small amount of negative feedback into the output stage and raises the amplifier quiescent voltage well into its linear region. Capacitor  $C_1$  modifies the gain/frequency characteristic to maintain stability at the higher loop gain. Output voltage swing can be up to 95% of the supply voltage if lightly loaded and the negative feedback keeps the output impedance reasonably low.

## **Component changes**

R<sub>1</sub>, R<sub>2</sub> These set the input resistance and the voltage gain

 $R_2 \gg R_3$  to minimize loading on the output.  $R_2$  1M to 22M $\Omega$ . R<sub>3</sub> Load resistance, dictated by user requirements. For use as an r.c.-coupled amplifier,  $\mathbf{R}_{a}$  has to carry a quiescent current greater than peak load current required. R<sub>4</sub> Sets d.c. conditions at output. If  $V_0 = +HT$  required for  $V_{in} = 0$ , then  $+ V/R_4 =$  $+HT/R_2$ . If output quiescent to be +HT/2 as when used for amplifying a.c. input signal, then  $+ V/R_4 = + HT/2R_2$ . R<sub>5</sub> Not critical. Raises amplifier output quiescent voltage to 1 to 3V range, i.e. into linear region. Value dependent on output quiescent current but might be  $50\Omega$  to  $5k\Omega$ . +VChosen to suit amplifier,

and available supplies (+4 to +36V). +HT Dictated by load

requirements. Tr<sub>1</sub> Must have voltage rating

in excess of + HT particularly if inductive loading possible.

## **Circuit modifications**

 A non-inverting amplifier again using a common emitter output stage has the feedback applied to the amplifier non-inverting input, while the signal is applied to the inverting input (the inversion in the transistor ensuring the correct overall sign for the gain). For a transfer function that passes through the origin a pre-biasing network is required. Again the gain is set by the ratio of the feedback to the input resistor. For a high output-impedance, the feedback is taken from the transistor emitter. The method

can be employed where the

Typical performance Supply: +15V, +300V (HT) R<sub>1</sub>:  $330k\Omega$ R<sub>3</sub>:  $10M\Omega$ R<sub>5</sub>:  $1k\Omega$ C<sub>1</sub>: 100pFVoltage gain: -31.2 (d.c. to 1kHz) Cut-off frequency: 3kHzOutput impedance:  $<10k\Omega$ (d.c. to 1kHz) Input impedance:  $330k\Omega$ (d.c. to 1kHz)

load is to be transformer coupled and the direct voltage drop across the primary is too small to allow of the d.c. feedback. The two 10-M $\Omega$ resistors define the potential at the transistor emitter as about  $2V_{be}$  and allows the output current to be fixed reasonably accurately, falling with rise in temperature. By decoupling the emitter and taking overall a.c. feedback from the load, the output impedance can still be made low if required. • Increased output resistance is possible by using compound output stages of various kinds that can also increase the

that can also increase the current capability, subject to device power limitations. An f.e.t. draws no current from



the amplifier ensuring that the load and emitter resistor currents change together if the current  $I_{\text{bias}}$  is made constant, either because of the high supply voltage and correspondingly high resistance R, or by a separate low-voltage constant-current stage.

**Cross references** Set 7, cards 5, 7. Set 16, cards 1, 2, 6.



## Set 16: c.d.as—signal processing—6

+V = +12VR<sub>L</sub> = 50Ω

## **Power amplifiers**



#### **Circuit description**

Addition of a Darlingtonconnected pair of transistors increases the output current capability from 10mA to 1-5A depending on the ratings of the transistors used. One restriction is that the  $V_{be}$ 's of the transistors limit the output voltage to around 2.5V below the supply level allowing for the amplifier internal saturation. The additional phase shifts that may occur even in an emitter follower make external compensation desirable. For supply, input and output to be all positive, the configuration shown is adequate, where with  $R_2 = R_3$ , V<sub>o</sub> varies linearly with  $V_i$  provided  $V_i$  is above the amplifier internal V<sub>be</sub>. The relationship is then  $V_0 \approx V_{be}$  $+(R_4/R_1)(V_1 - V_{be})$  as the currents in R1 and R4 have to be equal. This means that as a d.c. amplifier it is of relatively low accuracy but is quite suitable for supplying small d.c. motors under the control of a phase-locked loop. A combination of the techniques for increasing the current ratings and voltage could allow the production of high-power amplifiers. Replacing the emitter follower stages by common emitter amplifier increases the available positive output swing to within a hundred millivolts of the positive supply.

#### **Component changes**

 $R_1$  This sets the voltage gain in conjunction with R<sub>4</sub>. Because of the Vbe offset, the output voltage becomes temperature dependent particularly for V<sub>1</sub> comparable to V<sub>be</sub>, i.e. high gains are not compatible with good stability in this configuration.  $R_1$  100k to  $10M\Omega$ .

 $R_2$ ,  $R_3$  These provide forward bias for each of the inputs allowing the output to be controlled for inputs down to zero. This is best achieved for  $R_1 = R_4$  when  $V_0 = V_1$  is the first-order approximation, the Vbe effects at the two inputs cancelling.

 $R_4$  As suggested,  $R_4$  may equal R<sub>1</sub> for optimum stability but with gain restricted to +1.  $R_4 \ge 2.0R_1$ except for a.c. amplifiers where some drift in quiescent conditions is acceptable.  $R_5$  1k $\Omega$ . Together with  $C_1$  is part of the suggested network for maintaining stability. Limits amplifier current under load short circuit conditions providing protection for amplifier and output stage. Not adequate unless proper heat-sinking used since limit of output current depends on transistors' current-gains and is ill defined.

C1, C2 Control high frequency performance. C1: 330p to 2.2nF, C<sub>2</sub>: 5.6 to 22pF. Choose lowest values giving stability under operating conditions.

R<sub>6</sub> Load resistance. This may range down to  $10\Omega$  as currents of several amps are possible.

## **Circuit modifications**

• A simpler circuit based on the same principles is Fig. M1. The limiting and compensation components have been eliminated together with the high-power transistor. The bias network shown is suitable for a source with resistive path to ground  $\ll 470 k\Omega$ , with the non-inverting input grounding. The bias method is then basically the "nVbe" method as in the "amplified diode". Alternatively a capacitively coupled source may be used with no direct current in R<sub>2</sub> and  $R_1 \approx 2R_3$  to set  $V_0 \approx + V/_3$ . • For higher efficiency the usual Class B technique can be applied with a complementarysymmetry output (Fig. M2). Not recommended for low-distortion applications, since additional bias networks to overcome crossover problems would lose the advantage of simplicity normally offered by this amplifier.

• Although the supply voltage capability is good (up to 36V)

the output swing can be extended to  $\approx 70V$  (Fig. M3) using a bridge configuration. The simplest arrangement for a.c. signals has the  $1M\Omega$  and  $470k\Omega$  resistors setting the output quiescent voltages to + V/2 for maximum undistorted voltage swing. Replacing the two input resistors by a potentiometer and applying the signal via the pot to opposite-phase inputs gives anti-phase outputs that can be set for equal magnitude with an overall voltage gain of about 18.

#### Cross references

Set 7, cards 2, 4, 7, 8, 10, 11. Set 16, card 5.



## Set 16: c.d.as—signal processing—7

## **Bandpass filters**



Typical performance R:  $47k\Omega$ C: 10nFR<sub>1</sub>, R<sub>2</sub>:  $1M\Omega$ R<sub>3</sub>:  $6.8M\Omega$ R<sub>4</sub>:  $100k\Omega$ Supply: +15Vf<sub>0</sub>: 320HzFor Q = 15 k = 0.640N.B. Onset of oscillation at k = 0.648 compared with theoretical value of 0.66.

## **Circuit description**

Active filter techniques based on operational amplifiers are applicable to currentdifferencing circuits. That shown is a direct adaptation of Circard 1, Set 1. A Wien network is placed between source and output with the potential at its junction monitored by the amplifier input terminal. Resistor R<sub>1</sub> is a high-value resistance to minimize loading on the network. Resistor R<sub>2</sub> provides a variable amount of positive feedback to increase the Q of the circuit with minimal effect on the centre-frequency while R<sub>a</sub> sets the quiescent output voltage to allow for maximum signal swing. Because  $R_2 > R_1$ would be required for stability, with R<sub>2</sub> taken directly to the output, and variable high-value resistors are inconvenient, the alternative is to tap R<sub>2</sub> onto a variable portion of the output voltage. The result is a band-pass filter with centre frequency given by  $f = 1/2\pi CR$ and with Q controlled by R<sub>4</sub>. The limited gain and bandwidth capabilities of the amplifier does not allow the circuit to provide large stable Os nor to operate successfully at frequencies  $\gg 10$ kHz. The input impedance falls as the Q is increased because of the increased output swing (gain  $\propto$  Q to a first order). It is also true that the sensitivity of such circuits to variations in component values is proportional to the Q, i.e. for Q = 10 a 1% change in a critical resistor might produce >10% variation in the Q itself.

## **Component changes**

R, C The load on the output should not fall much below  $10k\Omega$  at any frequency to avoid loss of loop gain, distortion, etc. R 10k to 1M $\Omega$ , C 220p to  $10\mu$ F

 $R_1$  Used to set a convenient level of current at amplifier inputs; too low and loading on Wien network disturbs Q,  $f_0$ ; too high and stray capacitive coupling disturbs response. 220k to 10M $\Omega$ .

 $R_2$  Convenient to set  $R_2 = R_1$ using  $R_4$  to control feedback. Alternatively for fixed Q particularly if low, eliminate  $R_4$ taking  $R_2$  directly to output. Then  $R_2 > 1.5R_1$ .  $R_4$  Not critical, but ≥ 10kΩ

and  $\langle R_2$ .  $R_3$  Sets quiescent output voltage for maximum swing. Should be chosen after other components have been selected for desired response. Typically 2 to  $10 \times R_1$ , but for a given value, small variations in k and hence Q can be accommodated without serious bias changes.

#### **Circuit modifications**

• Multiple feedback circuits (Fig. M1) are the equivalent of the virtual earth circuits used with conventional operational amplifiers. With correct scaling of resistors, capacitors Q > 1is achieved simultaneously with centre-frequency gain of unity. Again both are strongly dependent on component stability at high-Q values. • A better approach to filter design with currentdifferencing amplifiers is to select passive networks whose transfer-function gives a



defined output current. It is possible to adapt the Wien and similar networks, Fig. M2. Normally R<sub>2</sub> would be grounded and the p.d. across it fed back to a high impedance point on the amplifier. By feeding back the current in R<sub>2</sub> into what is almost equivalent to an a.c. virtual earth point, the overall transfer function can be varied more easily. Components R<sub>1,2</sub> C<sub>1,2</sub> determine  $f_0 = 1/2\pi (R_1R_2C_1C_2)^{\frac{1}{2}}$ , while R<sub>5</sub> sets the Q and R<sub>4</sub> the quiescent output value.

• Any of the other RC networks used in bandpass filter designs are applicable and Fig. M3 gives almost identical performance for  $R_1 = R_2 = R$ ,  $C_1 = C_2 = C$ .

Cross references Set 1, cards 1, 3, 6, 7, 8, 12. Set 5, cards 6, 10. Set 10, card 8. Set 16, cards 8, 9.



## Set 16: c.d.as—signal processing—8

## Notch filters



Typical performance Supply: +20VR<sub>1</sub>:  $100k\Omega$ R<sub>2</sub>, R<sub>3</sub>: 1MR<sub>4</sub>:  $1k\Omega$ R:  $47k\Omega$ C: 10nFNotch frequency: 319HzAchieved for  $m \approx 0.35$ for k = 0 and k = 1



#### Circuit description

The circuit is again derived from a Wien Bridge to demonstrate the principles by which known circuits can be adapted to current-differencing amplifiers. It provides a notch or null in the response at a frequency set by the RC values  $(f = 1/2\pi RC)$  with  $R_1$ providing a trimming action to get as true a null as possible. If  $R_4$  is a low resistance potentiometer, then the depth of the null is unaffected as the tapping point is varied, since both ends of the potentiometer are at zero for this condition. However, the positive feedback introduced has the effect off-null of sharpening up the response so that the gain remains close to unity for frequencies close to the null. This makes the circuit useful for nulling out the fundamental (or particular harmonic) of a complex waveform with minimal effect on the other harmonics. A weakness of this particular circuit is that it relies on the matching of the inverting and non inverting input sensitivities making it prone to variation with supply/ temperature changes, etc.

### **Component changes**

R, C The difficulty with this particular circuit is that these should have a low impedance compared with R<sub>3</sub> so that the latter does not load them, while not in turn being disturbed by the varying source resistance of R<sub>4</sub> as k is varied. Typically R 10 to  $100k\Omega$ with C to give frequency as  $f = 1/2\pi RC C$ , 220p to  $10\mu$ F.  $R_2$ ,  $R_3$  1 to 10M $\Omega$ . At higher frequencies stray capacitance effects bring phase shifts preventing good null.  $R_1$  10 to 250k $\Omega$ .  $R_4$  1 to 10k $\Omega$ . This circuit experiences shift in

the null because of supply temperature changes and requires careful adjustment before use if sharp notch required.

## Circuit modifications

• T-networks normally have the disadvantage that to control the frequency while retaining a sharp notch requires adjustment of a number of interacting components. Interconnected with an amplifier (top circuit), using positive feedback to steepen the sides of the notch (equivalent to varying the Q but not the centre frequency), it has the advantage that variations in the amplifier gain only affect the steepness leaving the depth unaffected. A further advantage accrues using the currentdifferencing amplifier in that the gain can be adjusted, while retaining a fixed input impedance, by varying R<sub>f</sub>. This also affects the gain at frequencies well away from the notch, but the approach is often used where the gain is just less than unity the sides of the notch very steep and unity gain can be assumed off resonance. Independent adjustment of C and R are required to obtain a complete null and the system is easier to use if fine frequency control is possible at the oscillator, as may be possible when used as part of a distortion measuring system.

• Other passive networks have been described which give a good null at a specified frequency and though normally used as shown with a separate feedback attenuator and buffer, the network can also be used as above. By deriving the feedback directly from the output of the amplifier variable resistance of the attenuator is avoided and the need for the buffer eliminated.

## Further reading

Rowe, N. B. Designing a low frequency active notch filter, *Electronic Engineering*, April 1972.

Dance, J. L. & Edwards, K. H. Simple null filter with variable notch frequency, *Electronic Engineering*. vol. 36 1964, pp.478/9.

#### Cross references

Set 1, cards 9, 10. Set 16, cards 7, 9.



## Set 16: c.d.as—signal processing—§

## Low pass/high pass filters



#### **Circuit description**

The most convenient configuration for most feedback circuits with these amplifiers is the virtual earth configuration; it is not possible to use series applied feedback as with standard Sallen & Key type filters. The network is not easy to analyse component-bycomponent because of the interactions between them but the overall transfer function is well defined particularly if the gain in the pass-band is low, e.g. unity. Then it is sufficient if the amplifier voltage gain is >100 to have a transfer function that is very close to the theoretical value. A second-order low-pass filter results where the cut-off frequency may range from 1Hz to >100kHz. A convenient means of adjustment of the filter properties is via  $C_1$ ,  $C_2$ . It is their product that fixes the cut-off frequency (for R<sub>1</sub>, R<sub>2</sub>, R<sub>3</sub> fixed) while their ratio determines the shape of the transfer function (i.e. Butterworth, etc.). The output impedance is low and the input impedance may be up to  $1M\Omega$  if required allowing such filters to be cascaded with negligible loading. As shown the output voltage is defined as  $(R_s/R_1 + 1)V_{be}$ provided the input has a quiescent value of zero, as would be achieved by capacitive coupling from the previous stage. Since  $R_1 = R_3$ is a convenient value for design of the filter characteristics this restricts the output to a quiescent value of about 1.2V regardless of the supply. This may be inconvenient where larger voltage swings are desired and

## Typical performance Supply: 5VR<sub>1</sub>, R<sub>3</sub>: $150k\Omega$ R<sub>2</sub>: $47k\Omega$ C<sub>1</sub>: 470pFC<sub>2</sub>: 120pF-3dB point $\approx 8kHz$ (low frequency gain about -1) Max. input/output swing $\approx 1V$ pk-pk.

alternative biasing schemes may be needed (see Circuit modifications).

#### Component changes

Transfer function (voltage gain) is

$$\frac{\frac{1}{R_1R_2C_2C_1}}{s^2 + \left(\frac{1}{R_2} + \frac{2}{R_1}\right)\frac{1}{C_1}s + \frac{1}{R_1R_2C_1C_2}}$$

where  $s = j\omega$ provided  $R_1 = R_3$ . This is the basic form of a second-order low-pass filter giving  $\omega_n = 1/\sqrt{R_1R_2C_1C_2}$  and equivalent  $Q = 1/2\xi$  where  $\xi$  is the damping factor given by

$$Q = \frac{\sqrt{R_1 R_2}}{R_1 + R_2} \cdot \sqrt{\frac{C_1}{C_2}}$$

Setting R<sub>1</sub>, R<sub>2</sub> to determine input impedance, C<sub>1</sub> and C<sub>2</sub> may be varied to control cut-off frequency and shape of response. Note that the equations proposed in National Semiconductor applications note AN72 pp.14-15 show a circuit as having Q = 1, for which the above equations suggest  $Q \approx 0.4$ . Measurements support the equations quoted above. R<sub>1</sub>, R<sub>2</sub> 10k to 1M $\Omega$ . C<sub>1</sub>, C<sub>2</sub> 47p to 10 $\mu$ F.

### **Circuit modifications**

• If the filter is part of the system where it is convenient to bias input and output to + V/2 then the addition of resistor R<sub>4</sub> to the positive supply rail as shown permits this for  $R_2+(R_1+R_3/2)=R_4$ . The resistors R<sub>1</sub>, R<sub>2</sub>, R<sub>3</sub> are first selected to set the transfer function and the nearest



preferred value for  $R_4$  chosen using the given equation (Fig. M1.)

• The corresponding high-pass filter has the locations of the capacitors and resistors interchanged. Resistor R<sub>2</sub> is the only one contributing bias current to the inverting input leaving  $R_4 = 2R_2$  as the condition for a quiescent output voltage of + V/2. (Fig. M2.) • A current differencing amplifier can be adapted to behave as a voltage amplifier of defined gain-unity gain in the example, Fig. M3. R can be high in value and the loading effect on a separate passive network is light enough that filters similar to the Sallen & Key filters using voltage followers are possible.

### **Cross references**

Set 1, cards 4, 5, 6, 7, 11 Set 5, cards 2, 3, 7, 10. Set 16, cards 7, 8.



## Set 16: c.d.as—signal processing—10

## Gain-controlled amplifiers



#### Typical performance +15VSupply **1LM3900** A<sub>1</sub> R1 $2.2k\Omega$ R۶ 100Ω R<sub>3</sub>,R<sub>4</sub> 100kΩ R<sub>5</sub> 33kΩ C<sub>1</sub> $10\mu F$ 400mV pk-pk 10kHz Vin for V<sub>B</sub> 0.9V 5V pk-pk ٧o



The non-inverting input has a diode-connected transistor as part of a current-mirror. For a given feedback resistor to the inverting input, the current flowing in the non-inverting input gives a proportional output voltage. The nonlinearity of the input impedance leads to output waveform distortion if any low-resistance parallel path is used to attenuate the signal. If the path consists of a suitablybiased silicon diode then its non-linearity is comparable with that of the input stage and the input signal division between the two paths has little variation with signal amplitude, i.e. little distortion results. This remains true while the signal current is well below the bias level.

As the direct current in the diode is increased, the corresponding fall in sloperesistance by-passes more signal current reducing the gain. Resistor R4 provides sufficient bias current to set the output potential to a minimum of  $V_s/3$ . This applies when the bias voltage is low with the direct current in  $R_3$  as well as the a.c. signal current via R<sub>1</sub> are both shunted away through  $R_2$ ,  $D_2$ . The gain is virtually zero in this condition. As VB approaches one diode voltage, the direct currents in  $D_1$ ,  $D_2$ and the amplifier input become comparable. The slope resistances are also comparable and the input current divides equally between the amplifier and  $D_2$ . For any further increase in  $V_B$  the signal current flows on through D, to the amplifier with negligible attenuation by

D<sub>2</sub>. At this extreme the direct current in R<sub>3</sub> contributes to the amplifier bias and the output d.c. level rises to  $2V_s/3$ . The bias range from 0 to 1V may be raised to any higher level by adding a potential divider.

#### **Component changes**

R<sub>1</sub> Sets input impedance 1k to  $1M\Omega$ . R<sub>2</sub> Not critical but  $\ll R_1$ . R<sub>3</sub>, R<sub>4</sub> Normally equal, setting limits to output quiescent voltage. R<sub>3</sub>== R<sub>4</sub>= 3R<sub>5</sub> for output levels ranging from V<sub>s</sub>/3 to  $2V_s/3$ .

 $R_5$  Ratio  $R_5/R_1$  sets maximum value of voltage gain.

 $C_1$  Not critical, but reactance  $\ll R_1$  at lowest signal frequency.  $D_1$ ,  $D_2$  Small signal silicon diodes.

V<sub>s</sub> Whole voltage range of amplifier, 4 to 36V.

 $V_{\rm B}$  Centre of gain-control region occurs for 0.6V but this is temperature dependent.

#### **Circuit variations**

• In some applications it may be more convenient to make the control voltage compatible with the normal output of other amplifiers. The potential divider raises the range of bias voltages to 0 to 10V. The component values, Fig. M1, give comparable gain figures at much higher impedance levels (taken from manufacturers application notes).

• Automatic control of output voltage against input signal variations (a.g.c.) is possible by detecting the output amplitude and feeding a smoothed control signal back to the amplifier bias input. For best control the rectified output can be compared against a direct reference voltage with the amplified difference being fed back, Fig. M2.

• A simpler circuit, Fig. M3, uses a.c. coupling between the diode being controlled by R<sub>3</sub> and the amplifier non-inverting input. An alternative d.c. biasing technique sets the output voltage to  $(n + 1)V_{be}$  provided the value of R<sub>1</sub> is sufficiently low that the current in it  $(\approx 0.6 V/R_1)$  is  $\gg 30 nA$ , the typical input current of the amplifier. Increasing the bias voltage lowers the slope resistance of D<sub>1</sub>, diverting signal current from the amplifier input and reducing the gain (maximum gain  $\approx nR_1/R_2$ ). The gain cannot be reduced to zero with this circuit.

Cross references Set 16, cards 2, 3.





## Set 16: c.d.as-signal processing Up-date

1. Of the standard op-amp active filters, the two-integrator loop is one of the best known-though under a variety of names including "bi-quad" and "state-space". The overall loop gain at d.c. has to be inverting to prevent d.c. latchup. This is usually achieved by inserting a unity-gain inverting stage in the loop. An alternative is to apply the feedback to the non-inverting input of one of the amplifiers while leaving the integrating capacitor at the inverting input. Using op-lamps, a high Q is only attainable using a wide



spread of component values (see ref. 2). With a c.d.a. or Norton amplifier such as the LM3900 or MC3401, the natural Q of the system shown is in excess of 200. Lower, controlled values can be obtained by adding the damping resistor  $R_1$ , when the Q will be given by  $R_1/R$ to a reasonable approximation for Q in the range of 2 to 20. The Q can be increased by removing  $R_1$ and adding  $R_2$ , even with high values of  $R_2$  sinusoidal oscillations may occur. Both bandpass and low-pass outputs are obtained from the filter (quadrature outputs when oscillating).

## References

 Croskey, C. State variable filter uses only two op-amps, *Electronics*, 1974, Mar. 21, pp. 98/9.
 Attikiouzel, J. Low-cost active low-pass and band-pass network, *Circuit Theory and*

Applications, 1974, vol. 1.

2. The technique shown provides (i) a zener diode with an operating current stabilized by the zener voltage (ii) an output voltage controlled by the zener but having a temperature drift corresponding to a single  $V_{be}$  if the zener is a low-drift type. The p.d. across the resistor R must be Vz to a good accuracy if the  $V_{bes}$  of  $Tr_1$ , Tr<sub>3</sub> are well matched. The lower the current in R the more accurately this holds but in any case the current variation is very small (corresponding to a few tens



of millivolts in perhaps 5 to 10V). Thus  $V_z$  is well stabilized and the output voltage is  $V_z + V_{bes}$ . If this

voltage is to be used to define the currents at other points in the system, for example the inputs of other comparators, then it has a drift which is small and of the right magnitude and sense to stabilize those currents against variations in the V<sub>bes</sub> of the inputs. For the output voltage to be temperature-independent, the zener would need a temperature coefficient of voltage of around +2.5mV/deg C.

### Reference

Fergus, R. W. Use current-mode op-amps in reference circuits, *EDN*, 1974, June 20, pp. 80/1.

3. This modification to the v.c.o. described in set 17, number 3 was proposed a reader\*, and converts the circuit into a sawtooth generator, accompanied by a pulse train with narrow pulse widths. The input potentiometer varies the frequency with the values of  $R_1$ ,  $R_2$  and  $R_3$  controlling the waveform. If  $R_1 + R_2$  are replaced by a  $470k\Omega$ potentiometer with  $R_1$  set to 340k $\Omega$  and  $\mathbf{R}_3$  to  $7k\Omega$  then the pulse width is lms and



the frequency range is 0 to 140Hz. Pulse width can be increased to 4ms with a frequency range of 6 to 86Hz with  $R_1 22k\Omega$ ,  $R_3 4k\Omega$ . In each case the supply voltage was about 16.5V, but in common with most versions of this circuit the voltage is not critical.

#### Reference

\*Saunders, L. R. Voltagecontrolled pulse generator, Private communication, Auckland, N.Z., Sept. 6, 1975.

## Set 17: c.d.as - signal generation

The following article discusses differences between c.d.as and op-amps as far as cut-off frequency and slew rate are concerned. It also gives a circuit that simulates a c.d.a. using a five-transistor i.c., such as the CA3046, so that the effects of different operating currents and compensation can be investigated. That it is the current into c.d.as that is of prime concern is reinforced in the article and best results with c.d.as obtain from networks requiring a low impedance load—the inverting input—with shunt feedback. An example is given on card 2 which shows an RC oscillator followed by a level sensing circuit to give a distortion of  $0.1\frac{6}{20}$ .

The circuits on the cards show that some functions are achieved much more simply with c.d.as than previously: rectangular and triangular-wave generators are shown on cards 1, 3 & 10. The Schmitt trigger circuit (card 6) is also very simple, provided precise switching levels are not required. Constant-voltage circuits are easily produced by using a Zener diode or planar transistor in a negative feedback loop (card 4), while constantcurrent circuits require balancing of positive and negative feedback to give very high output impedance (card 5).

Generators 1 RC oscillators 2 Voltage-controlled oscillators 3 Voltage regulators 4 Constant-current circuits 5 Schmitts and comparators 6 Astable multivibrators 7 Monostable multivibrators 8 Flip-flops 9 Staircase generators 10 109

# **Current-differencing amplifiers**

signal generation

The simple model of the current-differencing amplifier discussed in the previous article (August issue) is sufficient to explain the principles-but not enough to satisfy the customer placing his pennies on the counter. A fuller circuit is shown in Fig. 1 representing the relevant sections of one of these amplifiers, in this case the LM3900, though other manufacturers produce similar circuits. Transistors Trg. 10 constitute the input current mirror coupling a current into the external feedback network that is the difference between the two input currents. Transistor  $Tr_8$  is the only stage contributing voltage gain and its collector is the highest impedance point in the system-the most convenient point to place the compensation capacitor Csince a small capacitance is sufficient to bring the cut-off frequency down to the required level. The single stage of voltage gain is buffered by  $Tr_{4,2}$  to give a reasonably low output impedance with a current source capability of tens of milliamps.

The open-loop voltage gain is very much less than is available from standard opamps, but at 60 to 70dB (1,000 to 3,000) is ample for most applications. The reduced gain allows the open-loop cut-off frequency to be increased to about 1kHz (c.f. the value of around 10Hz for 741 op-amp) without instability occurring at high frequencies when 100% negative feedback is applied (Fig. 2). As a result the open-loop gain is 10dB greater for these current-differencing amplifiers from 1kHz to 1MHz.

This is a fair statement for small-signal applications, but the slewing characteristics of the amplifiers are quite different. In the 741 and similar amplifiers the maximum current available for the capacitor is comparable for both positive and negative swings, bringing a slew-rate of about  $0.5V/\mu s$  in both directions. In the current differencing amplifier described here, the capacitor C (Fig. 1) can be discharged rapidly by  $Tr_s$  if the latter is over-driven, and the negative slew-rate is about 20V/ $\mu s$ . The charging path for the

capacitor is via  $Tr_4$  base and the slew rate is limited by the low base current to about  $0.5V/\mu s$ , giving asymmetry to the rise and fall times of a pulsed output (Fig. 3). The resulting large-signal response when used as an amplifier is limited to around 10kHz by this positive slew-rate.

This is but the first generation of currentdifferencing amplifiers, designed for simplicity and economy. It is to be expected that circuits will gradually appear offering improvements in this and other directions. With the example of operational amplifiers as a guide, we can hope to see multi-megahertz current-differencing amplifiers before long. This could be achieved by removing or reducing the compensation capacitance, provided the circuit was not then used with heavy feedback.

It is possible to experiment with a similar circuit to see the general effects of operating at different currents and with different degrees of compensation. The











Fig. 3. Positive slew-rate is limited by the low base current in  $Tr_4$  (Fig. 1) to about  $0.5V/\mu s$ , giving asymmetry to the rise and fall time of a pulsed output.

circuit is shown in Fig. 4 and is based on one of the low-cost five-transistor packages such as CA3086, CA3046 etc. These have gain-bandwidths in excess of 500MHz demanding care in construction if good results are to be obtained. Transistors  $Tr_1$ ,  $Tr_2$  comprise the current mirror,  $Tr_3$  is the voltage amplifier and  $Tr_4$  the emitter follower. Transistor  $Tr_5$ acts as a constant-current load to the emitter follower though the slope resistance is less than that achieved by current mirrors. Bootstrapping the collector load of  $Tr_3$  increases the voltage gain giving some of the effects provided by the constant-current stage in the commercial amplifier. This circuit is in no sense a competitor for the complete i.c. but may help in understanding the techniques and limitations. (Possible values are  $R_1$ ,  $R_2$  $47k\Omega$ ,  $R_3$   $470k\Omega$ , C  $10\mu$ F, with a supply of + 10V.)

The control of direct voltages and currents is readily achieved with amplifiers of this class, with the simplest circuits requiring only the addition of a zener diode. Care has to be exercised if high stability is required since, as shown in Fig. 5, the output voltage depends on the direct voltage between the inverting input and ground. This is approximately 0.55V, changing with temperature by about -2.2mV/degC. As drawn, the zener current would be restricted to the amplifier input current of 30nA and an additional resistor between inverting input and ground would be needed to bring the current up to the level appropriate to the zener.

Sine-wave generation is by passive resonant or phase-shift networks, with the one change; that it is the current into the amplifier that is of concern. While conventional passive networks such as the phase-shift network of Fig. 6 can be adapted by using a suitably large resistance R' to force a current into the amplifier without loading the network, better results follow from designing alternative networks requiring a low-impedance



Fig. 4. By making a c.d.a. from a fivetransistor i.c. the effect of altering the compensation capacitor can be investigated, gain-bandwidth products of 500MHz or more being possible.



Fig. 5. Stability of voltage level in c.d.as can be improved by simple addition of a zener diode.







Fig. 7. Waveforms can be generated by subjecting a capacitor to alternate positive and negative current flows. Square/ triangle generators can be simplified by fixing  $V_1$  or  $V_2$  and switching the other by a circuit that monitors integrator output.

load (i.e. the virtual-earth of the amplifier inverting input when used with shunt feedback).

A wide variety of waveforms can be generated by using the voltage across a capacitor subjected to alternate positive and negative current flows. Where the net charging current depends on the currentdifference at the two inputs, novel circuits result. In particular, simplification of square-triangle generators is achieved by keeping  $V_1$  or  $V_2$  (Fig. 7) constant while switching the other from some positive value to zero under the control of a levelsensing circuit that monitors the output of the integrator.

With suitable scaling of the voltages and resistors the polarity of the net current is reversed using only a single diode/ transistor/f.e.t., while the magnitude of that current is determined by an external control voltage. The resulting voltagecontrolled oscillator is markedly simpler than is normally possible. If one or more of the voltages is replaced by a pulsed source, then staircase/ramp waveforms are produced depending on the magnitude, polarity and timing of the pulses. In each of these circuits, the use of a second amplifier can cancel the input current of the integrator amplifier to a first order, reducing the drift to a very small level.

There is no one-to-one correspondence between the circuits designed around operational and current-differencing amplifiers. It will take considerable time and effort to make sure that the advantages of the latter are exploited. The effort will not be wasted.

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## Set 17: c.d.as—signal generation—

## Generators

Circuit description-1 Four types of generator are considered here. They can be summarized as sine, relaxation, square/triangle and d.c. The last of these, the group of circuits normally called voltage/current regulators, can be considered as generators of direct voltage or current. Alternatively, they can be dealt with as d.c. amplifiers with input signals provided by constant voltage/current reference elements, e.g. Zener diodes (see set 16). For cisoidal oscillators the basic requirement is a passive network giving a phase-shift of 0 or  $\pi$  at a single frequency. This may or may not be accompanied by a maximum in the magnitude of the transfer

#### Circuit description—2

The common element to many non-sinusoidal waveform generators is a Schmitt trigger, implemented by an amplifier with positive feedback as shown below. In this amplifier the output changes state rapidly as the input passes through a



given voltage level, reverting to its original state when the input returns to a different level. The difference between the switching levels is the hysteresis, and by using the circuit to monitor the voltage across a capacitor the charging current can have its direction reversed at each transition of the voltage through a switching level. The simplest arrangement is at top right where a single amplifier acts both as Schmitt trigger and as source of charging/ discharging current. The current varies during the cycle according to the usual RC relationships since the output voltage has a constant value for each switched state. If the capacitor forms part of an integrator then the charging current has one of two fixed values, because the virtual earth action prevents the charging voltage across the capacitor from modifying the current. A further modification of this circuit uses an additional diode or transistor to reverse the net current flow to the amplifier by interrupting the current to either the inverting or non-inverting input. Selection of suitable resistors gives a net-charging current that reverses its direction while remaining of equal magnitude and being controlled by an external direct voltage i.e. a

dictates to which input the

The circuit oscillates at the

amplifier phase-shifts cause

be used as above, with the

errors at high frequencies. The

classic phase-shift oscillator can

has zero phase-shift and

feedback should be returned.

frequency for which the loop



nR

amplifier providing more than enough gain. Negative feedback from output to inverting input sets the amplifier gain to overcome the losses in the passive network. Largest signal at amplifier output, least distortion across last capacitor. Make  $R' \ge R$  to minimize loading of network. Any

voltage controlled oscillator results. If the first circuit is adapted to have one stable state then it can be triggered by an external pulse into the unstable state where it remains until the capacitor has

band-pass filter may be inserted into the feedback loop of a system, allowing the overall loop gain to reach the critical condition only at the filter centre frequency. To adapt the more usual Wien networks to current differencing amplifiers, it is sufficient to take any such network whose output has a grounded resistor, and use the current in the resistor as a feedback-the amplifier having a low enough input resistance to act as a virtual earth. No mechanism is shown for amplitude control, but the precisely defined input non-linearities allow for simple solutions as shown later.

completed its charge (or discharge) sequence. The resulting monostable circuit may be edge- or level-triggered.



## Set 17: c.d.as—signal generation—2

## **RC** oscillators



Typical performance  $A_1, A_2: \frac{1}{2} LM3900$ Supply: +20V  $R_1: 220k\Omega$   $R_2, R_3: 33k\Omega$   $R_4: 120k\Omega$   $R_6: 1.8M\Omega$   $R_7: 2.2M\Omega$   $R_8: 5.6M\Omega$   $R_9: 15k\Omega$   $C_1: 0.68\mu$ F  $C_2, C_3: 10\mu$ F  $C_4: 0.1\mu$ F  $C_5: 1\mu$ F D<sub>1</sub>: 1N914 f: 500Hz  $f \approx \frac{1}{2\pi (R_2 R_3 C_2 C_3)^{\dagger}}$  $v_0$ : 7V pk-pk  $\pm$  0.2dB for V<sub>s</sub> 12 to 20V.



current in and hence slope resistance of  $D_1$ . High values to minimize required value of  $C_{\delta}$ . Time constant period of waveform.

 $V_{ref}$ : 0 to 4V. Controls output voltage;  $R_4$  can be varied to increase or decrease the control effect required via  $D_1$ .  $V_8$ : Oscillations sustained at lower supply voltages but amplitude control difficult.

#### **Circuit variations**

• The principle of the circuit is that of an RC oscillator using  $A_1$  and a modified Wien-network, followed by a signal level sensing circuit (A<sub>2</sub>) that modifies the loop-gain of the oscillator feedback loop. Any other RC oscillator can be substituted for A<sub>1</sub> circuit, but amplitude control via a biased diode is optimum, as the diode and amplifier input non-linearities cancel. A peak-sensing circuit is an alternative, and a diode peak detector can replace A<sub>2</sub> with power control of amplitude. • A recently described oscillator uses all four

amplifiers and achieves the very low distortion of 0.1%. Amplifier A<sub>2</sub> is a band-pass filter with  $A_3$  setting the Q. Feedback via the  $100-k\Omega$ resistance sustains oscillations at the filter centre-frequency. Amplifier A4 senses the mean-rectified output and controls the amount of feedback via the attenuation due to the diodes so as to keep the output as a well-defined function of the reference voltage. Resistor values are optimized to stabilize the output against temperature variation.

#### **Circuit description**

The circuit has two distinct sections. Amplifier A1 together with  $R_1$  to  $R_4$  and  $C_2$ ,  $C_3$ constitutes a complete RC oscillator, in which the positive feedback via R<sub>3</sub> exceeds the negative feedback via R4. Choosing  $R_1 \approx 2R_4$ , the amplifier output is biased to  $\approx$  $V_{\rm s}/2$  since the d.c. feedback ensures almost equal currents in R<sub>1</sub>, R<sub>4</sub>. The second amplifier amplifies and filters the output of  $\hat{A}_1$  after a.c. coupling via  $C_4$ , R<sub>5</sub> and half-wave rectification via the non-inverting input of  $A_2$ . The addition of a reference voltage acting via R<sub>7</sub> causes the d.c. output of A<sub>2</sub> to vary the bias current in  $D_1$  in such a way that it diverts a varying proportion of the a.c. feedback from the non-inverting input of A<sub>1</sub>. Allowing for the time constants of the loop, the output of A<sub>2</sub> will vary until the half-wave rectified sine-wave output of A<sub>2</sub> is comparable with Vref. The non-linearity of  $D_1 V/I$  characteristics does not introduce serious distortion, since it matches that of the input characteristics of A<sub>1</sub>. Because the amplifiers are current-differencing units, the conventional Wien bridge has been rearranged, i.e. instead of grounding R<sub>3</sub> and using the p.d. across it as the frequency-dependent feedback, the current through it is fed back, taking the amplifier input as a virtual-earth. The output voltage of A1 has a d.c. term of  $\approx V_{\rm s}/2$  and the superposed a.c. may have a peak-peak value of up to 80% of the supply voltage before the onset of clipping. The relatively low open-loop gain of the amplifier

does not allow very low distortion levels but the ability to control the amplitude of oscillation by an external direct voltage, while using only two out of the four amplifiers within the i.c. is very useful.

#### **Component changes**

R<sub>1</sub>, R<sub>4</sub>: Ratio  $\approx 2:1$  to bias A<sub>1</sub> output  $V_8/2$ , R<sub>4</sub> is chosen to provide less negative feedback than the positive feedback via R<sub>3</sub>, R<sub>3</sub>, C<sub>2</sub>, C<sub>3</sub> i.e. R<sub>4</sub> > 3R<sub>2</sub>. R<sub>2</sub>, R<sub>3</sub>: R<sub>2</sub> = R<sub>3</sub>. Together with C<sub>2</sub>, C<sub>3</sub> defines the frequency of oscillation. R<sub>2</sub> 10k to 1M $\Omega$ . C<sub>2</sub>, C<sub>3</sub>: C<sub>2</sub> = C<sub>3</sub>. 1000pF to 10 $\mu$ F. Large R values can be used with small C values for economy, but layout/screening need to be watched to avoid hum.

C<sub>4</sub>: Not critical.  $0.01\mu$  to  $100\mu$ F. Higher values only required for lower load resistances. Circuit output current should not exceed 1mA for lowest distortion.

C<sub>1</sub>: Must have low reactance at signal frequencies—at least  $0.5\mu$ F for 1kHz.

R<sub>8</sub>, R<sub>7</sub>, R<sub>8</sub>, R<sub>9</sub>: Control mean



## Set 17: c.d.as—signal generation—3

## Voltage-controlled oscillators



Typical performance IC:  $\frac{1}{2}$  LM3900 Supply: 10V R<sub>1</sub>: 220k $\Omega$ R<sub>2</sub>: 270k $\Omega$ R<sub>3</sub>, R<sub>4</sub>: 1M $\Omega$ R<sub>5</sub>, R<sub>6</sub>: 1.2M $\Omega$ C<sub>1</sub>: 1nF Square-wave output: 9V pk-pk Triangular wave: 7.5V pk-pk Frequency: 610Hz



 $V_s$ : +4 to +30V.  $V_{ret}$ : 0 to 30V (higher if  $R_1$  to  $R_s$  increased).

## **Circuit variations**

• Using an inverting Schmitt, but with the feedback shunting the current from the non-inverting input of the integrator leaves the phasing correct for sustained oscillations. Diode compensation can be added to this and the previous circuit such that  $f \rightarrow 0$  as the input pot. is set to zero. • Improved performance at high frequencies is possible using a transistor rather than a diode to direct the integrator input current; inversion provided by the transistor requires inversion of one or other of the other circuits. Because the transistor conducts for a rise in the Schmitt output of little more than one  $V_{be}$ , the rise time of the Schmitt no longer presents such a serious limitation to the frequency at which oscillation can be sustained.

• Adding bias resistors to +V sets a minimum value to the frequency of oscillation for  $V_{ref} = 0$  i.e. the frequency can be controlled over any prescribed range. (Bottom)

### **Circuit description**

A basic method of generating square/triangle waves is to interconnect an integrator and a Schmitt trigger circuit. With correct phasing the integrator output changes linearly with time between the threshold voltages of the Schmitt, causing the output to reverse and with it the direction of the integrator output slope. To bring the frequency under the control of an external voltage the switched output is used to reverse the net current into the integrator, but leaving the magnitude defined by the control voltage. A current differencing amplifier is particularly suited to this technique, since the net charging current is the difference between the two input currents. If these are set in the ratio 2:1 and the larger current is switched off, the net current changes sign, remaining of equal magnitude i.e. a triangular wave results at the integrator output with peak-to-peak voltage equal to the hysteresis of the Schmitt trigger circuit. Consider the Schmitt output

positive; the diode is reverse-biased and with  $R_1 + R_2 = R_3/2$ , the net integrator input current is negative. Integrator output ramps negatively until the lower threshold voltage of the Schmitt is reached. Output of the Schmitt falls almost to zero and the diode conducts by passing the current in  $R_1$  to ground through the output of the Schmitt circuit. (A residual current flows in  $R_2$  since the output of the Schmitt is typically a 100mV but the current is small compared with the normal charging current.) The net charging current is now positive and the integrator output ramps positively until the cycle is completed on chargeover at the upper threshold level. The rate of charging is proportional to  $V_{ref}$  which may be derived from a potentiometer across the supply or from an external source with  $f \propto V_{ref}$  to a good approximation.

#### **Component changes**

R<sub>1</sub>, R<sub>2</sub>, R<sub>3</sub>: For equal positive and negative slopes to the triangular wave (and hence unity mark-space ratio for the square wave).  $R_1 + R_2 =$  $R_{\rm s}/2$ . To minimize loading on source these resistors may range from 100k to 10M $\Omega$ .  $R_1 \approx R_2$ but may be different to use preferred values while meeting above conditions. C: sets frequency of oscillation and  $f \propto 1/C$  except at high frequencies where finite rise and fall times limit the response. Typically C is 1n to  $10\mu$ F. R<sub>4</sub>, R<sub>5</sub>, R<sub>4</sub>: control upper and lower threshold voltages and have peak-peak value of triangular wave. For  $R_5 = R_6$ and  $\bar{R}_4 = kR_5$ , the hysteresis is  $\approx k$  (square-wave output). Reducing the hysteresis increases the frequency of oscillation, other components being equal. Amplifier slew-rate of  $0.5V/\mu s$  makes it difficult to obtain outputs above 10kHz. D: Not critical. Any general-purpose silicon diode.



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## Set 17: c.d.as—signal generation—4

## Voltage regulators



### **Circuit description**

When negative feedback is applied around a current-differencing amplifier such as the LM3900, the inverting input terminal remains at 0.6V at 0 °C falling to nearly 0.4V at 75 °C. In the voltage regulator shown the voltage across  $R_1$  is defined by these values, fixing the current in Tr<sub>1</sub> which acts as a Zener diode. Amplifier input current is very small ( $\approx$  30nA) allowing currents between 1µA and 1mA to be adequately defined. Output voltage is the sum of the input  $V_{be}$  and the Zener voltage. Two sources of variation exist:

-direct variation in input voltage  $(V_{be})$  as above of about -2.5mV/deg Ctogether with a fall of 2 to 3mV for each 1V increase in the supply. -corresponding variation in Zener voltage caused by the resulting current variation

together with its own

temperature coefficient. If a Zener diode is available with a positive temperature drift cancellation between that drift and the input drift is possible, though complicated by the current variations. The use of a small-signal planar transistor with Veb breakdown of about 7V gives partial cancellation while using the Vec breakdown adds about 0.6V because of the forward biased base-collector diode. The additional negative drift over-compensates, for the particular device tested, but the method allows good cancellation. Conventional

Typical performance Supply: +15V IC: 1 LM3900 Tr1: BC125\*  $R_1$ : 10k $\Omega$ Vo: 7.7V  $R_1: 2k\Omega$ ⊿V₀: +15mV (for  $\Delta V_s = -5V$ ) +25 mV (for  $\Delta T =$  $+25 \deg C$ \*Use of reverse base-emitter junction gives good low-current breakdown. Using reverse collector-emitter raises voltage by 0.6V, changes drift by about -2mV/deg C.

Zener diodes would be preferred where a quick range of voltages is required or close tolerance (but see circuit modifications).

#### **Component changes**

Tr<sub>1</sub>: Any Zener diode may be used in principle since the output current is limited low-current operation is an advantage, and any planar transistor may be used as reverse base-emitter breakdown typically lies between 6 and 12V, being well-defined for a particular transistor type. R1: defines diode current as  $V_{be}/R_1$  where  $V_{be}$  is that of the i.c. input transistor ( $\approx 550 \text{mV}$ at room temperature, -2.5mV/deg C). Select for Zener current of say 100µA to 2mA depending on type, down to  $1\mu A$  for planar diode/transistor.  $V_{s}$ : 4 to 36V (down to 3V in some cases).  $V_0$ : up to 36V. Limited by available diodes but see circuit modifications.

 $I_0$ : 0 to 10mA. May be boosted by adding emitter follower at output of amplifier (*inside* feedback loop).

## **Circuit modifications**

• Additional temperature compensation is introduced by adding resistance in series with the Zener diode. Currents in the diode and resistors are almost equal since the amplifier input is small ( $\approx$  30nA). Hence  $V_0 = V_z + V_{be} + nV_{be}$ . Each equivalent  $V_{be}$  increases the output by  $\approx$  0.55V, adding a temperature drift term of -2.5mV/deg C. (Top circuit.)



• To increase the output voltage Zener/transistor (e.g. selected for its temperature dependence) the diode is tapped onto a portion of the output voltage. The potential divider should not be significantly loaded, nor should it constitute a significant load on the output so a low current breakdown device is preferred. With a chain current of 1mA the diode current should not exceed  $10\mu$ A for good accuracy. (Middle circuit.)



Cross references Set 6, cards 1, 2, 7 & 10.



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## Set 17: c.d.as—signal generation—

## **Constant-current circuits**



### **Circuit description**

Many circuits used with conventional operational amplifiers are adaptable for use. with current differencing amplifiers. To provide a constant-current drive to a grounded load is difficult in both cases since each amplifier has its output referred to ground so that feedback can only be taken in shunt with the load, i.e. lowering the output impedance. If the load can float then the feedback may be taken across a grounded resistor in series with the load. An alternative technique combines negative and positive feedback in such a way that at some point in the network the output impedance tends to infinity. By injecting voltages/currents at other points in the system, a load connected between that particular point and ground will carry a current independent of the load resistance. As shown, R4 is used to adjust the negative feedback such that for very high load resistances, the negative and positive feedback are almost in balance. In the ideal case, this is achieved for  $(R_3 + R_4)/R_1 = R/R_2.$ Current in the load may then be varied in one of three ways, each of which has minimal effect on the output resistance: variation of the input voltage V; insertion of an alternative/additional voltage in series with R<sub>1</sub>; variation in the tapping point of the resistor R. The last method can be used to obtain a 5:1 spread in sensitivity while retaining a high output resistance. Using

a single polarity supply restricts

the output to one polarity (+)

but dual-polarity supplies

Typical performance IC:  $\frac{1}{4}$  LM3900 Supply: +15V R<sub>1</sub>, R<sub>2</sub>: 10k $\Omega$ R<sub>3</sub>: 8.2k $\Omega$ R<sub>4</sub>: 2.2k $\Omega$  for particular circuit R: 10k $\Omega$ k: 0.5 V: 7.7V for IL 1.5mA

would permit bipolar outputs for handling d.c. signals without the need for blocking capacitors.

### **Component changes**

Vs: +4 to +36V. Because of losses in feedback-defining resistors load voltage swing is always less than amplifier capability.

R<sub>1</sub>, R<sub>2</sub>:  $10k\Omega$  is near lower end of working range, with amplifier normally operating with input resistor of  $100k\Omega$  upwards. However, values required where current of several mA required in load.

R<sub>3</sub>, R<sub>4</sub>: If high impedance is sufficient and it is not required to trim for output impedance  $\rightarrow \infty$ , replace by fixed resistance  $(R_3 + R_4 = R \text{ for } R_1 = R_2.$ Negative output resistance if feedback resistance to inverting input is too large. k: Can be varied from 0.2 to 0.9. At high values, relatively large currents obtainable but with some loss of accuracy. V: Variable over range 2 to 10V. For input voltages too low, V<sub>be</sub> errors at input prevent linearity of IL against V.

## **Circuit modifications**

• By adding transistors, the feedback may be taken from across one emitter resistor as would be the case for a voltage amplifier (top circuit.) The output is taken from the transistor collectors each or which carries a collector current almost equal to its emitter current if the transistors have high current gains. In the simplest arrangement the current is defined by the amplifier internal  $V_{be}$  with a consequent temperature



dependence to the currents (which may be set to any required ratios by varying  $R_1$ ,  $R_2$ , etc.). Any of the techniques used in voltage regulators may be used to achieve better control over the voltage across  $R_1$ .

• A voltage-controlled current sink as shown forces the voltage across the  $1k\Omega$  resistor to match the input voltage giving an overall transconductance of 1mA/V. Bias provided by the  $1M\Omega$  resistors linearizes the characteristic at low  $V_{in}/IL$ .



• A current source using this principle is possible but has some disadvantages. It is the voltage across R1 that needs to be defined, but approximate equality of the p.ds across the  $1M\Omega$  resistors that is achieved. Where the voltage across  $R_1$  is to be low (say 1V) to maximize load voltage, small unbalance in the input current sensitivities produce small voltage differences at A & B. These small errors then represent a large error in the small voltage across  $R_1$  and hence in the load current. (Bottom circuit.)

Cross references Set 6, cards 4 & 12.

## Set 17: c.d.as—signal generation—6

## Schmitts & comparators



Typical performance IC:  $\frac{1}{2}$  LM3900 Supply: + 15V R<sub>1</sub>: 1M $\Omega$ R<sub>2</sub>: 2.2M $\Omega$ R<sub>3</sub>: 10M $\Omega$ 

$$egin{aligned} V_{\mathrm{u}} &= + V igg( rac{R_1}{R_2} + rac{R_1}{R_3} igg) \ &+ V_{\mathrm{be}} igg( 1 - rac{R_1}{R_2} - rac{R_1}{R_3} igg) \ V_1 &= + V igg( rac{R_1}{R_2} igg) \ &+ V_{\mathrm{be}} igg( 1 - rac{R_1}{R_2} - rac{R_1}{R_3} igg) \end{aligned}$$

V<sub>be</sub>: of input transistors assumed equal and  $\approx 0.5$ V. Hysteresis,  $V_u - V_1 = + V R_1/R_3$ Measured hysteresis: 1.51V. Output rise-time:  $30\mu s$ (corresponding to 0.5V/ $\mu s$ slew rate limitation for positive-going output) Output full-time:  $2\mu s$ 

## **Circuit description** The input section of current-differencing amplifiers rely on the accuracy of matching of transistors in a current mirror to define the operating levels. It is not possible to obtain the precision of switching levels by such a method as can be achieved using a long-tailed pair and more conventional circuitry. Where a moderately wide hysteresis, say, 1V or greater, is acceptable and neither upper or lower thresholds need be defined to within a small percentage, Schmitt circuits can be simply constructed offering the other advantages of positive and rapid switching. The limited slew-rate of the amplifier makes it comparable in performance in this respect to op-amps such as the 741, but the internal compensation cannot be removed to allow speeds comparable to the uncompensated op-amps (301, 748, etc) let alone high-speed comparators. The main uses are likely to be as part of signal processing systems, alarm

Switching takes place, assuming ideal current mirror action, when the current in  $R_1$  equals the sum of the currents in  $R_2$ ,  $R_3$ . In each case the amplifier internal  $V_{be}$  of about 0.5V has to be taken into account, effectively raising both upper and lower thresholds by somewhat less than 0.5V. To a first order approximation, hysteresis depends only on the resistor values, while the input resistance may be made > 1M $\Omega$ .

circuits, waveform generators,

etc.

This is because the inverting stage input current of  $\approx 30$ nA allows reasonable accuracy of switching levels even though the currents in the resistors are only a few microamperes. Increasing R<sub>3</sub> reduces the upper threshold without changing the lower one. Increasing R<sub>2</sub> reduces both thresholds by the same amount leaving the hysteresis unchanged. Increasing R<sub>1</sub> raises both threshold voltages proportionately,

leaving the hysteresis in the same ratio to each as prior to the increase.

## **Component changes**

Supply V: +4 to 36V. Usual range for given i.c.  $R_1$ : 100k to 10M $\Omega$ . The lower value is particularly useful if voltage comparator action is required for voltages < 1V. Each input is pre-biased to +V through 1M to  $5M\Omega$  resistors. R<sub>2</sub>: As above. If used as a comparator R<sub>s</sub> is absent, and comparison is possible between unequal voltages i.e. the output can be made to swing between its limits as V<sub>in</sub> passes  $V_{+}/n$  if  $R_{2} = R_{1}/n$ . Comparison of negative voltages are also possible provided bias network accommodates common mode currents. Since currents are being compared, the use of suitably large resistors allows comparison of very large positive or negative voltages, but with the limited accuracy implied by the current-mirror input.

## $R_8$ : 1M to 22M $\Omega$ . Sets



hysteresis and would be absent for comparator action. Can generally be added to any comparator arrangement to introduce, for example, small amounts of hysteresis to prevent output jitter due to noise.

### **Circuit modifications**

• The position of the input and reference voltages (+V was used in the circuit overleaf) may be interchanged to give a non-inverting Schmitt. The availability of inverting and non-inverting Schmitts is of importance in the design of such circuits as wave-form generators. The thresholds and hysteresis are approximately:  $V_n = + V R_n/R_n$ 

$$V_{\rm u} = + V R_{\rm s}/R_{\rm i}, V_{\rm l} = + V \left( \frac{R_{\rm s}}{R_{\rm l}} - \frac{R_{\rm s}}{R_{\rm s}} \right)$$

 $\therefore$  hysteresis =  $+ V R_2/R_3$ . • If a p-n-p long-tailed pair is added at the input the resulting amplifier may be used as a comparator or as a Schmitt (as shown). The input impedance is extremely high if high-gain low current transistors are available, and any degree of matching for the input transistors, may be chosen for high precision. It is necessary to clip the output voltage swing more precisely (with a resistor-Zener combination for example) if the hysteresis is to be equally precise.

• If two amplifiers are used then the matching is that between corresponding inputs which can be better than between + and — inputs of a single amplifier. One amplifier plus the output transistor has overall negative feedback (transistor is an inverter) and the equal current forced into the two non-inverting inputs forces the transition point in the second amplifier output to occur for equal currents at the two inverting inputs. Again comparison can be made between unequal voltages but more account has to be taken of the effects of the small input bias currents.

Cross references Set 2, all cards. Set 17, cards 3, 7, 8 & 9. 117

## Set 17: c.d.as—signal generation—7

FREQUENCY (kHz)

VOLTAGE (V)



### **Circuit description**

The amplifier together with resistors R<sub>2</sub>, R<sub>3</sub>, R<sub>4</sub> constitutes a Schmitt trigger. The output switches between the two extremes ( $\approx +100$ mV and  $\approx 0.8V$  below V+). When the output is low  $C_1$  is discharged through R<sub>1</sub> until the lower threshold of the Schmitt is reached. The output switches rapidly to its most positive value and the capacitor is charged positively. When the upper threshold is reached the Schmitt output is reversed and the cycle restarts. If R<sub>2</sub>, R<sub>3</sub>, R<sub>4</sub> are chosen to make the threshold voltages  $V_+/3$  and  $2V_{\pm}/3$  respectively then the voltage across the capacitor swings between these limits, while the output approximates to a square wave with almost unity mark-space ratio. For the highest accuracy a number of conditions must apply: The current in R<sub>2</sub> must be very low compared with the charging current through  $R_1$ ; the output voltage swing should be as nearly as possible equal to the supply voltage; and the accuracy of the current-differencing action should be high (this places a restriction on the minimum currents in  $R_2$ ,  $R_3$ ,  $R_4$ ). If these constraints are met then both the frequency and the mark-space ratio remain substantially constant over a wide range of supply voltages. Any increase in supply voltage produces proportional and hence self-cancelling increases in the magnitude and rate-of-change of voltage across the capacitor. The potential difference across the capacitor is uni-directional permitting the

use of electrolytic capacitors for low frequency generation.

## **Component changes**

 $R_2, R_3, R_4$ : These must be chosen jointly such that the firing points of the Schmitt trigger they constitute are  $V_{+}/3$ and  $2V_+/3$  (see circuit description).  $R_3 = R_4 = 3R_2$ gives this condition. Make resistors as high as possible to avoid loading R1, C1 charging circuit i.e.  $R_2 > 100R_1$ . Typically  $R_2$  1M to 10M $\Omega$ . R1: This sets oscillator frequency together with C. Too low a value loads the output. too low is loaded by R<sub>2</sub>. Typically 4.7k to  $100k\Omega$ . C: Max frequency of oscillation  $\approx$  20kHz corresponding to C = 470 pF with  $R_1 = 15 \text{k}\Omega$ . Capacitor voltage always positive allowing use of electrolytic to give low frequencies. R<sub>L</sub>: Maximum output current  $\approx$  10mA giving minimum  $R_{\rm L} = 1 \mathrm{k} \Omega$  at  $V_{\rm s} = 10 \mathrm{V}$ . Generally keep  $R_{\rm L} > 2.2 k \Omega$  to

Generally keep  $R_L > 2.2k\Omega$  to allow output to swing close to supply.

## Circuit modifications

• As with other astable multivibrators, this oscillator can be synchronized to an external waveform. Because the Schmitt action is achieved using high value resistors, the synchronizing input may also have a high input impedance, while locking can be achieved with 100mV. Locking to a sub-harmonic of an incoming signal is also possible, though for frequency division ratios in excess of 5:1 carry the risk of locking to the wrong sub-harmonic. (Top circuit.) • Where relatively narrow pulses are required, the charge and discharge time constants should be as widely different as possible. The charging action is via  $R_1$ , a low resistance with a small current diverted into  $R_2$ . When the output is low the diode is non-conducting, and the small current in  $R_2$  gives a long discharge time, i.e. a long space. (Middle circuit.)

VOLTAGE, V<sub>o</sub> (V pk-pk)

OUTPUT



Other forms of multivibrator may require more than one amplifier but the bottom circuit has some similarity of behaviour to the programmable as uni-junction. The capacitor is charges through R<sub>1</sub> until the upper threshold is reached. The output swings towards zero he rapidly discharging the capacitor. The long mark has its duration controlled by the

direct voltage applied to  $R_1$ (+V may be replaced by an external control voltage giving a voltage controlled oscillator). The space is of short duration because of the relatively large discharge current.

## **Cross references**

Set 3, cards 2, 4, 6 & 8. Set 8, all cards. Set 10, card 2. Set 17, cards 3, 8 & 10.

## Set 17: c.d.as—signal generation—8

## Monostable multivibrators



Typical performance Supply: +10VR<sub>1</sub>, R<sub>3</sub>:  $1M\Omega$ R<sub>2</sub>:  $10M\Omega$ R<sub>4</sub>:  $100k\Omega$ C<sub>1</sub>: 220pFC<sub>2</sub>: 100pFD<sub>1</sub>: 1N4148f: 100Hzv<sub>in</sub>: 0.5V pk-pk square-wave Pulse-width: 0.66ms



## **Circuit description**

The circuit is closely related to the pulse generator modification of the astable circuit, and to the unijunction equivalent (card 7). The feedback network is designed so that the circuit normally latches into one particular state, with the output high. This is because in the low stage the diode inhibits current flow into the inverting input. Current in R<sub>2</sub> drives the output high initially and with the sum of the currents in R. and  $R_3$  exceeding that in  $R_1$ . The first negative-going pulse drives the output temporarily low, the diode becoming reverse-biased. The capacitor discharges through R<sub>1</sub> with only R<sub>2</sub> contributing any current to the non-inverting input. Discharging continues until the voltage across the capacitor is down to a fraction of the supply given by  $R_1/R_2$ . At this point, the inverting input current comes into balance with that at the non-inverting input, the output begins to rise and the positive feedback causes the cycle to complete itself rapidly. The diode conducts, rapidly changing the capacitor to a volt or so below the positive supply voltage and the circuit is ready for the following trigger pulse. The positive-going slew-rate limit of  $0.5V/\mu s$  limits the minimum pulse-width that can be defined with any accuracy. For a 20-V supply and a pulse of nominal length 100µs this slew rate would add up to 45% to its effective length.

#### **Component changes**

+V: +4 to +36V. Monostable action with increasing pulse-width and lowered output is possible to below 2V with some samples though outside specification.

 $C_1$ : Large values may be used (provided leakage resistance is  $1M\Omega$ ) and with  $R_1$  at  $1M\Omega$ time-delays of minutes are possible.

 $C_3$ : Used to differentiate input signal to provide short-duration positive transients. 47p to 220pF for first input pulses; increase for slower input rise times.

R<sub>1</sub>: Ratio  $R_1/R_2$  sets fraction of supply voltage across C<sub>1</sub> at which the pulse period is completed. Typically  $R_1/R_2 =$ 0.1 to 0.3. If ratio is too low the period lengthens considerably but is ill-defined. R<sub>2</sub>: Current in R<sub>2</sub> should not fall below 1 $\mu$ A if timing is to be controlled by RC values. R<sub>3</sub>: With R<sub>8</sub> = R<sub>1</sub> the smaller current in R<sub>2</sub> is sufficient to keep the output high in the quiescent state.

 $D_1$ : Any general-purpose silicon diode.

#### **Circuit modifications**

• Positive-going output pulses can be obtained with a similar circuit if the permanent bias is applied to the inverting input, the direction of the diode reversed and a changing path provided in parallel with the diode. The quiescent condition has the output low with the diode conducting and significant current only in the  $10M\Omega$  resistor. A positive-going input transition drives the input high, which state is held until the capacitor charges sufficiently to allow the current in the inverting input to exceed that in the non-inverting input. A major

advantage of an amplifier having both inputs accessible is that the signal pulses can be transferred to the other input. In this version, the output pulses would then be triggered by negative rather than positive transitions, while in the previous case which required negative transitions the response would then be to positive edges.

• Two-amplifier monostable circuits can have two opposite polarity outputs and allows separation of the triggering timing and output sections more readily. The circuit shown is equivalent to a Schmitt trigger followed by a monostable although the functions are combined in a single circuit. Out, is normally low contributing no current to the inverting input of the lower amplifier. Hence Out<sub>2</sub> is only driven low when Vin reaches  $\approx 0.8 V_{in}$ . Provided the input

falls below this level prior to the end of the pulse then generated, the circuit is suitable for resetting an oscillator when its ramp output exceeds a given d.c. level.



Set 17: c.d.as—signal generation—9

## **Flip-flops**



### **Circuit description**

The circuit has strong affinities with standard two transistor flip-flop and could be triggered at either of the inverting inputs via diodes to provide a set/reset function. Non-inverting inputs provide the additional facility of triggering the flip-flop to provide a  $\div 2$  action. Equal currents are fed into the non-inverting inputs on receipt of a positive-going edge but that amplifier whose output is already high because its inverting input receives no current will be initially unaffected. The other amplifier receives a short-duration current flow at its non-inverting input that exceeds that at its inverting input, and its output begins its positive-going transition. The speed-up capacitor couples this to the inverting input of the other amplifier and the positive feedback around the loop causes the transition to complete itself. Thus each output changes state in opposite directions on the receipt of each positive-going input transition. Two such transitions are required to complete each cycle of the output, and the output waveforms are complementary square waves at half the frequency of the input.

## Such circuits may be readily cascaded to develop simple

counters. Their low speed offers no competition to c.m.o.s., t.t.l. etc. but the ease with which they can be combined with other analogue/digital functions gives them a considerable advantage in many systems. Typical performance IC:  $\frac{1}{2}LM3900$ Supply: +10V R<sub>1</sub>, R<sub>2</sub>: 100k $\Omega$ R<sub>3</sub>, R<sub>4</sub>: 1M $\Omega$ C<sub>1</sub>: 100pF C<sub>2</sub>, C<sub>3</sub>: 56pF v<sub>0</sub>: 9V pk-pk Max input frequency for satisfactory toggle action: 10kHz Min. input amplitude: 4V pk-pk

#### **Component variations**

+V: Standard supply range +4 to 36V. Samples may operate down to < 2V. R<sub>1</sub>, R<sub>2</sub>: Sensitivity can be controlled by varying R1, R2 for fixed values of R<sub>3</sub>, R<sub>4</sub> and vice versa. Values are not critical and may be adjusted to suit a wide range of input pulses. 10k to  $1M\Omega$ .  $R_a$ ,  $R_4$ : These provide the current to keep the opposing amplifier saturated and are again not critical. 330k to  $10M\Omega$ could be used but as in any pulse system high resistances can be accompanied by spurious triggering. Generally the lower resistances are required at low supply voltages.  $C_2$ ,  $C_3$ : Speed-up capacitors ensure positive transitions but the maximum speed is limited by amplifier slew-rate. C<sub>1</sub>: Not critical. 47p to 470pF.

## **Circuit modifications**

• Using a single amplifier, one or more diodes are used to channel the incoming pulses to the two inputs alternately, with overall positive feedback to latch the circuit. The quiescent state with the output high is sustained because both R2 and R<sub>a</sub> contribute current to the non-inverting input, exceeding that in  $R_1$ . When the output goes low it remains there because now the non-inverting input receives current only via  $R_2$ , leaving the current through R, into the inverting input to dominate. In the latter state  $D_1$ prevents any input positive-going pulse from reaching the inverting input and the circuit is triggered via R<sub>4</sub> into the high state. The



succeeding positive pulse produces approximately twice the current through  $R_5$ ,  $R_6$  as through  $R_4$  ( $D_1$  is reverse biased), the output going low.  $C_2$  is the speed-up capacitor and  $D_2$  absorbs the negative transitions.

• Careful choice of the positive feedback in this circuit (middle) provides sufficient hysteresis to keep the output latched into a given state after an input pulse has returned to zero. Any further pulses on that input have no effect, and it requires



a similar pulse on the other input to reverse the output. The input which drives the output to logic 1 is the set input, the other the reset and the circuit is an asynchronous RS flip-flop. • To add set, reset facilities to the initial circuit, one of the amplifiers must be directly driven into its off-state by a current greater than the current pulse received from the input trigger. This overrides the trigger for that amplifier, while the other is automatically set with its output to logic 1 and the input pulses are again ineffective. A transistor may replace the switch. Shown in the bottom circuit.

## Cross references

Set 2, cards 3, 8 & 9. Set 14, card 1. Set 17, cards 7 & 8.

## Set 17: c.d.as—signal generation—10

## Staircase generators



## Typical performance

Supply: 7.5V R<sub>1</sub>, R<sub>4</sub>: 1MΩ R<sub>2</sub>: 120kΩ R<sub>3</sub>: 330kΩ R<sub>6</sub>: 10MΩ R<sub>6</sub>: 1.8MΩ C: 1μF D<sub>1</sub>: 1N914 Input: pulse train +12V peak Mark/space ratio: 2:1 Repetition rate: 30kHz Output 1: staircase waveform 3.6V pk-pk, frequency 800Hz Output 2: 6.8V pk-pk, frequency 800Hz

### **Circuit description**

A staircase waveform can be generated by feeding pulses of current to a capacitor. The resulting circuits are closely related to those used in frequency meters such as the diode-pump circuit and its variants. One version as shown feeds the pulses to either the inverting or the non-inverting input of the first amplifier depending on the state of the output of the second. Consider output 2 positive. The diode is reverse biased and with  $(R_2+R_3) = R_1/2$  the net positive current into amplifier 1 is positive. During each positive pulse output 1 ramps positively. remaining relatively constant between pulses. The smaller the mark-space ratio the more nearly the output waveform approaches a staircase waveform, with the rising steps occupying a small proportion of the time. At some positive voltage set by the resistors R4, R5, R6 the Schmitt trigger operates, its output switching to near ground level. The diode becomes forward-biased diverting all subsequent positive pulses from the non-inverting input. The net negative input current via R1 produces a staircase waveform in the negative direction until the lower threshold voltage of the Schmitt is reached and the cycle recommences. In essence the circuit is that of a square-triangle generator in which the direct control voltage is replaced by a succession of positive pulses giving a stepped characteristic to the triangular wave. The mean input voltage determines the mean charging

current and the average slope of the ramp. If the frequency of the input signal is varied while maintaining a constant pulse height and constant mark-space ratio, then the basic repetition rate of the output remains unchanged. (It is composed of a number of steps dependent on the ratio of input frequency/output frequency.) Varying the pulse height, the mark-space ratio or Schmitt trigger points varies the output frequency.

## Component changes

 $R_4$ ,  $R_5$ ,  $R_6$ : The magnitudes of these resistors are not critical and they may range from 10k to  $10M\Omega$ . Design procedure for the trigger levels is given on Schmitts and comparators card. Hysteresis  $\approx + VR_{\bullet}/R_{\bullet}$ . R1, R2, R3: For comparable positive and negative slopes,  $R_{2}+R_{3}=R_{1}/2$ . Ratio  $R_{2}:R_{3}$  is not critical but neither should be  $< R_1/10$ . R<sub>1</sub>: 10k to 1M $\Omega$ . C: Controls the basic repetition rate of the output for given input waveform. 220p to  $10\mu$ F. Output frequency can be very much less than input frequency with division ratios in excess of 100

## **Circuit modifications**

• When the input is at zero, current flow to ground through  $R_1$  allows the stored voltage on C to vary. Inserting a diode prevents this at the expense of a reduced effective pulse height. Most other v.c.o. circuits can be adapted to perform this function. Replacing the diode by a transistor and controlling the pulses to the inverting input leaves the phasing correct. (Top.)

 To obtain a staircase waveform followed by a sharp transition, the Schmitt may be followed by a monostable circuit that provides a pulse of current to discharge the capacitor rapidly. Input pulses are fed to a single input as shown, and again alternative arrangements are possible by phasing input/feedback appropriately. Switching and timing actions can be combined in the single monostable shown, where the positive output pulse is initiated at a particular input

amplitude, the timing remaining correct if the input voltage to the monostable falls below this firing level before the completion of the monostable output pulse. (Middle circuit.) • Circuits designed for tachometer applications give an inherent staircase output. Further, the step height is controlled only by the input pulse height and the values of  $C_1$ ,  $C_2$ , provided that the input pulse width is sufficient to allow full charge and discharge of the capacitor  $C_1$ . The version shown has a frequency doubling action with the positive input transitions being fed to the non-inverting inputs and the negative transitions to the inverting inputs.

Cross references Set 3, cards 3 & 7.



Set 17: c.d.as-signal generation Up-date

1. A voltage-controlled oscillator is a basic tool in a number of instrumentation systems. In this example, a thermistor network having a linear variation of resistance with temperature is used as a temperature sensor. Placed in the feedback path of a current-differencing amplifier. the current through it can be controlled by the current injected into the non-inverting input. Thus the voltage output of A<sub>1</sub> is a linear function of temperature, and is buffered by  $A_2$  to give a suitable offset scale-this allows the output frequency to be a



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temperature on the scale chosen  $(10 \times \text{temperature in} \text{deg F in the example given} \text{in the reference})$ . The resistance of the thermistor falls with rising temperature causing the output of  $A_1$  to fall, and of  $A_2$  to rise, thus increasing the frequency. The reference gives full calibration procedures for a particular thermistor network of U.S. origin but any element or network having a linear p.d. against temperature could be substituted. Using a diode or amplified diode network, the rate-of-change of voltage with temperature would be less and the temperature range might conveniently be 0 to 100°C rather than 14-113°F in the original circuit.

## Reference

Rufer, R. P. Circuit built with quad op-amp measures temperature digitally, *Electronic Design*, 1974, vol. 6, March 15, p. 160.

2. It is a commonplace of design that functions can be performed by particular devices/circuits, not planned for by the original manufacturer. They may be well within the device ratings, but just not mentioned on the data sheets. The circuit shown is a simplified form of basic uncompensated op-amps such as the 101/748 types and shows that both a current mirror and an output stage with inverting driver are present. These are normally driven by the input differential amplifier. The author of the reference article observed that the input stage can be



inhibited by returning both inputs to the negative supply rail. This leaves pins 1 and 5 (normally used for compensation purposes) available to act as the inputs of what is effectively a current-differencing or Norton amplifier. In principle, all the circuits in sets 16-18 can be implemented, while returning the inputs 2 and 3 to their usual voltage levels would allow one set of inputs to override the other and provide complex functions. The reference uses an audio mixer circuit as an illustrative example using both singleended and dual supply voltages.

## Reference

Jung, W. G. Op-amp in current-differencing mode becomes a non-inverting audio mixer. *Electronic Design*, 1974, vol. 6, May 10, p. 130.

3. Some of the circuits described in sets 16-18 use diodes and/or the input current mirror to rectify the input signal: careful design allows the mean or peak output to be a linear function of the input, but the circuits still cannot respond to a.c. signals with amplitudes below p.ds. If the rectifying circuit is placed in the feedback path of the amplifier this limit is



removed. With a bridge rectifier a peak-peak output swing of at least 2.5V is required excluding voltage drops across the meter movement. This is well within the amplifier capability for supply voltages of 5V and above, provided the output is biased appropriately—to the supply mid-point for maximum swing. Two additional components are then needed in the feedback path, viz. a blocking capacitor  $(220\mu F)$ and a resistor  $(4.7M \Omega)$  to set the quiescent output voltage in conjunction with the resistor to the non-inverting input ( $10M \Omega$ ). The meter reads the full-wave-rectified mean value of the input current assuming an ideal amplifier, and that current is given by  $v/R_{in}$ . The frequency response covers the audio band but it is not suited to high frequency measurements.

## Set 18: c.d.as — measurement and detection

A favourite theme of one of the Circard authors is to identify differences between and similarities of circuits of related function. Fitting circuits into a kind of family tree not only aids understanding, but can lead to development of new circuits. Peter Williams' article on Wien oscillators (WW Nov. 1971 pp. 541-7) is a good example of studying circuits collectively, and you will see the theme recurring in the Circard background articles from time to time. Often different realizations of similar circuits are quite unrecognizable when dressed in the clothing appropriate to the application. As the next article shows, there are strong similarities between rectifier circuits and sampling circuits, the last-mentioned using a switch to replace the rectifier diode. Similarly, when the resistive load is replaced by a capacitative one, the peak rectifier becomes a sample-and-hold circuit (cards 7 & 8). A feature of current differencing amplifiers in sampling circuits is the relatively long hold time possible but so far they are not able to compete with standard op-amps in respect of accuracy. The feature common to most of the circuits in this set is simplicity, whether the application be providing negative resistance, interfacing with transducer, frequency measurement or defining currents for subsequent voltage measurement. For example, use of matched transducers delivering a variable current and followed by the current differencing circuit is a very simple way to sense differences in light intensity and temperature. In this set, card 1 forms an excellent summary card: read it first for a quick look at some of the possibilities.

Measurement and detection 1 Logic circuits 2 Phase-locked loop 3 Transducer driving 4 Semiconductor device testing 5 Negative resistance 6 Peak/mean rectifiers 7 Sample and hold circuits 8 High-frequency circuits 9 Tachometers 10

# **Measurement and detection**

Pattern recognition is one sign that a technology is reaching maturity. The early stages following new advances are a succession of bright ideas, half-worked-out theories and unrelated developments. This is inevitable as workers in many areas take from the original material that which meets their needs—or appeals to their prejudices.

In circuit design the same configurations appear under many guises and names, developed quite independently and for different applications. If we can recognize these similarities and construct the appropriate family tree this is worthwhile in itself.

But we can do more. If two circuits are similar in form because related in function, then by finding any other circuit designed for one of the functions there is a good chance that it can be modified to provide the other. A good designer is one who picks the best brains.\*

The present topic is a particularly good illustration of this thesis. The problem is to measure some property of the amplitude of an a.c. waveform. Four circuits have their properties listed in the table and circuit diagrams representing a basic feedback form of each are shown in Figs 1 to 4. The configurations are identical, the differences lying only in whether conduction is through a diode or a switch, and whether the load is resistive or capacitive. This identity of form is far from apparent in practical versions since there are so many additional components and sub-circuits to optimize the response or effect coupling between other circuits/transducers.

The half-wave rectifier uses a diode as does the peak rectifier. It begins conduction through the diode as soon as the input goes positive remaining in conduction for the phase angle range 0 to  $\pi$  for sine-wave input. The mean value of the output is normally required, and a moving-coil meter is suitable as the deflection is proportional to the mean current.

\*To quote Tom Lehrer: Plagiarise Plagiarise Remember why the good Lord made your eyes So don't shade your eyes But Plagiarise Plagiarise Plagiarise —only please to call it Research When the resistive load is replaced by a capacitor, conduction of the diode only takes place for those instants when the input voltage exceeds the voltage stored on the capacitor. For a steady-state a.c. signal this corresponds to the positive peak of the input, and assuming no discharge of the capacitor in the intervening period the conduction angle is vanishingly small and is centred on  $\pi/_2$ . The resulting constant voltage across the capacitor is measurable with any d.c. voltmeter whose input current requirements are so small as to avoid significant capacitor discharge.

To accommodate varying signal amplitudes some discharge must be permitted since a small amplitude would otherwise never be sensed if following a larger input. The resistive path leads to a compromise time constant between maximum holding time of the peak voltage and minimum recovery time after large peaks. Conversely, the half-wave rectifier suffers from capacitive effects at high frequency with stray capacitance leading to partial peak rectification. The resulting output/frequency characteristic often shows a rise of I to 3dB prior to the cut-off frequency limits of the amplifier.

The sampling circuit replaces the diode of the half-wave rectifier by a switch which closes for a brief interval at some phase angle determined by external circuits. The output is zero for all instants except the sampling instant. With capacitive loading, provided the switch closure is for a period of time greater than the time constant of the capacitance together with the amplifier output resistance, then the capacitor volt-

# Four types of circuit, listed here, to measure the amplitude of an a.c. waveform—see Figs. 1 to 4.

Circuit	Load	Conduction Conduction angles, $\phi_1, \phi_2$ device		Voltmeter
Half-wave rectifier	R	0, π	diode	mean/d.c. moving coi
Peak rectifier	C	$\frac{\pi}{2}, \frac{\pi}{2}$	diode	d.c.
Sample	R	arbitrary ⊿ø→0	switch	instantaneou
Sample and hold	C	arbitrary ∆ø→0	switch	d.c.



Figs. 1–4. Types of circuit used to measure amplitude of a.c. waveforms (see Table). Complete circuits are given in cards 7 and 8 in set 18.



Fig. 5. LM3900 c.d.a. is well-suited to measurement of time period and frequency. An input capacitor can alternatively be charged through a diode to form a "pump" circuit (see card 10).



Fig. 6. Defining operating conditions for testing a zener diode with a c.d.a. (see card 5).

age becomes equal to the input voltage (again a compromise since the sampling period should not be so long as to allow a significant change in the input). If the switch is closed periodically at the same instant in successive cycles then the sampling time may be reduced, with the capacitor voltage increasing to the required level over a number of periods. With the switch open, as it is for most of the time, the capacitor stores or holds the sampled voltage, provided the measuring instrument is suitably buffered.

The sampling circuits are readily constructed with current-differencing amplifiers, and long hold times are possible. With careful adjustment the output drift can be < 5%/hour under controlled conditions which is a good performance from such a general-purpose circuit. The accuracy is less impressive since the currentmirror match is involved, and it cannot compete with standard op-amp circuits in this respect.

### Measuring period and frequency

The measurement of time period and frequency is another field to which the circuit is well-suited. A pulse waveform of constant width and height but variable frequency is fed as in Fig. 5 to the amplifier with parallel RC feedback. The mean voltage across the capacitor is then directly proportional to the input frequency. Alternatively frequency and pulse height may be kept constant when the output becomes a measure of pulse width. The availability of two inputs extends this capability to the measurement of frequency difference or sum. Alternatively an input capacitor may be charged and discharged through a diode network to give the equivalent of a diode pump/transistor pump type of frequency meter (tachometer).

The d.c. characteristics of the amplifier can be used to simultaneously define the operating conditions of diodes, zeners etc, while providing a low output impedance point for ease of measurement (Fig. 6). Finally, the circuit may be used in conjunction with an external network of resistors and diodes to perform quite complex logic functions such as exclusive-OR. Though offering no competition for the usual logic families for large-scale applications, they are very convenient for providing a small number of logic functions in an existing system. The wide range of supply voltages particularly commend them for such applications.

## Measurement and detection

## **Circuit description**

These circuits operate largely in a switched or non-linear mode; one card deals with negative resistance circuits which can be considered as having controlled non-linearity.

The amplifier gain is large enough that simple logic functions are readily performed as described in set 16 (Fig. 1). By scaling the resistors the output of Fig. 2 becomes  $\overline{A} + B$ and the smoothed output has a d.c. value dependent on the phase difference of the inputs if these are square waves. The phase detector combined with a v.c.o. (set 17) gives a p.p.1., see card 3. Scaled resistors can also be used to provide simple forms of threshold logic including the weighted comparator of Fig. 3, while the comparator, using equal-valued resistors if desired, forms part

**Circuit description** 

The integrator plays a large

as in waveform generators.

part in measurement circuits

Sample-and-hold circuits are

closely related to peak-detectors,

in that a voltage is stored on a

capacitor that is proportional

to the input voltage at some

defined instant (the sampling

voltage respectively). The ramp

and hold circuit of Fig. 5 is an

intermediate step between the

two circuits. If a resistive path

capacitor then the mean voltage

is placed in parallel with the

developed at the output is

defined by the mean input.

Access to both inverting and

non-inverting inputs makes it

possible to measure the mean

value of the sum or difference of two inputs, with the

capacitor minimizing the output ripple. The circuit of Fig. 6 formed part of a controlledamplitude RC oscillator (set 17) detecting the mean value of a half-wave rectified sine-wave; that of Fig. 7 acts as a

instant and the input peak



of the peak detector of Fig. 4. The capacitor charges through the diode until it matches the peak value of the input. Another application for the comparator is in coupling signals into and out of transducers. Positive feedback may be added to the system to provide hysteresis, or negative feedback to control the level of current/voltage in a load as when testing or measuring the properties of a device e.g. semiconductor breakdown voltages.



Further reading Frederiksen, T. M., Howard, W. M. Sleeth, R. S., The LM3900-A New Current-Differencing Quad of  $\pm$ Input Amplifiers, National Semiconductor application note AN72. Frederiksen, T. M., Norton quad amplifier subtracts from costs, adds to design options, Electronics, Dec. 6, 1973, pp.116-20. Motorola Linear Integrated Circuit Data Book, pp.7-446, 7-453, 7-456 and 7-463; data sheets on MC3301P and MC3401P amplifiers. National Semiconductor, Linear Integrated Circuits, pp.226-33, data sheets on LM3900. Frederiksen, T. M., Howard, W. M., Sleeth, R. S., Use Current-mode IC amplifiers, Electronic Design, vol. 21, no. 2, Jan. 18, 1973, pp.48-55. Mortensen, H., Use a quad amplifier to handle transducer bridge signals, *Electronic* Design, vol. 21, no. 3, Feb. 1, 1973, pp.74-6.

tachometer provided the inputs are of controlled amplitude and pulse width. The non-inverting input has a non-linear v/i characteristic. Overdriven with two high-frequency signals it injects a current into the amplifier which contains sum and difference frequencies. If  $f_1 - f_2$  is a low frequency the filtering action of the RC network removes the highfrequency components while giving an output at  $f_1 - f_2$ . This input stage is outside the feedback loop and is not frequency limited by the amplifier compensation. Low-level high-frequency amplification is possible using this input.

## Logic circuits



#### **Circuit description**

Just as t.t.l., c.m.o.s. gates may be interconnected to produce more complex functions, so the simple gate functions described on card 1 may be used in identical configurations. The standard methods for producing such functions as the exclusive-OR gate which has the output high for one and only one high input require four two-input NAND gates. This may be acceptable where it is desired to implement a system with a single device type even though multiple exclusive-OR circuits are available within a single package.

The flexibility of the currentdifferencing configuration allows such circuits to be designed economically while retaining the advantages of wide supply voltage range and good noise immunityprovided again that the circuit is restricted to industrial applications not requiring fast response. The lower amplifier is a standard AND gate as discussed in set 16, card 4, while the upper amplifier is an OR gate. By feeding back the AND gate output to the inverting input of the OR gate the logic 1 resulting from A=1 B=1 is inhibited by the logic 1 simultaneously appearing at the AND gate output. The resulting truth-table shows the output to be an exclusive-OR gate, but the simultaneous availability of the AND function creates a half-adder. It is obvious that such a circuit is no competitor to c.m.o.s. or t.t.l. for complex logic functions in view of the speed limitation. Where a small number of logic

Typical performance IC:  $\frac{1}{2}LM3900$ +V: +20V R<sub>1</sub>:  $82k\Omega$ R<sub>2</sub>:  $120k\Omega$ R<sub>3</sub>:  $33k\Omega$ Output logic 0:  $\approx 150$ mV Output logic 1:  $\approx 19.2$ V (with load current 5mA output logic 1 still within 1V of positive supply)

functions are to be introduced into an industrial system the availability of a circuit that will operate off virtually any available supply is a clear advantage. Being able to mix functions within a given package allows analogue and digital processes to be tackled efficiently.

#### **Component changes**

+V: +4 to +36V. In practice several samples of the device operated satisfactorily for supply voltages down to 2.7V at room temperatures. R<sub>1</sub>, R<sub>2</sub>: Values may range from < 10k to > 1M $\Omega$ . There is no obvious advantage to lowering the resistances, but at high values capacitive strays may produce input current spikes significantly larger than the quiescent currents. For the AND gate, the current in  $R_1$ must be less than current in R<sub>2</sub> while with both inputs high the current through the pair of  $R_1$ must exceed that in R<sub>2</sub>. For a logic 1 signal nearly equal to +V this gives  $R_2/2 < R_1 < R_2$ . R<sub>3</sub>: This has to provide enough current to override the OR gate action for A=1 B=1. Value is not critical and values from  $R_1/2$  down to  $R_2/2$  are satisfactory. Where the input is noisy, the input resistors may be centre tapped and the tapping point decoupled to ground. The high value of resistors used together with the wide noise margins possible at high supply voltages allows the circuit to accommodate high noise environments provided the supply is adequately decoupled.

### **Circuit modifications**

• Interchanging inverting and non-inverting inputs on the OR gate converts it into a NOR gate (Fig. 1). Then the output in the absence of feedback through R<sub>3</sub> would be logic 1 only when both inputs are at logic 0. The feedback forces the output of the NOR gate to go to logic 1 when both inputs are at logic 1, overriding the normal NOR gate function. The net result is that the output of the NOR gate becomes logic 1 for A = B and the circuit is an equality comparator. The output can also be seen as the negation of the exclusive-OR function.

 It is not necessary to use two amplifiers to perform the exclusive-OR function. Instead, a passive AND gate using  $D_1$ ,  $D_2$  and  $R_1$  keeps  $D_3$  out of conduction except when A=B=1 (Fig. 2). Under these circumstances the current in R1 is diverted through  $D_3$  into the amplifier inverting input and overrides the OR gate action to provide a logic 0 at the amplifier output. A limitation in this circuit is the low threshold to the AND gate, with input levels rising above about 0.4V bringing D<sub>3</sub> into conduction i.e. 0-level noise margin is small.

• If the inputs of the original circuit are linked the result

becomes a simple and readilycontrolled window comparator. Replacing the paralleled resistors by single resistors the simplified form of Fig. 3 is obtained. For low input voltages both amplifier outputs are low. At some intermediate value, the lower threshold voltage  $V_L$ , output X goes high assuming  $R_2/R_1 < R_3/R_4$ (Fig. 4). Output Y remains low. At the upper threshold output Y goes high driving X low. At some still higher input voltage a suitable choice of R<sub>2</sub>, R<sub>5</sub> also allows X to go high again if required as the current in R<sub>2</sub> continues to increase while that in  $R_5$  is restricted to  $\langle V/R_5$ .

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## **Phase-locked loop**



### **Circuit description**

There are three identifiable subsections to this phase-locked loop. The first is a phasedetector which receives one input from the signal source and one from the square-wave output of the voltage-controlled oscillator. The output of the phase detector, when the oscillator is locked to the signal source, is a negativegoing pulse train of the same frequency but of mark-space ratio dependent on the phase difference between the two waveforms (see circuit modifications). Smoothed by a simple RC filter, a direct voltage results that controls the rate of charging of a capacitor in the second subsection, the integrator. When the output voltage from the integrator reaches a critical level it causes a change in level from the output of the third sub-section, a Schmitt-trigger circuit. This switches the transistor reversing the net current into the integrator and with it the direction of charging until the second trigger level is reached. This completes the cycle of the v.c.o., and if its frequency is close to that of the incoming signal, locking is achieved. If the incoming signal changes its frequency slowly, the resulting phase difference produces a corresponding change in the direct voltage that controls the v.c.o. i.e. the frequency changes, remaining in lock, but with a new value of phase difference. If the frequency difference between the input and the v.c.o. becomes too great or if either varies too rapidly, \_ocking can be lost. The lock-in range with this circuit is over a frequency

range of about 2:1 while the capture range is somewhat smaller. In summary, frequency modulation of the signal source, if slow enough, produces pulse-duration modulation at the phase detector output. After filtering, which converts this p.d.m. into a varying direct voltage, the v.c.o. is frequency modulated causing it to track the original frequency variations of the signal.

## **Component changes**

Schmitt:  $R_6$ ,  $R_7$ ,  $R_8$ . Design ranges are indicated on set 17 card 6. One change would be to design for smaller hysteresis. Increasing  $R_7$  reduces the positive feedback to achieve this.

*Integrator:* To operate at low frequencies  $R_4$ ,  $R_5$  may be increased. This disturbs the frequency stability of the v.c.o. if used in a free-running mode, but closure of the feedback loop via the phase detector accommodates this at the expense of drift in the phase difference.

*Filter*: If the signal is expected to change frequency only slowly if at all, the filter time-constant can be increased with advantage.

### **Circuit modifications**

• The phase-detector action is as follows: when both inputs are low the output is high. For equal amplitude inputs, and  $R_1 > R_2$ , then when both inputs are high the output is high since the current to the . . non-inverting input is greater than that to the inverting input. When only one is high the output is controlled by that input.



• Because the phasecomparator has an output with a minimum mark-space ratio of 1:1, the mean output voltage swings over the range + V/2 to + V allowing a maximum frequency range for locking of about 2:1 assuming a linear v.c.o. By introducing an amplifier this range can be increased to any desired extent since the amplifier may be biased to give an output  $\rightarrow 0$ i.e. the v.c.o. frequency may be driven to a very low value while lock is retained (Fig. 2). Phase difference between the received and generated signals is frequency-dependent since at any given frequency there is a specific direct voltage required by the v.c.o. and hence a specific phase difference at which the phase comparator can deliver that voltage. If the filter is replaced by an integrator with a pre-set bias, then provided the integrator has a high gain (approaches the ideal integrator), its mean output takes up a value



appropriate to the incoming frequency, while the phase difference can be controlled by varying R (35k to 70k $\Omega$  to provide phase differences from 0 to  $\pi$  with values shown in Fig. 3). This is because the net current into the integrator  $\rightarrow 0$ when the output is to remain constant.



## Set 18: c.d.as-measurement, detection-4

## **Transducer driving**



Typical performance +V: +10V  $R_1$ ,  $R_2$ : 220k $\Omega$  $R_3$ ,  $R_5$ : 5.6M $\Omega$  $R_4$ : 10M $\Omega$ Hysteresis: 400mV

#### **Circuit description**

Where the input signal is from a transducer which has a floating and preferably low resistance output, the circuit shown gives a convenient means of setting the trip points symmetrically about the zerovoltage state of the transducer. With the output at zero and  $R_1 = R_2, R_3 = R_5 = R_4/2$  then with the transducer output at zero, the net current into the amplifier is  $V(1/R_4 - 1/R_3)$ . With the output at +V the net input current is  $V(1/R_4 + 1R_5 1/R_3$ ). With the above conditions these currents reduce to  $\pm (V/R_4)$  respectively. The input sensitivity/impedance depend on R1, R2 increasing the former reducing the latter. Sensing of signals  $\ll 100 \text{mV}$  is not recommended but the circuit has reasonable rejection of high frequency noise/strays because of the relatively large input resistances used together with the effective input capacitance (said to be around 2nF by the manufacturer).



#### **Typical performance**

Inputs can be any pair of matched linear or non-linear transducers, which deliver a variable current for a fixed p.d. as the physical parameter under test varies e.g. reverse-biased silicon photodiodes giving current proportional to light intensity.

## **Circuit description**

Current-differencing characteristics of the amplifier make it particularly suitable as a comparator of input currents from whatever source they may come. Accuracy of matching of the input stages is not such as to allow very high precision but in routine applications it can be used for rapid detection of out-of-tolerance resistors excessive leakage currents in diodes etc. One of the inputs is fed from a standard element or with a current defined by an external resistor while the device under test is connected between the other input and the positive supply line. Transducers such as photodiodes and transistors, photoconductive cells operated at low light intensities, and thermistors, can be used in matched pairs to sense the difference between physical parameters at two different locations while offering compensation against changes in any common variable (temperature effects on photoconductors etc.).



 $\begin{array}{l} \textbf{Typical performance} \\ + V: 12V \\ R_L: 600 \\ \Omega \\ R: 100 \\ k \\ \nu_i: > + 10V \end{array}$ 

### **Circuit description**

The normal sink current is provided by a constant current stage of about 1.3mA. An additional internal transistor when over-driven can provide an output sink current of up to 30mA with a reasonable saturation voltage (>0.5V). This output current depends on the combined current gains of two internal transistors, allowing input currents as low as  $100\mu$ A to provide sufficient drive. Output current is sufficient to drive small filament lamps and relays. As usual to protect the amplifier against reverse voltage spikes on switch-off a parallel diode is added. Driving of l.e.ds requires precautions since the device p.d. is  $\ll + V$  leaving a large amplifier dissipation at high supply voltages. Either reduce supply voltage or use resistance in series with l.e.d. to limit current.



Typical performance + V: 10VR<sub>1</sub>:  $10k\Omega$ R<sub>2</sub>, R<sub>3</sub>:  $470\Omega$ R<sub>4</sub>:  $100k\Omega$ Tr<sub>1</sub>: BFR81 R<sub>L</sub>:  $40\Omega$ 

## Circuit description

A single transistor with a current gain in saturation of > 10 increases the load drive capability to > 250mA. Resistor  $R_3$  is included to minimize the p.d. across the amplifier output transistor when switched to the low state. Feedback via  $R_4$  to the inverting input provides positive feedback for sharply

defined switching (because of additional inversion by the transistor). Output swing can be to within a few hundred millivolts of the supply with a suitable power transistor and this allows for a well-defined hysteresis. The value of  $R_1$  is reduced to allow sufficient input current to the amplifier to hold the required output state as the switching point is approached. If hysteresis is not required,  $R_4$  is eliminated and  $\mathbf{R}_1$  reverts to the value used in the previous circuit i.e. that value that results in an input current of about 0.1mA at the given size of input pulse.

#### **Cross references**

Set 9, cards 1, 11. Set 13, cards 2, 4, 8. Set 16, cards 2, 3, 5, 6. Set 17, card 6.

## Semiconductor device testing



### **Circuit description**

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One application to which the matching characteristics of the amplifiers can be applied is the testing of semiconductor devices. To find the d.c. characteristics of a zener diode for example, the current should be defined (or measured) and the terminal p.d. measured over a wide range of currents. If the currents are low then this makes the measurement of voltage more difficult. In the circuit shown the direct current in the diode is given by  $V_{be}/R$ where  $V_{be}$  is that of the amplifier input transistor. At very low currents the diode carries the 30-nA amplifier input current as well. If the amplifier output voltage to ground is measured it gives the zener voltage after subtraction of the V<sub>be</sub> term. Output impedance is low even at low zener currents. simplifying the voltage measurement. For these amplifiers the V<sub>be</sub> term is little dependent on the feedback current (though somewhat more so on the output voltage). As the zener voltage is substantially constant, the V<sub>be</sub> voltage can be conveniently measured when operating at a relatively high current yet will be valid for the lower currents. It is not necessary to make these two separate measurements since there are other amplifiers in the package with almost identical Vbe values. One of these amplifiers is biased as shown and the voltage measurement taken between the two outputs. These are both low impedance points allowing a wide choice of voltmeters without significant loading. Matching between the amplifier V<sub>be</sub> values is not perfect and improvement follows from operating the amplifiers at equal currents i.e. by placing

equal resistance values to ground. By superposing an alternating voltage of peak-peak value  $< V_{be}$  at the input, as from a generator whose internal d.c. resistance  $\geq R$ , an a.c. current is defined in the device under test. The alternating voltage at the output then gives a measure of the dynamic or slope resistance of the device.

### **Component variations**

+V: Must be >( $V_z$ +2V) to allow for amplifier Vbe plus output saturation voltage. Not critical within these limits. Rejection of supply changes improved when using compensating amplifier (to as little as  $1\mu V/V$  effect on V<sub>be</sub> differentials). R: 50 $\Omega$  to 10M $\Omega$ . The range corresponds to maximum output current of about 10mA down to minimum current of 50nA. At this lower level accuracy is poor because of amplifier input current of 30nA. If  $v_i$ , the input alternating voltage, is kept constant as R is varied, then since V<sub>be1</sub> is substantially constant, the alternating current superposed on the d.c. remains a constant fraction. With zener diodes operated in their normal region the zener voltage varies little, any variation in slope resistance shows clearly in the variation of the output a.c. component.  $V_i$ : Peak-peak voltage  $< V_{be_1}$ . Typical values of 0.1 to  $0.3 V_{be1}$ .

## **Circuit modifications**

• To define the current accurately it is not enough to rely on the constancy of the input  $V_{be}$  particularly if the temperature is subject to variation. A simple alternative is to take the resistor to the negative supply rail (if available) through a diode to give compensation against the

### Typical performance

IC:  $\frac{1}{2}LM3900$ +V: +15V R: 5.6k $\Omega$ Zener current:  $\approx 100\mu A$ Amplifier input V<sub>bc</sub> unbalance,  $\Delta V \approx 2mV$  for V<sub>z</sub> of 6V N.B.  $\Delta V$  varies with both V<sub>0</sub> and I<sub>0</sub> but V<sub>0</sub> in turn is approximately constant for a given zener. Zener d.c.  $\approx V_{be1}/R$ Zener a.c.  $\approx v_i/R$ 



remaining small effect of V<sub>be</sub> variations (Fig. 1).

• Fig. 2 circuit varies the d.c. resistance to ground from the amplifier input terminal in 10:1 steps. The equivalent alternating voltage generated in series with this resistance is about 110mV pk-pk or about 20% of the V<sub>be</sub> value. Current range as shown is 100nA to 10mA.

• A p-n-p transistor may have its  $V_{be}/I_e$  characteristic plotted using a similar arrangement. Collector current is defined with the emitter taken to the output of one amplifier, the base to the other.  $R_1$ ,  $C_1$  control the h.f. characteristics to maintain stability. By varying  $R_3$ , the  $V_{ce}$  value for the transistor can be controlled to show the small effect of  $V_{ce}$  on the  $I_c/V_{be}$ characteristics; n-p-n transistors cannot be dealt with so simply using this form of circuit.



## Set 18: c.d.as-measurement, detection-6

## Negative resistance



#### Circuit description

A potent tool in the design of oscillators and certain classes of active filters is the negativeresistance circuit, which may also be a particular case of a negative-impedance converter. A previously described constant-current circuit may be adapted to give a controlled negative output resistance, by allowing the overall positive feedback to exceed the negative feedback. The circuit remains stable only when the external load re-establishes the supremacy of the negative feedback. In the circuit shown, a short-circuit load eliminates the positive feedback altogether indicating that the circuit is short-circuit stable. For  $R_2/R_1 > R/R_3$ , the open-circuit load condition has excess positive feedback and switching of the output would occur. For intermediate values of  $R_L$ , the load current as defined by the voltage V and potentiometer setting k is observed to fall as the load-resistance is reduced. For the values shown,  $R_2 = 10 k\Omega$  gives the limiting case of output resistance  $\rightarrow \infty$ , and increasing values of R<sub>2</sub> produce a negative resistance of falling magnitude. Maximum current in the load is restricted to about 2mA with k=0.5 but the amplifier itself restricts the current to around 10mA regardless of the resistance values chosen. In principle, emitter followers can be added at the amplifier output with all the resistor values lowered to increase the drive capability, but for most applications it is the small-signal performance that is of interest.

IC:  $\frac{1}{4}$ LM3900 k: 0.5 R<sub>1</sub>, R<sub>3</sub>, R: 10k $\Omega$ R<sub>2</sub>: 15k $\Omega$ Dynamic output resistance:  $-9.3k\Omega$ V<sub>s</sub>: +15V V: 5.3V (set for output current of 1mA with R<sub>2</sub> trimmed for output resistance  $\rightarrow \infty$ )

Typical performance

## **Component changes**

 $R_1, R_3, R$ : It is often convenient to have these equal. Minimum value is dictated by loading effects on output and would not generally be much below  $10k\Omega$ . Negative slope resistance available is comparable to these values. R<sub>2</sub>: Controls the slope of the resistance, together with the tap on R. Range 10 to  $20k\Omega$ . k: 0.2 to 0.8. For application suggested later, where only small negative resistance effect required k=0.5 is satisfactory. V+:+4 to +36V. Normal operating range for amplifier. V: Used to set the output current to some range over which R<sub>2</sub> controls the negative slope +1V to +8V for  $V_{+} = 15V.$ 

## **Circuit modifications**

A typical application is to introduce a negative slope resistance into a circuit normally used as a constantcurrent source. This may be either to overcome some positive resistance due to leakage paths etc, or simply to modify the waveform produced in a generator. Here the circuit provides the charging current to a capacitor that is part of an astable circuit having some similarity to a unijunction oscillator (Fig. 1). Capacitor imperfections plus external resistors would result in a ramp whose dV/dt fell at higher amplitudes. By increasing the negative slope, more current is fed into the capacitor as the p.d. across it increases. This can linearize the slope or with overcompensation cause the slope of the waveform to increase at higher voltages.



• For other applications a direct equivalent of the negative-impedance converter is required. In Fig. 2 the voltage gain of the amplifier is +2. Hence the voltage appearing across the resistor R' is equal to the input voltage in magnitude but in a sense that returns a proportional current to the source. This leads to an input resistance of -R' (shunted by +R where R can be up to  $10M\Omega$  is required representing a small error).

• The previous circuit is short-circuit stable in that a low source resistance ensure that the positive feedback is bypassed, leaving the negative feedback in control. With the alternative configuration in Fig. 3, the negative feedback overrides the positive feedback provided the source impedance is greater than R'. Additional bias resistors from the amplifier inputs to the positive supply line may be used to set desired output quiescent levels

Cross references Set 3, card 8. Set 6, card 6 (AC constantcurrent circuits, see note on front). Set 17, card 5.



## Set 18: c.d.as—measurement, detection—



 Typical performance

 +V: +15V 

  $Tr_1: BC125$ 
 $R_1: 1M\Omega$ 
 $R_2: 1.2M\Omega$ 
 $R_3: 10k\Omega$ 
 $R_4: 6.8k\Omega$ 
 $C: 10\mu F$ 
 $D_1: 1N914$ 
 $v_1: 2.5V r.m.s., 1kHz$ 
 $v_0: 1.35V d.c.$ 



## **Circuit description**

The mean value of an alternating voltage can be obtained by rectifying the signal (half or full-wave) and applying the resulting output to a mean-sensing instrument such as a moving-coil meter. Alternatively the rectified waveform may be filtered and applied to a d.c. voltmeter (including a d.v.m.). In each case one or more diodes or transistors are required to perform the rectifying function, and they introduce two related errors into the response of the circuit. Non-linearity in the diode causes a corresponding non-linearity in the transfer function. In particular, at very low levels the signal is insufficient to bring the diode into conduction. This gives an offset to the transfer function. If other diode effects can be added, then the offset together with its temperature drift can be largely cancelled. Rectification of signals below about 0.5V r.m.s. is still impossible but above this level the error can be small. In the circuit shown the amplifier integrates the difference between the currents caused by the rectified input signal and the mean voltage at the transistor collector. For a particular value of  $R_2/R_1$  the diode and Vbe drops provide partial cancellation, and the transistor collector potential becomes a linear function of the mean half-wave rectified input. (Although the slope if extended would pass through the origin, rectification of low-level signals cannot be achieved with this circuit.) The value of R<sub>2</sub> obtained empirically for this circuit differs somewhat from that previously proposed for the related section of the

ģ.

Wien bridge oscillator. The values of R4, R3 can be adjusted over a wide range to accommodate differing load resistors/meters etc while the capacitor can be increased to give adequate smoothing for frequencies <1Hz. At high frequencies partial peak rectification may occur in the input section because of stray capacitance while shunt capacitance across R1 will increase the input current. A slowly rising response is observed up to 1MHz.

## **Component changes**

+V: Not critical, provided it is >  $(V_0 + 1V)$ . For higher output currents R<sub>3</sub> would become too low if supply/ output differential is too small. Tr<sub>1</sub>: Any silicon n-p-n smallsignal transistor. No significant h.f. response required because of integrating action of C on rectified waveform. R<sub>1</sub> sets input impedance and should be high to minimize loading on source (important because of non-linear input impedance). If too high then capacitive strays become more significant. 100k to  $10M\Omega$ .  $R_{2}$ : Ratio to  $R_{1}$  sets optimum slope to minimize offset effect. Empirical result suggests  $R_2/R_1 \approx 1.25$ , manufacturers suggest  $R_2/R_1 \approx 1.6$  for similar circuit.

 $R_s$  has to carry maximum meter current at maximum output voltage i.e. should be <10k $\Omega$  for 1mA meter movement at 5V full-scale and 15V supply.

R<sub>4</sub>: Not critical, avoids excessive non-linear loading on amplifier output. 3.3k to  $33k\Omega$ . C sets degree of ripple reduction at low frequencies. C>22 $\mu$ F is necessary for f<1Hz if needle flutter is to be reasonable. C may be  $< 1\mu$ F for higher frequencies (self-inductance of electrolytics may cause problems at h.f.) D<sub>1</sub>: Small-signal silicon diode.

#### **Circuit modifications**

• A very simple alternative circuit is that used for the basic tachometer (see also set 17 card 2). In this (Fig. 1) the input transistor of the amplifier non-inverting input provides the rectifying action directly. Reverse half cycles inject current into the damping circuit which may cause disturbance to the other amplifiers in the package. A low voltage-drop diode (germanium, Schottky) may be used to absorb this reverse current but leakage effects with germanium would force the use of lower resistance values. The mean current in the two inputs will become comparable because of the negative feedback while the capacitor removes the ripple as before. A peak detector with gain • is shown in Fig. 2, in which the capacitor is charged until the



current in R<sub>3</sub> equals the sum of the currents in R<sub>2</sub> and R<sub>1</sub> taken at the positive peak of the input voltage. At all other points in the cycle, the diode is non-conducting and some loss of charge takes place through R<sub>3</sub>. Resistor R<sub>2</sub> provides a small forward bias. Resistor R<sub>3</sub> cannot be increased greatly without introducing a large error term due to the amplifier input current. The diode is within the feedback loop and the error term due to the diode drop  $\rightarrow 0$  at low frequencies. Amplifier slew-rate limiting prevents the circuit from approaching the frequency response of the previous circuits.

### **Cross references**

Set 4, cards 2, 3, 5, 9, 10. Set 9, card 12. Set 17, cards 2, 3.

## Sample and hold circuits



Circuit description

If an ideal integrator is incorporated into an overall d.c. feedback loop, then the capacitor charges until it provides an output voltage equal to (or some set multiple of) the input voltage. It introduces a time delay into the loop by virtue of the maximum charging current that can be provided by the amplifiers and so the speed of response of the amplifier is limited. If the loop is interrupted at any instant such that the integrator input current is reduced to zero, the p.d. across the capacitor remains substantially constant at the value corresponding to the input just prior to that interruption i.e. the signal can be continuously or periodically sampled with the facility to hold the value of the input at any required instant. Amplifier input currents, capacitor leakage etc cause drift in this held value but with care drifts of < 5% are possible for periods of 5 to 30 minutes with simple circuitry. With the control input low (sample mode) the integrator receives current to its inputs via R<sub>3</sub> and  $R_6/D_1$ . It integrates the difference between those currents until the currents in  $R_1$ ,  $R_2$  are substantially equal, bringing v' into its linear region. Any increase in vi raises v' and hence the current into the integrator non-inverting input until vo tracks the charge in vi. Allowing for an ideal current mirror at the amplifier input  $v_0/v_1 = R_2/R_1$  though V<sub>be</sub> effects only cancel well for  $R_2 \approx R_1$ . In the hold mode currents in R<sub>3</sub>, R<sub>6</sub> are grounded and the only currents affecting

v<sub>o</sub> are the amplifier input current plus capacitor leakage.

#### Component changes

R<sub>1</sub>: Sets input resistance. If too high then amplifier input current becomes significant. 100k $\Omega$  to 10M $\Omega$ . R<sub>2</sub>: R<sub>2</sub>=R<sub>1</sub> gives best compensation against V<sub>be</sub> drifts, but unequal sensitivities at two inputs can be compensated for by trimming the ratio for v<sub>0</sub>=v<sub>1</sub> at maximum output.

R<sub>a</sub>: Set for sufficient current at v' maximum to override current in R<sub>6</sub>. Typically  $R_3 \approx$ 0.5 to 0.8  $R_6$ . Maximum current should not exceed, say 5mA; too low a current restricts maximum rate of charge.  $R_4$ ,  $R_5$ : In conjunction with hold/sample pulse must be sufficient to saturate Tr<sub>1</sub>, Tr<sub>2</sub>. D<sub>1</sub>: Low reverse leakage diode for least error in hold mode.  $C_1$ : High value increases time-constant for minimum drift in hold mode. Highleakage capacitors unsuitable, since leakage should be ≤ 30nA if capacitor is not to be limiting feature of circuit. Too high capacitance slows down charging rate forcing longer sample period.

## **Circuit modifications**

• The four amplifiers in the package are closely matched, and the input current of one can be used to generate an output voltage that drives a comparable current into the other. Trimming the  $2M\Omega$  resistor (Fig. 1) compensates for input-current induced drift while over-compensation can be used to minimize the additional leakage in C<sub>1</sub>, D<sub>1</sub>. Compensation can be exact at

**Typical performance** + V: +15V R<sub>1</sub>, R<sub>2</sub>: 1M $\Omega$ R<sub>3</sub>: 22k $\Omega$ R<sub>4</sub>, R<sub>5</sub>: 47k $\Omega$ R<sub>6</sub>: 33k $\Omega$ C<sub>1</sub>: 0.1 $\mu$ F D<sub>1</sub>: 1N4148 Tr<sub>1</sub>, Tr<sub>2</sub>: BC125 Input range: +1 to +14V Output drift: +0.2V/s in hold mode

only one combination of  $C_1$ temperature output voltage etc, but with care the output can remain stable to within 1% for several minutes using low cost capacitors.

• By using only one of the switching transistors, Fig. 2, the sample mode remains unchanged, while in the hold mode only input voltages below the stored value are ignored i.e. the output voltage is updated each time the input exceeds its previous peak value, holding the new peak pending any further increase. It is a form of peak-reading voltmeter.

• Using the other switching transistor (Fig. 3), then with the pulse input low the transistor is non-conducting and the sample mode is attained. When the pulse input is high the amplifier output is driven to its zero state regardless of the input.

Cross references Set 4, card 7. Set 11, card 11. Set 18, card 7.



## Set 18: c.d.as-measurement, detection-9

gain set to 60 at 1kHz

-ATIVE TO 1kHz (dB)

REL

**NUAS** 

## **High-frequency circuits**



Typical performance + V: 10V R<sub>1</sub>: 1k $\Omega$ R<sub>2</sub>: 220k $\Omega$ R<sub>3</sub>: 100k $\Omega$ C<sub>1</sub>: 10 $\mu$ F C<sub>2</sub>: 1 $\mu$ F Voltage gain: 60 @ 100kHz Input impedance: 1.5k $\Omega$ Output impedance: 100 $\Omega$ 

## **Circuit description**

The gain-bandwidth of a feedback amplifier is a good measure of performance over a wide range of frequencies if the fall-off in gain is due to a single dominant lag as for these current-differencing amplifiers. If there is no external path to ground from the inverting input with such an amplifier while a conducting path is provided between that input and the output, there is effectively 100% feedback and the amplifier attains its maximum bandwidth. To inject a signal without reducing the effect of this feedback, the signal is fed directly into the non-inverting input. This is outside the feedback path with this amplifier but couples the signal in via a high impedance path internally (see circuit modifications). The

disadvantage is that the feedback exercises no control over the input impedance at the non-inverting input i.e. the input current depends on that impedance and hence on the bias level. For most operating conditions the practical gain is 20 to 40% below that predicted on the basis of  $R_3/R_1$  alone. Output swing available is small at high frequencies being limited by the positive slew-rate limitation of about  $0.5V/\mu s$ . Hence at 500kHz, the peak-peak output cannot exceed about 500mV without distortion becoming serious. There is a variation in gain with supply voltage largely due to the effect of bias current on the input impedance at the non-inverting input but this can be held to  $\pm 10\%$  for a 2:1 variation in supply voltage.

#### **Component variations**

+V: Normal supply range for amplifier. Higher voltages allow higher input bias current for given value of R<sub>3</sub>. Hence input resistance at noninverting input falls allowing gain to approach more nearly to  $R_3/R_1$ . R1: Sets input impedance; add on input resistance of noninverting input  $r_i \approx 25 k \Omega / I_{\text{bias}}$ , where  $I_{\text{bias}}$  is current in  $R_2$  in  $\mu$ A. Range 1k to 10k $\Omega$ .  $R_2$ : Sets bias as  $I_{bias} = + V/R_2$ and direct output voltage as  $+ VR_3/R_2$ . Range 100k to 1M $\Omega$ . R<sub>3</sub>: Sets gain in conjunction with  $R_1$ . For  $R_1 \rightarrow 0$  overall voltage gain  $\rightarrow 20 \times$  magnitude of supply voltage in volts. C<sub>1</sub>, C<sub>2</sub>: Chosen in conjunction with  $R_1$  and load resistance to define low-frequency cut-off. Typically 0.1 to  $10\mu$ F.

### **Circuit modifications**

• The non-inverting input is a current mirror with the input transistor behaving as an almost ideal p-n junction having transconductance g<sub>m</sub> of





frequency is available;  $(f_1 - f_2)$  is then measurable on audiofrequency equipment.

Cross references Set 12, cards 1, 3, 8, 9.



## Set 18: c.d.as—measurement, detection—10

## **Tachometers**



#### **Circuit description**

A tachometer using both inputs of the current-differencing amplifier can reduce the ripple in the output voltage for a given frequency. It has some similarity to the diode pump circuit, in which the two diodes are  $D_1$  and the equivalent input diode at the non-inverting input. When the input goes positive the amplifier input diode is driven into conduction and capacitor  $C_1$  charges through  $R_1$  until the p.d. across it is  $(V_i - V_{be})$ , assuming that the positive peak is held for a time appreciably greater than  $R_1C_1$ . The current mirror action at the amplifier input ensures that an equal current flows in the inverting input, causing  $C_2$  to charge with the output going positive. When the input returns to its most negative value, D1 comes into conduction and C<sub>1</sub> discharges. Current drawn through  $R_2/C_2$  is in the same sense in both cases i.e. the effects of charging and discharging  $C_1$  are additive at the output, doubling the effective frequency of the pulses. The mark/space ratio of the pulses is not important and need not even be constant provided only that the duration of each is sufficient to allow complete charge/discharge. Overall sensitivity is controlled by  $C_1$  and  $R_2$  with  $R_1$  serving

to limit the peak input current and  $C_2$  to smooth the output d.c. In each cycle  $C_1$  gains and loses charge of  $C_1v_i$ . This corresponds to a net current in  $R_2$  of about  $2fC_1v_i$  if f is the repetition frequency, a factor 2 coming from the fact that  $R_2$ has to carry current corresponding to the direct discharge of  $C_1$  via  $D_1$  and also

**Typical performance** +V: +5V v<sub>i</sub>: 5V pk-pk R<sub>1</sub>: 10k $\Omega$ R<sub>2</sub>: 1M $\Omega$ C<sub>1</sub>: 1nF C<sub>2</sub>: 0.1 $\mu$ F for f: 500Hz, v<sub>0</sub>: 2.78V

a current equal to the charging current because of the current mirror action as outlined above. Hence the mean output voltage is  $\approx 2fC_1v_1R_2$  ignoring V<sub>be</sub> effects. The time constant  $C_2R_2$ should be  $\gg$  period of waveform for low ripple.

## **Component changes**

R<sub>1</sub> limits peak current to < 1mA and preferably  $100\mu$ A or less for high temperature operation. If too high prevents completion of charge/discharge cycles at high frequencies  $C_1R_1 < 0.1f$ .

 $C_1$  sets sensitivity contributing two units of charge  $C_1V_1$  on each cycle to the output. 100pF to  $1\mu$ F.

R<sub>1</sub>: Average current flow in R<sub>1</sub> is equal to the average of the charge gained and lost by C<sub>1</sub> as above. Because amplifier input current is low a mean current in  $R_2$  of about  $1\mu A$  is still sufficient to give moderate accuracy i.e. R<sub>2</sub> can be as high as  $10M\Omega$  but better accuracy for values of  $1M\Omega$  or less.  $C_2$ : Time constant  $R_2C_2$ determines ripple remaining on output with  $R_2C_2 \gg 1/f$  for minimum ripple. C2 must be low leakage or leakage resistance will cause loss of accuracy. 0.01 to  $1\mu$ F. D1: Not critical. Any silicon diode.

### **Circuit modifications**

• Transfer function of circuit is linear in  $v_0/f$  but there is an output offset of  $V_{be}+(I_-)R_2$ , where  $I_-\approx 30$ nA and  $V_{be}\approx 0.55$ V. By using a second amplifier with an identical feedback resistance and taking the output signal as shown, the output is substantially zero for zero input signal i.e. the transfer-function remains



linear but passes through the transposed origin. The offset has good temperature compensation (Fig. 1).
Simpler tachometer circuits are possible with some relaxation in specifications. For input pulses of defined

width t<sub>1</sub> and current I, the mean current in the resistor R must be equal to that at the non-inverting input by current mirror action i.e.  $i_{\rm R} \approx f l t_1$  and the output voltage smoothed by C is  $v_0 \approx f I t_1 R$ . If the pulse width remains constant then  $v_0 \propto f$ . No current flows in the non-inverting input when the input signal is at zero (Fig. 2). The output may be made • to pass through zero by adding a diode at the output inside the feedback loop and with common-mode biasing. A load

resistance  $\ll 200 k\Omega$  should be present to ensure that the current in R<sub>f</sub> can flow to ground while developing a very small voltage (Fig. 3).

## Cross references Set 3, card 3. Set 13, card 6. Set 14, cards 6, 9, 10.


Set 18: c.d.as-measurement, detection Up-date

1. In any signal-processing circuit, the input may contain transients which might damage the amplifier or interfere with its operation by saturating it etc. The usual technique of adding a pair of back-to-back diodes across the input with a resistive limiter is equally



applicable to these amplifiers. It is shown preceding a Schmitt trigger, and clips the input to an approximate square-wave which is then applied to the Schmitt to produce a square-wave almost equal to the supply in amplitude.

2. Two other circuits help to augment the range of functions available to users of current-differencing amplifiers. In addition to active filters as usually understood there are amplifiers with defined frequency responses, i.e. having a number of low frequency and high frequency timeconstants often widely spaced. These can be realized using purely passive networks though loading of one by the other can be a problem. The circuit illustrates a combination of lead and lag networks, with  $C_1$ ,  $C_2$  reducing the gain at low frequencies and C<sub>3</sub> reducing it at high frequencies. Such circuits are useful in audio frequency instrumentation when a

response matching various CCIR, NAB, BS or DIN curves are required. In such applications information may be needed about the peak-peak value of the signal after such processing. An example is in the measurement of wow and flutter for a tape recorder, and the meter drive circuit shown is a simple solution.



Regardless of the quiescent output, the capacitors  $C_4$  and  $C_5$  charge until the sum of their p.ds equals the supply voltage. When the output swings positive, D<sub>2</sub> conducts charging C<sub>6</sub> positively; for negative-going swings D<sub>1</sub> conducts, pulling  $C_4$  in the negative direction. Provided the time constant of the meter circuit and the  $C_4$ ,  $C_5$ combination is long compared with the period of the lowest frequency component the ripple is small and the meter reads the peak-peak output voltage. The diodes should be low voltage-drop devices to minimize errors at low levels and can be replaced by germanium transistors with collector-base shorted.

3. Where many current differencing amplifiers are to be checked, the circuit shown can help. Each amplifier is connected to a l.e.d./transistor combination such that the transistor has to force a current into R (and hence the l.e.d.) sufficient to raise the inverting input to 0.5V, its normal value for linear operation. With  $R = 100\Omega$  the 5mA current in the l.e.d. brings it to the level of visibility. A gross failure with either drive the output high (l.e.d. bright) or low (l.e.d. off). A more

detailed check is to compare the potentials at the inverting inputs—these points have a low resistance to ground because of the  $100\Omega$  resistors, and the meter used need only



have a sensitivity capable of registering a p.d. of a few millivolts but its resistance is not critical. The matching between inputs on a given chip is quoted in the reference article as typically 2mV and between different amplifiers from a production batch as 20mV.

#### Reference

Tenny, R. Check Norton amplifiers quickly, *EDN*, 1974, March 5, p. 72.

### Set 19: Monostable circuits

Monostable circuits, often referred to as multivibrators, can take various forms. The term multivibrator seems to have been used for Abraham and Blochs astable two-triode circuit of 1918, and the related bistable Eccles-Jordan circuit of 1919. Flip-flop or monostable circuits were developed a few years later, for example the 1926 transitron of van der Pol. Many of the early circuits used two timing capacitors, being derived from the a.c.-coupled multivibrator, but all of the monostables described in this set of cards use one timing capacitor. The most well-known arrangement (card 1) can be thought of as derived from the crosscoupled bistable circuit, with one resistive coupling replaced by the timing capacitor. Recovery time is often too long with this straightforward circuit. Cards 1 & 2 show a complementary arrangement that has short recovery time  $(RC|\beta)$ , but a possible disadvantage is that both transistors conduct in the quiescent state. Other methods are shown in the article, e.g. use of a diverting diode to prevent saturation (Fig. 3), addition of an emitter follower as shown in Fig. 4, and a circuit in which the on and off transistors are transposed, Fig. 6 (see also card 10). Emitter-coupled circuits, card 6, have the advantage of isolating the output point from the remainder of the circuit.

As with the astable circuits of set 8 this collection includes t.t.l. circuits (card 4), op-amp circuits (card 3) and circuits using c.m.o.s. devices (cards 5 & 8). Long delay circuits use the D-type flip-flop or unijunction transistors (card 8) and improved duty cycle is obtained with the circuits of card 10. A voltage-variable delay is provided by the circuits of card 7 of around 7 to  $90\mu$ s and 40 to 200ms.

Basic discrete-component circuits 1 Complementary circuits 2 Op-amp/comparator circuits 3 Monostable using t.t.l. gates 4 Compensated c.m.o.s. circuits 5 Emitter-coupled circuits 6 Voltage-controlled monostables 7 Long-delay circuits using c.m.o.s. and unijunction devices 8 Dual monstable applications using 555 timer 9 High duty-cycle monostable 10

# The monostable . . . doth give us pause

Not quite the words of Hamlet, but delay is one outstanding property of the monostable circuit. De Bono, in The Mechanism of Mind, could be describing another function of the circuit when he writes, "a short-term memory is just a way of extending the influence of an event beyond the real time of its occurrence along the dimension of time", e.g. a monostable will accept a transition at its input and respond with an output pulse, but for a finite time. A more formal description of the monostable (sometimes called a one-shot) is a circuit having one stable state in which it remains until triggered by an external signal into a quasi-stable state, where it remains for a time determined by circuit parameters and subsequently returns to its stable state. This basic action allows the monostable to be used for a variety of purposes such as lengthening, delaying and regenerating pulses, sequential timing and delay applications and frequency division.

The nature of the monostable circuits can be widely different because they can be designed using n-p-n and p-n-p bipolar transistors (in cross-coupled emittercoupled and complementary modes), fieldeffect transistors, operational amplifiers, discrete and integrated-circuit logic gates, as well as purpose-designed monostable integrated circuits.

A very common type of monostable uses a cross-coupled configuration which can be thought of as being a modification of a symmetrical bistable where one resistive coupling is replaced by a capacitive coupling. Two possibilities are available; the timing capacitor can either be connected between the collector of a normally-off transistor to the base of a normally-on transistor, or between the collector of a normally-on transistor and the base of a normally-off transistor. Fig. 1 shows a circuit of the first type and Fig. 2 the associated current and voltage waveforms.

Transistor  $Tr_2$  is in a stable on state, until triggered, due to the base drive supplied through  $R_2$ . Transistor  $Tr_1$  is held in a stable off state as  $R_4$  is connected to  $Tr_2$ collector which is at the low saturation value of  $v_{ce2}$  and  $R_3$  is returned to the negative  $V_{BB}$  rail. When a positive-going trigger pulse is applied to  $Tr_1$  base via  $C_1$  this transistor turns on causing its collector voltage to fall to almost zero.

Because the charge on  $C_2$  cannot change instantaneously, this negative-going transition is passed to  $Tr_2$  base which switches off. The transistors remain in these states while the charge on  $C_2$  changes via  $R_2$  causing  $v_{be2}$  to rise exponentially towards  $+ V_{CC}$ . When  $v_{be2}$  passes through zero and rises positively to a value depending on the type of transistor used, which causes base current to flow in  $Tr_2$ , this transistor begins



Fig. 1. A common monostable can be thought of as a bistable circuit with one resistive coupling replaced by a capacitative coupling. Two can be connected as shown.

Fig. 2. Waveforms associated with circuit of Fig. 1.

Fig. 3. Switching speed of Fig. 1 circuit is improved by preventing saturation with diodes  $D_1$  and  $D_2$ . Clamping diode  $D_3$ reduces recovery time by connection to a supply less than  $V_{cc}$ .





to turn on. The resulting fall in  $v_{c_2}$  is coupled to  $Tr_1$  base via  $C_3$ ,  $Tr_1$  beginning to turn off; the regenerative feedback via  $C_2$  and  $C_3$  causes the circuit to return to its stable state of  $Tr_1$  off and  $Tr_2$  on, the capacitor  $C_2$  recharging through  $R_1$ .

The switching speed of this kind of circuit can be improved by using higher speed switching transistors in a non-saturating circuit. To prevent saturation, germanium diode  $D_1$  and silicon diode  $D_2$  can be added as shown in Fig. 3. When  $Tr_2$  begins to turn on, to return the circuit to its stable state,  $D_1$  will be reverse-biased until  $v_{C2}$  falls below  $(v_{be2} + v_{D2})$  causing the excessive base drive current, which would otherwise saturate  $Tr_2$ , to be diverted through  $D_1$ . Two series-connected germanium diodes can be used in place of the silicon diode  $D_2$ . The waveform at  $Tr_i$  collector can have a slow recovery time, especially when driving capacitive loads, and this can be reduced by the addition of the clamping diode  $D_{i}$ returned to a supply  $V_A < V_{CC}$ . The output voltage  $v_{CI}$  attempts to rise towards a higher value with  $D_i$  present but becomes clamped at  $(V_A + V_{D3})$  when  $D_3$  conducts.

Another method of reducing the recovery time is to include an emitter follower between  $R_1$  and  $C_2$  of Fig. 1, as shown in Fig. 4. As  $C_2$  is charged to almost the supply rail voltage, the emitter of  $Tr_3$  is normally close to  $+V_{CC}$ . The input trigger pulse switches the circuit to its quasi-stable state and as the charge on  $C_2$  changes, the emitter voltage of  $Tr_3$  rises above its base yoltage ( $v_{CI}$  on) cutting the transistor off. When  $Tr_2$  again begins to conduct, the circuit returns to its stable state with  $C_2$  being rapidly recharged by the emitter current of  $Tr_3$ .

The output from  $Tr_i$  collector can be made to more closely approach a rectangular pulse by the inclusion of an isolation diode  $D_4$  as shown in Fig. 5. When  $Tr_i$  is on the collector current flows through  $R_1$  and  $R_7$  in parallel and a faster recovery time is achieved by making  $R_7 \leq R_1$  so that when  $Tr_1$  switches off  $D_4$  is reverse biased and  $C_2$  recharges more rapidly, through  $R_7$ , than in the circuit shown in Fig. 1. Other methods of triggering this type of monostable include negative pulses to either  $Tr_1$ collector or  $Tr_2$  base or positive pulses to the base of another transistor having its collector and emitter respectively connected to  $Tr_1$  collector and emitter.

Another form of the cross-coupled monostable is shown in Fig. 6, the major difference compared with the foregoing circuits being that  $Tr_1$  is on and  $Tr_2$  is off in the stable state. This is achieved by correct choice of the potential-dividing chain and by  $D_2$  being forward-biased via  $R_4$ , holding the base-emitter junction of  $Tr_2$ , reverse-biased. A negative-going input trigger pulse causes  $Tr_i$  to switch off and hence  $Tr_2$  to switch on and to remain in that state as  $C_2$  charges, part of the charging current being base drive to  $Tr_2$ . Diode  $D_2$  is reverse-biased in this quasi-stable state which ends when the base drive to  $Tr_2$ has fallen to a level which will not maintain conduction. Transistor Tr, then switches off causing  $Tr_i$  to return to the stable on state.



Fig. 4. Recovery time can also be reduced with an emitter follower between  $R_1$  and  $C_2$  of Fig. 1.



Fig. 5. Output from  $Tr_1$  is made more rectangular by isolation diode  $D_4$  and by making  $R_7 < R_1$ .



Fig. 6. In this variant of the cross-coupled monostable,  $Tr_1$  is normally on and  $Tr_2$  normally off. Note trigger is of opposite polarity. Circuit has much faster recovery time.

No output is taken from  $Tr_1$  collector as a load at that point significantly changes the off time of  $Tr_2$ . This circuit has a much faster recovery time than the previous ones discussed.

A cross-coupled monostable which besides producing a time-constant-dependent output pulse may provide one due to the input pulse duration is shown in Fig. 7. In the stable state  $Tr_1$  is on and  $Tr_2$  is off, so when a short-duration trigger pulse is applied via  $D_1$  the circuit remains in its quasi-stable state for a time determined largely by  $C_1R_3$ . However, when a long input pulse is applied,  $Tr_1$  will remain off until the input is removed even if  $C_1$  completes its discharge during that interval.

Fig. 8 shows an emitter-coupled monostable where  $Tr_2$  is on and  $Tr_1$  is off in the stable state. Compared with the cross-

coupled circuits, this type has the advantages of only using a single supply and providing an output which is taken from a point having no internal coupling. When a negative-going trigger pulse is applied to  $Tr_1$  collector via  $C_1$  it is coupled to the base of  $Tr_2$  which switches off. The emitter voltage falls allowing  $Tr_1$  to switch on for a time determined by that required for  $C_2$  to discharge sufficiently to allow  $Tr_2$ to begin to conduct. The emitter voltage then rises, causing  $Tr_1$  to begin to switch off, and the resulting rise in  $Tr_1$  collector voltage is coupled to  $Tr_2$  base which switches on, and  $Tr_1$  switches off to regain the stable state. Due to the presence of  $R_4$ , the output voltage swing does not approach  $V_{CC}$  and the recovery time is not very fast, as  $R_3$  should be greater than  $R_{\delta}$  to ensure the correct switching action. 140



Fig. 7. This circuit produces an output pulse dependent on input pulse duration.







Complementary transistors enable both transistors to be normally off (Fig. 9, above) or normally on (Fig. 10, right). Opposite-polarity outputs are available from Fig. 10.

**Re**covery time can be improved by the addition of an emitter follower as was done in Fig. 4.

A monostable using a complementary pair of transistors, having both transistors off in the stable state, is shown in Fig. 9. A negative-going trigger pulse applied to  $Tr_i$  base causes this p-n-p transistor to switch on, its collector current in  $R_1$  and  $R_2$  causing the base of  $Tr_2$  to go positive causing this n-p-n transistor to switch on also. The resulting collector current in  $R_4$  causes the output voltage to go negative and this change is coupled to  $Tr_i$  base through  $C_1$  causing  $Tr_1$  and hence  $Tr_2$  to be switched hard on. In this quasi-stable state  $C_1$  charges through  $R_3$  and  $Tr_2$  towards  $-V_{EE}$  and when the charging current is insufficient to maintain conduction



in  $Tr_1$  this transistor switches off, as does  $Tr_2$  and the circuit returns to its stable state.

Fig. 10 shows a complementary monostable in which both transistors are on in the stable state, and which provides a pair of opposite-polarity outputs simultaneously. A positive-going trigger pulse applied to  $Tr_2$  base turns both transistors off and after  $C_1$  charges sufficiently, through  $R_2$ and  $R_4$ , both transistors return to the stable on state regeneratively.

### **Basic discrete-component circuits**



#### **Circuit description**

In the quiescent state with the trigger at 0V, Tr<sub>2</sub> is held on due to base drive via R<sub>3</sub>. Point  $\mathbf{\Phi}$  is therefore at zero volts and ence Tr<sub>1</sub> is off, its base being at 0V. In receipt of the trigger pulse the base of Tr<sub>1</sub> is forced positive,  $Tr_1$  conducts and its collector voltage drops. This is transmitted via  $C_1$  to the base of Tr<sub>2</sub> and hence Tr<sub>2</sub> is forced off, raising its collector voltage. By potential division via  $R_5$  and  $R_4$ , the base of  $Tr_1$  is forced positive, thus forcing  $Tr_1$ further into conduction. This bositive feedback action forces a rapid change in state of both  $Tr_1$  and  $Tr_2$ .  $C_1$  then has one side at 0V, point Q, and the other side connected via  $R_3$  to  $V_s$ . It thus commences to charge towards Vs. This continues until the potential at the base of  $Tr_2$  reaches a value which causes Tr<sub>2</sub> to start conducting. Point Q then starts to fall in potential causing the base of  $Tr_1$  to fall. This causes  $\bar{\mathbf{Q}}$  to rise, forcing the base of Tr, even higher and rapidly the two transistors change back to the quiescent state. Length of the output pulse depends on C<sub>1</sub>R<sub>3</sub>. See graph.

The original quiescent state is not reached until  $C_1$  has returned to its original uncharged state. The discharge path is via  $R_1$  and the base-emitter junction of  $Tr_2$ . Discharge time is what accounts for the recovery time,  $t_r$  minus the minimum time after the output pulse trailing edge at 
 Typical performance

  $Tr_1$ ,  $Tr_2$ : BC125

  $R_1$ ,  $R_2$ : 1kΩ

  $R_3$ : 22kΩ

  $R_4$ : 4.7kΩ

  $R_6$ : 10kΩ

  $C_2$ : 100pF

  $V_8$ : 5V

Min. trigger pulse width  $0.5\mu$ s Trigger pulse height in range 1.4 to 3.6V

Output pulse width  $t_p$  depends on  $C_1R_3$ —see graph Recovery time  $t_r$  depends on  $C_1R_1$ —see graph

Output pulse height 4.6V

which a further trigger pulse will give correct operation; see graph above.

Maximum trigger pulse height is fixed by the negative-going voltage applied to the base of  $Tr_1$  on receipt of the trailing edge of the trigger pulse. If this is greater than the bias set by  $R_4$  and  $R_5$  then  $Tr_1$  is switched off immediately and monostable action is prevented. Minimum trigger pulse height depends on  $C_2$  and on the pulse rise time.

### **Circuit modifications**

• Both of the above restrictions on the trigger pulse height may be reduced by applying the trigger pulse as shown Fig. 1. If this arrangement is used the min. pulse height should be about 0.7V and the maximum should be the breakdown voltage of the triggering transistor.

• A general-purpose diode e.g. 1N914 should be inserted in the base line of  $Tr_2$  as shown Fig. 2, if the supply voltage used is above 6V or so which is the base-emitter breakdown voltage of most planar transistors. If this is not done the linearity of  $t_p$  with respect to C is lost.

• If short recovery times are required between consecutive output pulses the circuit of Fig. 3 may be used. In the quiescent state both transistors are conducting, Q is low and  $\overline{Q}$  is high. On receipt of the trigger pulse the p-n-p transistor (BC126) is switched off causing



the base of the BC125 to fall in potential, thereby raising the voltage of Q and, via the  $10k\Omega$ resistor, the base of the BC126. The circuit thus latches into a condition where both transistors are off. Capacitor C<sub>1</sub> then charges via the 22k and  $1k\Omega$  resistors toward the supply and eventually turnsthe BC125 on and with it the BC126. Discharge of C<sub>1</sub> occurs more quickly in this case because the discharge path is through the collector-emitter junction of the BC126 and the base-emitter junction of the BC125. No external resistors are involved in this path as occurred previously. Hence, the circuit recovery time is considerably shorter.

### Further reading

Set 19: Monostable circuits—1

SGS Fairchild. Industrial Circuit Handbook 1967. Hemingway, T. K., Electronic Designer's Handbook. Business Publications 1967.



### Set 19: Monostable circuits—2

### **Complementary circuits**



**Circuit description** 

Monostable circuits using complementary transistors allow one to trigger either with respect to the positive supply line.

In the circuit shown both transistors are non-conducting in the quiescent state. On receipt of a positive-going pulse at trigger point 1, Tr<sub>2</sub> starts to conduct, dropping its collector voltage. This drop is transmitted via C to the base of Tr<sub>1</sub> and hence Tr<sub>1</sub> starts to conduct. This produces further base drive to  $Tr_2$  and pushes it further into conduction. Very quickly both transistors are fully conducting and the output. initially at Vs drops to 0V approximately.

Capacitor C, uncharged in the quiescent state, then begins to charge towards  $V_s$  via  $R_2$  and  $Tr_2$ . When it reaches  $V_s - V_{bc}$ ,  $Tr_1$  is cut off,  $Tr_2$  is cut off and the capacitor discharges via  $D_1$  and  $R_3$  principally. Insertion of  $D_1$  provides a low resistance discharge path giving a fast recovery time. When  $Tr_2$  is cut off the output

swings back to  $V_s$  so the output pulse period is determined by the charging time of the capacitor.

With the basic monostable circuit (card 1) the capacitor is, in the quiescent state, charged in one direction and this direction is reversed when triggered. Ignoring  $V_{be}$  effects, the capacitor during charging is going from  $+V_s$  to  $-V_s$ and the 0V or 50% point is sensed to bring the circuit back to quiescence. Because one is sensing a particular percentage of the attempted swing, the time taken is not supply voltage dependent. This is not the case here since a particular voltage is being sensed; the graph over shows the variation of pulse width with supply voltage. This could conceivably be used to produce a voltagedependent monostable action.

#### **Component changes**

Increasing the pulse height allows a reduced pulse width (trigger pulse). Reducing  $R_1$  to  $4.7k\Omega$  allows operation down to 6V and  $0.1\mu$ s trigger pulse width. Value of  $R_2$  is not critical;  $R_3$  depends on  $Tr_2$ ;  $R_4$  and  $R_5$  not critical.

#### Similar circuits

• The circuit of Fig. 1 will have zero stand-by power consumption but in this case has a positive-going output pulse. Again the output pulse width will depend on Vs, since in the quiescent state C is. uncharged. In addition the V<sub>CEsat</sub> of Tr<sub>2</sub> will affect the height of the pulse which should be  $V_s - V_{CEsat}$ . The 0V level should be well defined. In the circuit over the VCEsat affects the OV level but not the peak value (ref. 1). • The circuit of Fig. 2 has a pulse width given by

**Typical performance** Tr<sub>1</sub>: BC126, Tr<sub>2</sub>: BC125  $\mathbf{R}_1, \mathbf{R}_2: 33k\Omega$  $R_3$ : 1.2k $\Omega$ ,  $R_4$ ,  $R_5$ : 10k $\Omega$ **D**<sub>1</sub>: 1N914 With  $V_s = 12V$ , trigger frequency 500Hz, trigger pulse width and height  $10\mu s$ . and 2V respectively, graph opposite of output pulse width against C was obtained. With C = 47nF and pulse width of  $10\mu s$ , minimum trigger pulse height about 1.3V. With pulse height at 2V, minimum pulse width about  $2\mu$ s. Output pulse height about 12V; maximum mark-space ratio is 10: 1 at this frequency.

 $t_{\rm p} = 0.69(R_1 + R_3)C$  and a

 $t_r = R_s C/\beta$ . As recharging

Pulse width is not supply

dependent but since both

transistors conduct in the

quiescent state the standby

power is large. Two outputs

are available, at the collector

of both transistors. They are

Diodes are frequently

(ref. 2).

the complement of one another

included in monostable circuits

to prevent too large a reverse

voltage being applied to the

base-emitter junction of a

non-conducting transistor

current is supplied by Tr<sub>1</sub>, this

recovery time given by

recovery time is short.



(Fig. 3). A diode may be inserted in the emitter lead or in the base lead. The last-mentioned is generally preferred in that it will not affect the output pulse levels.

#### References

- 1 Electronic Circuit Design Handbook, Tab. p.174.
- 2 Electronic Circuit Design Handbook, Tab. p.85.

Cross references Set 18, card 1 Set 10, card 4



### **Op-amp/comparator circuits**



### **Circuit description**

This comparator features an open-collector output, and may be used with pull-up resistor R<sub>2</sub> and a separate supply to give a wide switching range. Resistors  $R_3$  and  $R_4$  set the reference voltage equal to  $V_s/2$ , but also determine the trigger pulse level necessary for switching. Resistor R<sub>1</sub> should be much greater than R<sub>2</sub> to avoid loading the output. With  $V_{\rm ref} = V_{\rm s}/2$  and the noninverting input near 0V, the output transistor of the comparator is conducting and the output is at VCEsat. A negative trigger pulse at the input causes this transistor to switch off and hence the voltage at the output rises to almost V<sub>s</sub>. Capacitor  $C_1$  now charges at a rate mainly determined by R1 and  $C_1$ , which also control the width of the output pulse. The potential of the non-inverting input now falls exponentially from Vs and the comparator switches to its original condition when the noninverting comes below  $V_s/2$ . Diodes  $D_1$  and  $D_2$  prevent the inputs from being driven below zero volts, and  $D_1$  also provides a low impedance discharge path for  $C_1$ .

Typical performance IC<sub>1</sub>:  $\frac{1}{4}$  LM139 V<sub>s</sub>: +10V C<sub>1</sub>: InF C<sub>2</sub>: 470pF R<sub>1</sub>: 330k $\Omega$ R<sub>2</sub>: 10k $\Omega$ R<sub>3</sub>, R<sub>4</sub>: 1M $\Omega$ D<sub>1</sub>, D<sub>2</sub>: 1N914 Trigger pulse width: 30 $\mu$ s Trigger pulse level: 6V Output pulse width: 190 $\mu$ s Output pulse level: 9V

Component changes IC<sub>1</sub>:  $\frac{1}{4} \times MC$  3302P Useful range of C<sub>1</sub>: 220p to 22nF

Useful range of  $R_1$ : 47k to 470k $\Omega$ 

Minimum trigger pulse width:  $5\mu s$ 

Minimum trigger pulse height:  $5V(\text{for } V_s = +10V)$ Range of V<sub>s</sub>: up to 28V. • Increasing trigger pulse

height to 8V decreases pulse width by 150 to 200µs.
Variation of pulse width with capacitance shown, for

trigger pulse 5V. • Output pulse is supply dependent, see graph above. (For  $C_1=1nF$ ,  $R_1=330k\Omega$ , p.w. is 50 $\mu$ s.)

#### **Circuit variations**

• Circuit shown left gives a pulse width that is supply independent, and is capable of a high duty cycle. Capacitor  $C_3$  is initially charged to  $+V_{ec}$  and  $V_{out}$  is at almost zero volts. Potentials at X and Y are defined by associated divider chains. A positive input trigger pulse to make  $V_X > V_Y$  causes IC<sub>1</sub> to switch (output driven low). Capacitor  $C_3$  rapidly discharges, and IC<sub>2</sub>



will switch when the voltage across  $C_a$  falls below  $V_Q$ , defined by R<sub>11</sub>, R<sub>10</sub>. Output voltage is now equal to Vee, which is fed back bia  $R_7$  to  $IC_1$ which again switches to off. condition. Hence C<sub>3</sub> now charges up via R<sub>8</sub>. Output pulse width is defined by the time for the voltage across C<sub>3</sub> to just exceed that of point Q; IC<sub>2</sub> then reverts to its on, or low-output state. IC<sub>1</sub>, IC<sub>2</sub>  $\frac{1}{4}$  LM139, R<sub>5</sub>, R<sub>4</sub> 100k $\Omega$ ,  $R_6 1M\Omega$ ,  $R_7 62k\Omega$ ,  $R_8$ ,  $R_9$ 10M $\Omega$ , R<sub>10</sub> 220k $\Omega$ , R<sub>11</sub> 560k $\Omega$ ,  $C_3 0.1 nF.$ 

Note that retriggering is not possible while IC<sub>2</sub> output is high-a useful lock-out feature. Circuit centre, using an op-amp, has a similar action to the comparator network. A negative reference voltage on the inverting input, with the non-inverting input tied to ground, gives a stable condition where  $V_{out} = + V_s$ . A positive input trigger sufficient to cause the differential input polarity to change, the output switches to  $-V_{\rm s}$ . At this instant the potential of R<sub>1</sub> with respect to the potential across R<sub>1</sub> rises

to initial condition. Component values:  $IC_3$ MC1530/1531, C<sub>2</sub> 23pF, R<sub>2</sub> 100 $\Omega$ , C<sub>1</sub> 10nF, R<sub>1</sub> 10k $\Omega$ , V<sub>ref</sub> = 2V. Pulse width typically 125 $\mu$ s. Diode and R<sub>3</sub> allow faster discharge of C<sub>1</sub>, permitting an increase in p.r.f. • Circuit right is a useful

interface for delaying logic signals. When logic level is 1, capacitor charges quickly via diode, discharges via  $R_x$  when logic level is 0, allowing  $V_{out}$ to rise when  $V_{Cx} \rightarrow +1V$ . Using LM311, claimed delay is typically 0.04 to 370ms.

Cross reference Set 17, card 8.

=+1V



### Set 19: Monostable circuits—3

### Set 19: Monostable circuits-4

### Monostable using t.t.l. gates



#### **Circuit description**

The circuit above uses two of the four NAND gates available in the integrated circuit package. Positive trigger pulses are applied to the first gate which is connected as an inverter, the unused input being taken to the positive supply rail (logic 1 state) via a 1-k $\Omega$  resistor to prevent the gate being damaged by a supply line transient. With the trigger input in the logic 0 state (0V), the output of the first gate is at 1, as are both the inputs to the second gate in the steady state, causing its output to be at logic 0. A positive trigger pulse causes the output of the first gate to go to its logic 0 level. One of the two inputs to the second gate immediately goes to the logic 0 level but its other input receives the voltage across C<sub>1</sub> which cannot change its charge instantaneously. The inputs to this gate are thus at logic 0 and 1 respectively, causing its output to switch to 1 where it remains until the trigger pulse returns to the 0-volt level. During this interval, capacitor C1 discharges towards 0 volt through  $R_1$  and the output impedance of the first gate and is recharged to the 1 level after the trigger pulse has returned to the 0-volt level.

### Circuit changes

Useful range of  $V_{ee}$ : +4.5V to +5.5V

Maximum useful  $R_1$ : 470 $\Omega$ Maximum useful  $C_1$ : 1.5 $\mu$ F Minimum trigger pulse amplitude: +3V Minimum trigger pulse width:

 $40\mu s$ Output pulse shape improved

by cascading a third gate (see waveform and graphs above).

### Typical performance

Logic gates:  $\frac{1}{4} \times 7400$ Supplies: +5V; 8mA R<sub>1</sub>: 100 $\Omega$ , R<sub>2</sub>: 1k $\Omega$ , C<sub>1</sub>: 1 $\mu$ F Input pulses: amplitude +5VWidth: 50 $\mu$ s, p.r.f. about 10 kHz Output pulse width about 74 $\mu$ s

Cascading the fourth gate in the package provides a pair of complementary outputs.

#### **Circuit modifications**

The simple two-gate circuit shown in Fig. 1 has positive trigger pulses fed simultaneously to one input of each gate. These pulses appear at the output of the first gate inverted and delayed by the propagation time of the gate. Both of the inputs to the second gate are therefore at the 1 level for a short period, providing a narrow negativegoing output pulse, delayed by the propagation time of one gate. This circuit has two defects: the output pulse is very narrow and non-adjustable and the trigger pulse must be wider than the output pulse.

The circuit shown in Fig. 2 overcomes the first defect but still requires a trigger pulse wider than the output pulse. When the trigger input is at logic 0 the output of the first gate is at 1. Inputs to the second gate are at logic 0 and 1 since C is charged from the first gate, hence the output is at 1. When the trigger pulse goes to 1 the output of the second gate switches to the 0 state for an interval determined by the time constant, as C discharges through R, and the output impedance of the first gate. When C has discharged to a level where its voltage crosses the switching threshold of the 1 to 0 transition the output from the second gate switches back to its 1 level. Trigger pulse width must still be greater than the output pulse width, as the output of the second gate will always



return to the 1 state whenever the trigger pulse returns to 0 volt.

• To overcome this problem the trigger pulses can be fed to the monstable via a third gate which has an output logic state determined by the monstable output, as shown in Fig. 3. Addition of this gate causes



the monostable to be triggered when the input transition is from the 1 to the 0 state and the width of the input pulses is only that required to cause the input and output gates to change state. Since the same time constant controls the charge and discharge rates of C both circuits have relatively low maximum duty cycles, typically around 25%.

• Circuit of Fig. 4 uses only two gates but has a very similar mode of operation, the output pulses controlling the state of the input gate. This gate will normally have a 0 output since both its inputs will be at 1 until the trigger pulse goes to 0 volt. This causes the output to switch states for a time determined by that required for the CR network to cross the logic threshold.

### Further reading

Malmstadt, H. V. and Enke, C. G., Digital Electronics for Scientists, Banjamin 1969, pp. 213-5. Wilke, W., Operating a logic gate as a flip-flop, *Electronics*, 21 March 1974, pp.120/1. Kaniel, A., Digital delay circuit for one-shot controls timing interval in programmable integer steps, *Electronic Design*, 18 Jan. 1974, p.94.

Cross reference Set 11, card 3.

### Set 19: Monostable circuits—5

### Compensated c.m.o.s. circuits



### **Circuit description**

Complementary m.o.s. monostables using one time constant suffer from the fact that the transfer voltage  $(V_{tr})$ of i.cs can vary by as much as  $\pm 33\%$  so that replacement of an i.c. by another can result in a considerable change in the output pulse width. This defect can be removed by using two identical time constants with two inverters on the same chip. In this case the transfer voltage of both inverters is almost identical and the effect is cancellation of the variation in transfer voltage. This is taken advantage of in the circuit shown.

In the quiescent state with the input pulse low (ground), point  $\hat{2}$  is high ( $V_{DD}$ ), points 3, 4 and 5 are therefore low, low and high respectively. Capacitor  $C_1$  is therefore in a charged state. On receipt of the input pulse this charge is removed rapidly via the source resistance and  $D_1$  which is an internal protection device to prevent the input rising beyond  $V_{DD}$ . On the negative-going edge of the input pulse point 2 is taken down to ground and  $C_1$  commences to charge towards V<sub>DD</sub> via R<sub>1</sub>. Point 3 will go high and remain in this state until C1 passes through the transfer voltage of inverter 1. When point 3 goes low C<sub>2</sub> will commence to discharge via  $R_2$ ,  $D_2$  acting to prevent discharge into inverter 1. When point 4 passes through the transfer voltage of inverter 2 point 5 will go high and the operation is complete. The graphs show that there are two possible outputs-at points 3 and 5. The output at point 5 is the only one in which there is compensation

for transfer voltage variation. Due to lack of feedback in the circuit the risetime of the output is decidedly long (see performance data) but this is acceptable if the circuit is supplying further c.m.o.s. circuitry.

Despite the protection of D<sub>1</sub> the fact that the circuit is being fed via a capacitor makes it advisable to ensure that the input pulse height is close to  $V_{DD}$ . This can be done if the pulse is fed via further c.m.o.s. supplied from  $V_{DD}$ . This circuit can be re-triggered, i.e. the output 5 can be maintained in its low state by feeding in a succession of pulses, provided each pulse occurs before point 2 reaches the transfer voltage of the first inverter.

### **Circuit modifications**

A positive-going output pulse, again triggered by the negative-going edge of the input is obtainable with the circuit shown top. In this case the output is obtained at the output of the first inverter viz point 3. This output is compensated, whereas that at 5 is not. Relevant waveforms are shown in Fig. 2. In this case re-triggering is not possible because any positivegoing trigger pulse would reduce  $T_1$  no matter the state of point 5.

• A negative-going pulse triggered on the positive-going edge of the input pulse is obtainable from the circuit shown in Fig. 3. For compensation, the output is taken from point 2 in this diagram. Relevant waveforms are as in Fig. 4. With this circuit re-triggering is also

### Components

 $D_2$ : general-purpose diode Inverter 1 and 2:  $\frac{1}{3} \times CD4007AE$ 

#### Performance

 $R_1C_1 = R_2C_2 = RC$   $T \approx 1.4RC$ ; tested in range  $10\mu s < T < 1ms$ With clock rate of 1kHz and pulse width 400 $\mu s$ , T remains within 4% of the nominal value over supply range 5V to

impossible because point 2 will stay low so long as point 5 is high, the trigger signal having no effect due to the normal NOR gate action.

#### **Further reading**

RCA applications note ICAN6267. Murphy, C. Simple reconnection reduces rise time of CMOS delay circuit. *Electronic Design*, 4 Jan. 1974.





10V. With T of  $330\mu$ s fall time is minimal but rise time is  $120\mu$ s.

### Set 19: Monostable circuits—6

OUTPUT PULSEWIDTH (µs)

10

### **Emitter-coupled circuits**



#### **Circuit description**

In the stable state Tr<sub>2</sub> is on and saturated due to the base drive via R5. The emitter current of Tr<sub>2</sub> causes the common emitter voltage to be at approximately 1.5V. Thus  $Tr_1$  is held off by the  $R_1$ ,  $R_2$ potential divider which maintains its base at about 1.3V, slightly reverse-biasing the base-emitter junction. Output voltage at Tr<sub>2</sub> collector is at approximately 1.6V in this stable state. The C<sub>1</sub>, R<sub>1</sub>, R<sub>2</sub> network differentiates the input trigger pulses and, if these have a positive-going transition large enough to exceed the emitter voltage by  $V_{BE}$ ,  $Tr_1$ will switch on and saturate. The collector voltage of Tr<sub>1</sub> falls by about 4.5V and this negative-going transition passes to  $Tr_2$  base through  $C_2$  so this transistor switches off. Output voltage rises to  $+V_{CC}$  and  $Tr_2$ remains in the off state while  $C_2$  charges via  $R_5$ ,  $Tr_1$  and  $R_4$ . As  $C_2$  charges, the base voltage of Tr<sub>2</sub> rises exponentially towards  $+V_{cc}$  but when it exceeds the emitter voltage, Tr<sub>2</sub> rapidly turns on and saturates. Current in R4 increases, producing a corresponding rise in the emitter voltage which reversebiases the base-emitter junction of Tr<sub>1</sub>, which switches off, and the circuit returns to its stable state.

#### **Circuit changes**

Minimum trigger pulse width  $0.5\mu$ s. Output pulse width could be varied by R<sub>5</sub> but this would also change the d.c. conditions. Larger supply voltages may be used but then a silicon diode should be included in series with Tr<sub>2</sub> base

Typical performance Supply: +6V, 4mA Tr<sub>1</sub>, Tr<sub>2</sub>: BC125 R<sub>1</sub>:  $6.8k\Omega$ R<sub>3</sub>:  $1.8k\Omega$ R<sub>3</sub>:  $2.7k\Omega$ , R<sub>4</sub>:  $330\Omega$ R<sub>5</sub>:  $22k\Omega$ , R<sub>6</sub>:  $1k\Omega$ C<sub>1</sub>: 100pF, C<sub>2</sub>: 100nF Input pulse amplitude: +2VInput pulse width:  $1\mu$ s p.r.f.: 1kHz

to prevent reverse breakdown of its base-emitter junction by the switch-off transient.

#### **Circuit modifications**

• Adding an emitter-follower to the original circuit, as shown in the top left diagram, allows the output pulses to be obtained from a low-impedance source and to be referred to ground. By adjustment of the supply voltage these pulses may be made compatible with t.t.l. logic circuitry.

• A p-n-p common-emitter stage may be added, as shown top right, to use the output pulses for driving a groundconnected load. For example, the load could be in the form of a filament lamp for alarm or monitoring purposes and as collector currents of at least Output pulse—see waveform below



50mA may be obtained, the load could be a relay coil to provide control of some other circuitry. The inductive load provided by the relay coil should be shunted with a protective diode.

The original circuit may be modified to be triggered by negative-going pulses as shown in the circuit shown bottom left. In this arrangement the trigger pulses are applied to the collector of  $Tr_1$  via the small coupling capacitor C<sub>3</sub>. The negative-going trigger pulse edge is passed to Tr<sub>2</sub> base through  $C_2$  switching this transistor off.  $Tr_1$  turns on and the remaining cycle of operations are as described previously. The circuit will function with  $C_1$  omitted but the transition to the quasi-stable state will be more rapid when it is included in the circuit.

10 C<sub>2</sub> (nF)

 $R_5 = 22 k \Omega$ 

• In all of the circuits discussed the output point was isolated from any internal coupling. If, see bottom right,  $R_1$  of the original circuit is taken to  $Tr_2$  collector instead of the  $+V_{CC}$  rail this is no longer the case and changes in emitter voltage may be very small.

The simplest way to provide faster switching is to prevent  $Tr_1$  and  $Tr_2$  from saturating by increasing  $R_4$ .

### Further reading

SGS-Fairchild, Industrial Circuit Handbook, 1967, pp.51/2. Budinsky, J., Techniques of Transistor Switching Circuits, Iliffe 1968, pp.486-91.

#### **Cross references**

Set 18, cards 1 and 7.



### Set 19: Monostable circuits—7

### Voltage-controlled monostables



### **Circuit description**

Transistors Tr<sub>1</sub>, Tr<sub>2</sub> form a complementary-pair positive feedback switch, and are either both off or both on. In the rest state, this switch is closed,  $Tr_3$  is on,  $D_1$  is non-conducting, and  $C_1$  is charged. A negative trigger pulse applied at the trigger terminal causes the regenerative switch  $(Tr_1, Tr_2)$ to open. Collector current of Tr<sub>3</sub> is now  $\alpha I$ , where  $\alpha$  is common base current gain and  $I = (V_{\rm S} - V_{\rm BE})/R_{\rm s}$ . Hence C<sub>1</sub> discharges at a constant rate giving a run-down linear sweep at point A. This sweep continues until the potential of point A is less than  $V_{\text{control}} - V_{D_1} - V_{BE(Tr_2)}$ , when the complementary switch again conducts. The switch also acts as a comparator, besides providing a low impedance path for recharging capacitor C<sub>1</sub>. Notice that the sweep termination depends on the level of the control voltage, which therefore controls the width of the output pulse.

#### **Component changes**

 $Tr_2$ ,  $Tr_3$ : BSX28,  $Tr_1$ : BSX29 Useful range of  $C_1$ : 1 to 47nF Useful range of  $R_3$ : 10k to 100k $\Omega$ 

Useful range of control voltage is in the range 10% to 90% of  $+V_s$ .

To minimize inductive and stray capacitive effects, this circuit was constructed on a p.c.b. to obtain above measurements.

Rise and fall times of output

Typical performance  $Tr_1: BC126$   $Tr_2, Tr_3: BC125$   $R_1: 22\Omega, R_2: 1.2k\Omega$   $R_3: 10k\Omega$   $C_1: 1nF, C_2: 100pF$   $+V_S: +10V$   $D_1: 1N914$ Trigger pulse magnitude: -13VTrigger pulse width:  $0.1\mu s$   $V_{control}: 8V$ Output pulse duration:  $7\mu s$ Waveforms and timing graph shown opposite, curve (a)

pulses less than  $0.5\mu s$ . Control voltage may be obtained from potential divider connected across supply.

### **Circuit modifications**

• Connect additional buffer stage  $Tr_4$  (BC126) and  $R_4$ (390 $\Omega$ ) for increased output pulse amplitude (below). Trigger pulse amplitude is 7V, when coupling capacitor  $C_2$ increased to 1nF; provides timing graph shown above a (b).

• Circuit (centre) is basically an emitter-coupled monostable circuit. Time duration may be extended by the inclusion of a complementary pair ( $Tr_6$ ,  $Tr_7$ ) with breakdown fixed by zener diode  $D_2$ . This effectively increases the cut-in voltage,



 $V_b$ , of  $Tr_8$ .  $Tr_8$  is normally on. Trigger pulses cause  $Tr_5$  to conduct ( $Tr_8$  off), and circuit returns to its stable state when the voltage at B brings the regenerative switch into conduction. Pulse duration depends on I, which is controlled by voltage applied to base of transistor  $Tr_5$ . Output pulse duration typically in the range 40 to 200ms for a control voltage of 3 to 6V.

• Operational amplifier  $IC_1$ (MC1530/MC1531), shown right, can provide variable delays by controlling the voltage at the inverting input. Diode clamps the negative amplitude of the output to about -0.7V. Positive output will approach +V.

#### Further reading

Rakovich, B. D. & Djurdjevich, B. Z., Wide-range voltage-to-time converter, *IEEE Transactions* vol. IM-22, no. 2, June 1973, p.162. Time delay circuit for SCR control, *Electronic Eng.* June 1974, p.14. Motorola application note AN-258.

Cross references Set 18, card 3, 6.



### Set 19: Monostable circuits—8

### Long-delay circuits using c.m.o.s. and unijunction devices



#### **Circuit description**

The high input impedance of c.m.o.s. integrated circuits offers a low leakage path for capacitor  $C_1$ , and thus allows a wide range of delays from this D-type flip-flop configuration. The D-input is permanently connected to the positive supply rail, hence Q goes high when a positive trigger pulse is applied at the clock input. A positive voltage approximately 50% ( $V_{\rm DD} - V_{\rm SS}$ ) at terminal R will reset O to its initial condition of OV. On the occurrence of a trigger pulse, the reset threshold will be reached in a time defined by  $C_1$  and  $R_1$ , which in turn defines the delay available at Q. This circuit is fairly insensitive to supply voltage change, because the threshold level for reset will alter pro-rata. However it should be noted that this level is not well defined, and hence the accuracy of time delays will depend on specific measurement with the components used. Unused inputs should be tied down to ground.

### Component changes

Useful range of  $R_1$ : 100k to 10M $\Omega$ Useful range of  $C_1$ : 47pF to 32 $\mu$ F (For long time delays, lowleakage capacitors must be used) Supply voltage range: ( $V_{DD}-V_{SS}$ ) 3 to 15V Minimum pulse width: 0.1 $\mu$ s Minimum pulse height: 2 to 4V Alternative IC: MM74C74

### UJT transistor monostable

The circuit shown right (top). employs a discrete transistor flip-flop, with the timing controlled by the firing of the unijunction transistor  $Tr_1$ . Typical performance IC<sub>1</sub>:  $\frac{1}{2} \times CD$  4013AE V<sub>DD</sub>: +5V, V<sub>SS</sub>: 0V R<sub>1</sub>: 470k $\Omega$ , C<sub>1</sub>: 10 $\mu$ F Trigger pulse width: 1 $\mu$ s Trigger pulse height: 4V Delay: 9s Typical waveforms and delays shown opposite.

The normal rest condition of bistable FF1 is  $Tr_3$  off and  $Tr_2$ conducting. Hence the available voltage at point A is approximately  $V_{CEsat}$  of  $Tr_2$ . This is much less than the firing voltage  $V_P$  of the unijunction transistor which therefore is non-conducting, and the output voltage is at zero volts.

A positive pulse applied at the trigger input brings Tr<sub>s</sub> into conduction, and due to positive feedback Tr<sub>2</sub> is rapidly switched off. Hence the potential of point A rises. Voltage across C<sub>2</sub> now rises exponentially toward  $+V_s$ . When it reaches  $V_{\rm P}$  in a time determined by  $R_{2}C_{2}$ , Tr<sub>1</sub> will conduct and discharge C2 via the emitter and the resistor R<sub>E</sub>. The output pulse developed across R<sub>E</sub> is applied to the base of Tr<sub>2</sub> to turn it on again and hence reset the bistable, and turn off Tr<sub>1</sub>. Well-defined output pulses are also available at the collectors of Tr<sub>2</sub> and Tr<sub>3</sub>.  $R_2 1M\Omega$ ,  $C_2 1\mu$ F,  $R_E 27\Omega$ , Tr<sub>1</sub> 2N4853, Tr<sub>2</sub>/Tr<sub>3</sub> 2N4123,  $V_{\rm S}$  + 10 to + 30V.

An advantage is that the initial voltage across C<sub>2</sub> is always the same at the start of a period no matter the pulse repetition frequency. It is claimed that with the supply voltage variation noted above, pulse duration changes are  $2^{\alpha/2}_{10}$ . Circuit (bottom) uses the transistor array CA3095E. Diodes  $D_1$ ,  $D_2$  and  $Tr_5$  act as a voltage-limiting network. Ouiescent condition is with transistors Tr<sub>1</sub> and Tr<sub>3</sub> off, Tr<sub>2</sub> and Tr<sub>4</sub> on. A negative-going pulse at the trigger-input terminal is coupled to the base of Tr<sub>2</sub> via diode-connected transistor Tr<sub>7</sub>. Hence Tr<sub>2</sub> (and Tr<sub>4</sub>) cut off and V<sub>out</sub> rises to  $+V_s$ . Because Tr<sub>2</sub> and Tr<sub>1</sub> are



emitter-coupled,  $Tr_1$  can now conduct when  $Tr_2$  is off, therefore the voltage across  $C_3$ i.e.  $V_L$  is near ground. Capacitor  $C_3$  now charges via  $R_3$ . When the base potential of  $Tr_2$  is high enough to bring it into conduction,  $Tr_1$  again goes off and  $V_{out}$  drops to its normal value of about 1.8V. Time duration is  $t \approx 0.47R_3C_3$ . Cross references Set 18, cards 5, 9.

#### Further reading

Unijunction Transistor Timers and Oscillators. Motorola application note AN-294.



### Dual monostable applications using 555 timer



#### **Circuit description**

The i.c. shown is a dual 555-type timer (internal details of which are shown in cross references 1 and 2). In this application the output of the first timer is capacitively coupled via  $C_{\delta}$  to the input of the second. Both timers are triggered when the input to each drops below  $\frac{1}{3}V_{\rm CC}$ . In this case the trigger pulse operates timer 1 and the output of timer 1 triggers timer 2. Both timers are connected in their monostable mode. Formulae for  $T_1$  and  $T_2$  quoted are valid for all conditions and are fixed entirely by the external  $R_1$ ,  $C_1$ and R<sub>1</sub>, C<sub>2</sub>. Any succeeding trigger pulses are ignored. Both  $T_1$  and  $T_2$  can be anywhere in the range  $10\mu$ s to 100s with two provisos:  $T_1$  cannot be reduced below  $T_p$ , but since  $T_p$  can be less than  $3\mu$ s this is not serious. Secondly low values of  $T_1$  and T, should not be obtained by reducing R<sub>1</sub> and R<sub>2</sub> to a level at which current drain from the supply is serious. A minimum value of  $1k\Omega$  is recommended for both R<sub>1</sub> and R<sub>2</sub>.

The only restriction on  $R_L$  is that it must not be so low as to exceed the current rating (200mA) of the i.c. Typical waveforms are shown above. These are suitable for sequential timing circuits. Note that trigger pulses occurring during the period set by  $R_1$ ,  $C_1$ and  $R_2$ ,  $C_2$  are ignored.

#### Circuit modifications

• Obviously the i.c. may be used as two separate timers or

monostables by omitting  $C_s$ and  $R_s$  and by applying the trigger pulses directly to pins 2 and 12.

• Referring to the main diagram, control over the total period  $(T_1+T_2)$  can be achieved by connecting the arm of a potentiometer to V<sub>CC</sub> and the ends to pins 3 and 9, R<sub>1</sub> and R<sub>2</sub> being now the two segments of the potentiometer shown above. If  $C_1=C_2=C$  then  $(T_1+T_2)=C(R_1+R_2)$  which is constant no matter the individual T<sub>1</sub> and T<sub>2</sub>.

• Each individual timer can be regarded as a frequency divider in its own right. For example, in the waveforms shown over, the lower two graphs indicate that output 1 is running at half the trigger pulse frequency and output 2 is running at half the frequency of output 1. By correct choice of  $T_1$  one can make output 1 run at any submultiple frequency of the trigger and likewise with correct choice of T<sub>2</sub> output 2 can run at any submultiple frequency of output 1.

• A combined frequency divider and pulse width controller can be achieved by setting  $T_1$  so that it is greater





Set 19: Monostable circuits—9

than the trigger period and by ensuring that  $T_2 < T_1$ . The value of  $T_1$  sets the frequency division and the value of  $T_2$ sets the pulse width, the output being output 2. Furthermore, the "phase" of the output with respect to the trigger can be controlled since, for example, frequency division by 3 allows  $T_1$  to lie anywhere between two and three trigger periods. See waveforms above.

### Further reading

XR-2556 Dual Timing Circuit, data sheet/application note, Exar Integrated Systems Inc.

Cross references Set 3, card 9 Set 13, cards 3 and 5

### High duty-cycle monostable



#### Circuit description

In this circuit the timing capacitor C<sub>3</sub> not only determines the width of the output pulse obtained from Tr<sub>2</sub> collector but also delays the setting-up of the circuit in readiness for the next cycle of operations. This prevents an early trigger pulse from producing an output pulse having too short a duration. Because the circuit may be re-triggered immediately after completing its cycle, an output-pulse duty cycle approaching 100% may be obtained. The conventional "cross-coupled" monostable allows only about 95% duty cycle, even when its timing capacitor is rapidly recharged from a low-resistance source. In the stable state,  $Tr_3$  is biased off and Tr<sub>1</sub>, Tr<sub>2</sub> and Tr<sub>4</sub> are on. Input trigger pulses are differentiated by  $C_1 R_1$ , the positive component passing to  $Tr_3$  base via  $D_1$ . If of sufficient amplitude, these overcome the reverse bias on  $Tr_3$ , which switches on,  $Tr_1$ ,  $Tr_2$  and  $Tr_4$ switching off. Transistor Tr<sub>4</sub> remains off for approximately  $0.7\tau_1$  as its base potential rises under the control of C<sub>8</sub>. When Tr<sub>4</sub> turns on again, Tr<sub>3</sub> collector potential rises as C<sub>3</sub> recharges with time constant  $\tau_2 = C_3 R_3$ until it exceeds approximately 0 V. Transistor Tr<sub>2</sub> and hence  $Tr_1$  then switch on regeneratively and the circuit returns to its stable state and may be re-triggered immediately.

### **Component changes**

Minimum useful supplies  $\pm 2 \text{ V}$ Useful range of R<sub>1</sub>: 1k to 100k $\Omega$ Useful range of R<sub>2</sub>: 1k to 470k $\Omega$  Typical performance V<sub>CC</sub>: +6V, 6 to 18mA\*, V<sub>EE</sub> -6V, 6 to 12mA\* Tr<sub>1</sub>: BC126, Tr<sub>2</sub>, Tr<sub>3</sub>, Tr<sub>4</sub> BC125 R<sub>1</sub>: 10k $\Omega$ , R<sub>2</sub>, R<sub>7</sub>: 56k $\Omega$ R<sub>3</sub>, R<sub>8</sub>: 47k $\Omega$ , R<sub>4</sub>: 1.8k $\Omega$ R<sub>5</sub>, R<sub>6</sub>: 1.2k $\Omega$ , C<sub>1</sub>, C<sub>2</sub>: 100pF C<sub>3</sub>: see graph opposite; D<sub>1</sub> PS101 Input: 4V positive pulses, pulse width 1 $\mu$ s, p.r.f. about 10kHz Output rise time: 0.5 $\mu$ s Output fall time: 0.2 $\mu$ s \*depending on value of C<sub>3</sub>.

Minimum trigger pulse amplitude approximately 2.5 V Minimum trigger pulse width less than about  $0.1\mu$ s Maximum useful p.r.f.: 200kHz.

#### **Circuit modification**

The circuit shown top is a logical representation of a high duty-cycle monostable which may be realised with logic gates or discrete transistor circuitry. A pair of pulse-steering AND gates and a flip-flop used as a toggle cause the positive-going input trigger pulses to be alternately applied to the pair of monostables (M). The monostable outputs are combined in the OR gate to provide the output pulse and to toggle the flip-flop which changes state when the output pulse is completed. Hence, the start of the next output is only delayed by the time taken for the flip-flop to change its state, which can be a small percentage of the output pulse duration. A discrete version produced 700ms pulses every 705ms.

Circuit shown next uses a ٠ flip-flop in conjunction with a Shockley diode (SD1) to achieve a high duty cycle. An input pulse causes the flip-flop to change its state producing an exponential voltage rise across  $C_4$ . When this voltage reaches a critical value the diode breaks down rapidly discharging  $C_4$ and producing a negative pulse to reset the flip-flop via  $C_3$  in readiness for the next input pulse. Pulses of  $2\mu$ s duration may be produced after only a 250ns recovery time.



Typically,  $R_1$ ,  $R_6 \quad 1M\Omega$ ,  $R_2$ ,  $R_3 \quad 10k\Omega$ ,  $R_4$ ,  $R_5 \quad 150k\Omega$ ,  $R_7$ , 2.2M $\Omega$ ,  $C_1$ ,  $C_2$ ,  $C_3 \quad 20pF$ ,  $C_4 \quad 500nF$ ,  $D_1 \quad 1N914$ , SD1 4E20-8, with V<sub>CC</sub> + 25V.

• Circuit shown bottom can provide a recovery time of only 1% of its period and may be triggered by positive pulses at  $Tr_1$  base or negative pulses at  $Tr_2$  base causing both transistors to switch from their on to off states. Transistor  $Tr_1$  can recharge  $C_1$  with a large collector current to rapidly regain its initial voltage. Diode  $D_1$  and  $D_2$  may be omitted if  $Tr_1$  and  $Tr_2$  have a large reverse  $V_{\rm BE}$  rating.

#### Further reading

Shagena, J. L. & Mall, A. Single-shot multivibrator has zero recovery time, *Electronics*, 27 Nov., 1967, p. 83. Electronic Circuit Design Handbook, 4th edition, Tab, 1971, pp. 85, 96, 342/3.

Cross reference

Set 19, card 9.



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### Set 19: Monostable circuits Up-date

Disadvantages of monostables in digital circuits caused by their sensitivity to noise on the output line or transients in the power supply are avoided by using integrated-circuit monostables such as the SN74121. Two applications are considered.

1. The normal timing resistor for the monostable connected between pins 14 and 11 is replaced by the impedance of a transistor. The input trigger pulses to pin 3 provide predetermined pulse width at the output of the monostable depending on  $C_2$ and the effective resistance provided by the transistor which is changed by the feedback loop to maintain the duty cycle. The output pulse defines the time for which

2. Integrated-circuit CA3081 contains seven n-p-n transistors with the emitters commoned; The control of pulse width from the integrated-circuit monostable is achieved by the **R-C** combination across pins 14, 11, 10. Hence by driving the base of one of the transistors  $Tr_1$  to  $Tr_2$ , a different capacitor is switched across pins 10, 11. The absolute value of capacitor depends on which of the input terminals are taken high or low respectively, and this may be



current  $I_1$  flows into  $C_1$ . For Vout high,



For Vout low







average current to the capacitor is zero, when  $\alpha$  has reached the correct value. The above expression for  $I_{av}$  is then zero and

$$x = \frac{V_{\text{ref}} - V_{\text{outL}}}{(V_{\text{outH}} - V_{\text{outL}})}$$

To avoid loading the monostable output and thus possibly varying  $V_{out_{\rm H}}$  and  $V_{\text{out}_{L}}$ , an alternative output is taken from pin 1 via an inverter. Variable base drive due to V<sub>f</sub> can cause the transistor impedance to vary between 500  $\Omega$  and 500k  $\Omega$ . allowing a pulse repetition frequency range of the order of 1000:1.

#### Reference

Langon, H. P. D. Electronics, July 24, 1975, p. 93.



controlled sequentially, as in the specific application of the reference, using a read-only memory package. It is suggested that temperature compensation for the pulse width might be achieved by including a thermistor as part of the tuning resistor network R.

Reference Pulford, I. P. & Risk, R. J.

Applied Ideas: Electronic Engineering, Sept. 1974.

3. The input impedance of c.m.o.s. gates is very high and therefore provide a useful means of obtaining long delays with relatively small capacitance values. This circuit provides a means of eliminating the slow rise of capacitor voltage, when it is normally connected between gate-input and the  $V_{SS}$  rail. When V<sub>C</sub> reaches the lower threshold value, then Vout begins to go positive. As the charge on the capacitor

cannot change instantaneously,  $V_{\rm C}$  follows the output at a rate determined only by the c.m.o.s. gate. A similar regenerative effect will occur

on a negative transition.

Reference

Electronic Design 1, January 4, 1974.



### notes

- 3

### Set 20: Transistor pairs

Tweedledum's query seems an especially appropriate way of starting the following article on transistor pairs. Not only are the advantages of each configuration not always clear, but sometimes it's difficult to see the true operational nature of the configuration. In the cards, 20 different connections of two transistors are described, excluding c.m.o.s. circuits and triple or mixed pair arrangements. This set of circuits is an important one in that many form the basic building bricks of integrated circuits-long-tailed pairs, current mirrors, and even cascode circuits and high-gain pairs. Both the article and card 4 discuss the merits of the different long-tailed pair configurations, the card giving useful formulae for gain, common-mode rejection and temperature drift. Cards 1 & 2 contain a thorough investigation of the properties of high current gain pairs-Darlington, complementary Darlington and four-terminal pairs-the results being given in the 16 graphs, all of course derived from the compiler's measurements. Various circuits can be developed by mixing various pairs and two examples are given on card 9. One uses a long-tailed pair with constantcurrent tail feeding a current mirror to produce a triangular-wave generator; the other is an operational amplifier using bipolar and m.o.s. devices with a single supply.

The complementary transistor switch of card 6 is the transistor equivalent of unijunction and thyristor switches, but has the advantage of being faster than either as high-speed transistors can be used. Other cards show how the current capability of a c.m.o.s. inverter can be simply increased (card 8); how a cascode circuit is re-arranged to avoid the need for a biasing voltage (card 3); and card 5 gives performance and descriptions of m.o.s. and bipolar current mirrors—now an integral part of operational amplifiers.

High current-gain pairs—I 1 High current-gain pairs—II 2 Cascode amplifier 3 Long-tailed pair 4 Current mirrors 5 Complementary switching transistors 6 Complementary emitter-follower 7 CMOS circuits 8 Triples and mixed pairs 9 Pot pourri 10

## Transistor pairs

"Supplementary or complementary. opposites or composites?" asked Tweedledum. "Contrariwise", said Tweedledee. It is not the intention to rewrite "Alice ....' but it is not always clear what the advantages are in d.c.-coupled pair configurations-voltage gain plus impedance change, super current-gains, p-n-p simulation, thermal compensation or perhaps a current source as a stabilizing network.

The pièce de resistance might be the long-tailed pair or differential amplifier (where would the linear i.c. be without it?) which provides some satisfying amplifying properties—some possible variations for small-signal operation are shown in Fig. 1. For example, the connection shown in Fig. 1(a) converts a signal difference between two inputs to equal, but antiphase collector signals. Signals common to both inputs (e.g. superposed noise signals from the same source) are reduced without affecting the differential gain, which is that due to a single transistor.

The ability of this amplifier to prevent amplification of a common signal is called the common-mode rejection ratio, being the ratio of the differential voltage gain to the common mode gain, usually expressed in dB. The single-ended inputs of Fig. 1(b) and (e) produce an output at  $Tr_2$  collector, equivalent to  $Tr_1$  as an emitter-follower driving  $Tr_2$  as a common-base amplifier. In addition, the output of  $Tr_1$  is equal in magnitude but antiphase to that of  $Tr_2$ , on the assumption of constant total current (long-tail). Hence the differential output in Fig. 1(b) is twice the single-ended output. The voltage gains are approximately  $R/h_{ib}$  and  $R/2h_{ib}$  for Fig. 1(b) and (e) respectively, where  $h_{ib}$  is the effective input resistance of the common base configuration. Fig. 1(c) and (f) are not used as amplifiers, but are useful as a means of determining the gain to common mode signals. This should be much less than unity to give good rejection.

The common-emitter, common-base, connection shown in Fig. 2(a) is the cascode amplifier, where the first stage transistor  $Tr_i$  has current gain, and the second stage  $Tr_2$  has voltage gain. The advantages are a large gain-bandwidth product and also, for high voltage outputs, the high breakdown value of the common base transistor  $Tr_2$  is essential.

The common collector-common collector pair of Fig. 2(b) is the familiar

(e)

Darlington connection, where the effective current would be approximately  $\beta^2$ , if the transistors were identical. This would be true for short-circuit outputs, but a practical value for the  $\beta$  of each transistor



Fig. 1. Variations of the long-tailed pair include circuits with differential input, differential output (a), single-input, differential output (b) & (c), differential input, single output (d), and single input and output (e) & (f).











Fig. 2. Other ways of connecting two transistors are common-emitter, common-base i.e. cascode (a), common-collector pair i.e. Darlington (b), complementary pair with only one  $V_{be}$  drop (c), complementary pair with no  $V_{be}$  drop (d), symmetrical complementary pair i.e. class B amplifier (e), and the regenerative pair (f).

would be to specify the minimum value quoted by a manufacturer (assuming that discrete transistors are being discussed, because it must be remembered that super- $\beta$  packages are also available with defined current gains). The complementary connection of Fig. 2(c) provides only one  $V_{be}$  drop compared with Fig. 2(b), and that of (d) cancels the  $V_{be}$  drops, but offers some second-order residual offset between input and output, and requires a separate bias path (not shown) for  $Tr_i$  emitter current and  $Tr_2$  base current. Complementary versions may provide an advantage, in that the transistor choice for the first stage can be that with the highest current gain at low currents.

When opposite types of transistors are connected in complementary form, such as Fig. 2(e), we have the basis of a class-B amplifier. The circuit for each half-cycle of signal is essentially identical, and the load is driven from a low-impedance source, because each transistor acts as an emitter follower. An advantage is that interstage and output transformers are not necessary when such a circuit is used at audio frequencies.

The interconnection of n-p-n and p-n-p transistors as in Fig. 2(f) provides a regenerative switching action due to the positive feedback between collectors and bases. It is similar in operation to a unijunction transistor, and there are of course similarities with the thyristor. Switching speeds can be faster than a single unijunction device because highfrequency transistors can be employed. The same arrangement is the basis for a constant-current circuit (Circards set 6, card 5), while those sources which may be classed as current controlled tend to use the current-mirror—no self-respecting operational amplifier would be without one!

The complementary-symmetry m.o.s. pair has been an excellent addition to the pair family. Besides being the basis of several logic gates, it also has application in multivibrators and as a linear amplifier.

A close study of most linear integratedcircuits will reveal the use of two or more of the pairs mentioned, to provide a composite arrangement which offers advantages that no single transistor or pair can do.

### Set 20: Transistor pairs—

### High current-gain pairs—1

Darlington pair



Components Tr1: BFR41 Tr2: TIP3055

**Description and characteristics** A Darlington pair as shown above is a frequently used two-transistor circuit, the purpose being to obtain high gain through cascading two transistors. Because large currents are obtained in the second transistor one frequently finds this to be a power transistor and that the arrangement is used in many switching applications. As the circuit has only three

terminals it can be regarded as a single high-gain transistor. The basic action is that Ib produces, through Tr<sub>1</sub>, a large base drive for Tr<sub>2</sub>. The total Ic is the sum of the two collector currents although the contribution of Tr<sub>1</sub> to I<sub>c</sub> will be small if the gain of Tr<sub>2</sub> is large.

The characteristics obtained for the components quoted are as shown. From graph 1 we obtained an hoe of about 25mS and an hre of about 11,000. Graph 2 shows an hie of around 70. These figures are in line with theory (see reference). Graph 2 also shows the dependence of  $I_c$  on  $I_b$ , the two graphs being indistinguishable because of the scales chosen. For switching applications the value of  $V_{CE}$  is important.  $V_{CE}$ sat is defined as that  $V_{CE}$ corresponding to an arbitrarily



low  $I_c/I_b$  ratio, in this case 1,000 (graph 3). With higher  $I_{\rm c}/I_{\rm b}$  ratios, corresponding to lower I<sub>B</sub> values the value of  $V_{CE}$  will be lower. The corresponding graph of V<sub>BE</sub>sat. is shown in graph 4. Graph 5 shows the dependence of  $h_{FE}$  on  $I_c$ , the drop in  $h_{FE}$  at high I<sub>c</sub> being due to saturation and that at low Io being due to lack of base drive to  $Tr_1$ .



#### Complementary Darlington pair



#### Components Tr<sub>1</sub>: BFR81 Tr<sub>2</sub>: TIP3055

**Description and characteristics** The basic action of this circuit is the same as that of the normal Darlington pair. Note that the circuit in this format is acting as a single p-n-p transistor. An n-p-n version is shown below. It is the input transistor which indicates whether the combination is p-n-p or n-p-n but one can readily check the bias voltages.

Graph I shows the output characteristics: not eminently suitable for a small-signal amplifier, but as the circuit tends to be used for switching applications this is not too serious. We obtained for these components an hre of 8,500 approx and an hoe of greater than 0.1S.

Graph 2 shows I<sub>B</sub> and I<sub>c</sub> against  $V_{BE}$  and these again coincide because of the scale chosen. Note in this case that V<sub>BE</sub> during conduction is about 0.7V as opposed to 1.5V for the normal Darlington; this is due to the fact that there is only one junction between b and e. This shows up again in graph 4.

By the same token, the graph of VCEsat (graph 3) is much the same as that for the normal Darlington.

Graph 5 again has much the same shape as that for the normal Darlington though with a lower maximum value, which is in line with the reduced h<sub>FE</sub>.



Darlington has low saturation voltage, Electronics, 3 Oct., 1974.

### Reference

Ajdler, J. Transistor circuits, Electronic Engineering 1965, p.757. See also p.338 and p.112, same year.



Set 20: Transistor pairs-2

### High current-gain pairs—2



Components Tr<sub>1</sub>: BFR41 Tr<sub>2</sub>: TIP3055

Performance and description The Darlington circuits on card 1 are both super  $\beta$  circuits but have the additional charactistic that they only have three terminals and so can be regarded as a single transistor. The circuit shown above also has high gain but has four terminals, three of which can

be regarded as base, emitter and collector, the fourth being connected to a voltage which gives some control over the characteristics. The principal difference in performance of this circuit is that a low saturation voltage is obtained. From graph 1 we obtained an hre of 7,000 and an hoe of 25mS for  $V_{\rm B}$ =2.0V. Characteristics for V<sub>s</sub>=4.0V are virtually identical. Graph 3 shows the low Vcesat values obtained, slightly lower values being obtained for  $V_{s} = 4.0V.$ 

Graph 4 shows the expected VBEsat values around 1.5V since two base-emitter junctions are between b and e. Again slightly lower values are obtained for  $V_s = 4.0V$ . The graph of  $h_{FE}$  is shown in graph 5, the figures being in line with



Note that in this configuration Tr<sub>1</sub> will have supply voltage V<sub>s</sub> across it and will normally require a current-limiting resistor in its collector lead.





#### Components Tr<sub>1</sub>: BFR81 Tr<sub>2</sub>: TIP3055

### **Circuit description**

First given by Baxandall and Shallow, this circuit has a gain producing action similar to that of the Darlington's and the circuit overleaf. Base current to Tr<sub>1</sub> produces large base drive to Tr<sub>2</sub> thereby giving large gain. In this case, however, bias  $(I_1)$ must be provided for correct action. As there are more than three terminals it cannot be considered as identical to a single transistor, and it displays some characteristics

that one does not obtain with single transistors. In particular, one obtains collector current Ie in the absence of any  $V_1$ , this being due to the action of  $I_1$ . The characteristics shown were obtained with a  $V_{CE}$  of 1V and exhibit an overall gm of 6A/V, obtained at  $V_1$  of zero. The components used were chosen for comparison with the other super  $\beta$  pairs but certain features would dictate other choices. For example, the circuit simulates a p-n-p transistor whilst having an n-p-n output stage. In monolithic i.cs with lateral

transistors, n-p-n transistors have low gain so that Tr<sub>1</sub> would dictate the overall gain. As  $Tr_1$  has a very low voltage across it, viz  $V_{be}$  of  $Tr_2$ , then  $Tr_1$  can be made to have high gain-see MC1538R. When used as shown above right, the circuit exhibits extremely high output resistance which, allied to the high current gain, gives large voltage gain. The reason for this is that  $v_1 = v_3 + v_2 + v_{1n}$  and as  $v_3$  and  $v_2$  cancel, both being base-emitter voltages, then  $v_1 = v_{in}$  and the current in R<sub>L</sub> is thus defined by  $I_1$  and the current in R, no matter the

value of R<sub>L</sub>.

The circuit therefore, must have a large output resistance. Values as high as  $50M\Omega$  have been quoted.

#### Further reading

Baxandall, P. J. and Shallow. E. W. Constant-current source with unusually high internal resistance and good temperature stability; Electronics Letters vol. 2, p.351 1966. Jarger, R. C. A high output resistance source, IEEE Journal of Solid-State Circuits, Aug. 1974, pp.192-4.

### Set 20: Transistor pairs—3

### **Cascode amplifier**



### **Circuit description**

Tr<sub>1</sub> is a common-emitter amplifier which is feeding Tr, in common-base mode. The load on  $Tr_1$  is therefore a near short-circuit a.c.-wise. In addition, the current gain of  $Tr_2$  is near unity but its output impedance is high, due to the common-base mode of operation. The combination therefore shows the gain characteristics of a shortcircuited c.e. stage plus the output impedance characteristics of c.b. It is therefore ideally suited for the driving of tuned resonant loads or indeed of any high impedance load.

The basic equations are  $I_c = g_m V_{1n}$   $I_c \approx I_L$   $g_m = I_c/26mV$  $V_o = Z_L I_L$ 

The second equation assumes unity current gain through  $Tr_2$ and the last equation assumes that  $h_{ob}$  (for  $Tr_2$ ) is very low. The net result is

 $V_{\rm o}/V_{\rm in} = g_{\rm m} Z_{\rm L}$ 

Since  $I_e$  is almost linearly related to  $I_B$ , particularly if  $V_{CE}$  of  $Tr_1$  is constant which it is in this case, then  $g_m$  is controllable by  $R_B$ . The net result is the graph shown. For each result the circuit was retuned for resonance, the change in resonant frequency being around 1.5% over the complete range.

With  $R_B=1M\Omega$ , the resonant frequency was 924kHz and the bandwidth 18.5kHz giving a Q of 50. V<sub>R</sub> is not critical—it is only required to produce correct transistor action. In these results it was 3V. A 1.5-nF supply decoupling capacitor was necessary. Note that the graph represents Components  $Tr_1$ ,  $Tr_2$ : BFR41  $V_8$ : 10V Z: 660mH coil plus stray capacitance plus c.r.o. loading C: coupling capacitor  $R_B$ : see graph  $V_0$ : see graph

a range of voltage gain from 50dB to 71dB and that the range of R<sub>B</sub> is 20:1. The circuit above has two disadvantages viz, it is not a three-terminal device and also requires a biasing voltage. The above circuit suffers from neither of these disadvantages. The circuit is self-biasing provided only that I<sub>c</sub> is less than I<sub>d</sub> by an amount sufficient that the resulting  $V_{gs}$ keeps the bipolar transistor well out of saturation. Typically Vce would need to be in the range 0.2 to 1.0V (though a higher value would be preferable for higher gain). This means that the operating current is restricted to the Id of the f.e.t. with  $V_{gs}$  in the range of -0.2 to -1.0V. In practice the current is defined by some other requirement (drift, matching etc) and is often much less than Idss-the f.e.t. current with  $V_{gs}=0$ (which would give no gain from the bipolar anyway). The f.e.t. will frequently be required to operate near pinch-off. The circuit, above centre, is a less common form which loses the merit of low capacitive feedback. It does, however,



have the merit of increased current capability since the f.e.t. can operate with  $V_{gs} = 0V$ . The circuit right is of a form found as the input of some commercially available differential amplifiers. Each two-transistor block can be regarded as a single transistor and the circuit is then recognisable as a long-tailed pair. The tail uses the high output impedance of the pair to provide an excellent constant-current source the other two pairs being used for their high voltage gain.

Cross references Set 20, cards 4, 10.



### Set 20: Transistor pairs—4

### Long-tailed pair

(a) Differential indifferential out



**Circuit description** This emitter-coupled differential amplifier should have a large value for R4 to provide a high common-mode rejection ratio (c.m.r.r.). This implies that the differential output voltage between collectors for a common signal at the bases, is very small and ideally zero. In general, the output signal depends on the difference between the signals at the

**Useful relationships** Bipolar transistor collector current related to base-emitter voltage by  $I_{\rm C} \approx I_{\rm S} \exp q V_{\rm BE}/kT$  $\frac{dI_{\rm C}}{dV_{\rm BE}} = \frac{qI_{\rm C}}{kT} = g_{\rm m}$ At room temperature  $g_{\rm m} \approx \frac{I_{\rm C}}{26} ({\rm mA/V})$ 



Differential input signal can be considered as



transistor bases. Each transistor receives half the signal i.e. the gain to each output is half of that provided by a single transistor under the same conditions.

### Performance data Tr<sub>1</sub>, Tr<sub>2</sub>: matched pair from CA3086. R<sub>1</sub>, R<sub>2</sub>, R<sub>4</sub>: $100k\Omega$ .



Gain slope: typically 230. Variation in gain:  $\pm 1\%$ for several choices of CA3086. Reducing R<sub>1</sub>, R<sub>2</sub> by similar ratios will maintain slope at same order of magnitude.

 $v_{\rm C_1} = -g_{\rm m} \frac{v_{\rm in}}{2} R_{\rm C}$ 

 $v_{\rm C_2} = -g_{\rm m} \frac{-v_{\rm in}}{2} R_{\rm C}$ 



Vin: 10mV d.c. (20mV) Vout: 2.39V d.c. (4.79V) i.e. gain slope similar to (a)  $\Delta V_{out} = 240$ AVin

For single-ended output:  $V_{out}$  = voltage between X and ground.

 $\frac{\Delta V_{\rm out}}{\Delta V_{\rm in}} = \frac{10.28 \text{V} - 9.06 \text{V}}{20 \text{mV} - 10 \text{mV}} = 122.$ i.e. gain is halved.



 $\frac{\Delta V_{\rm xy}}{\Delta V_{\rm in}} = \frac{0.149 - 0.142}{2 - 1} = 0.007$ For single-ended output (terminal X to ground):  $\frac{\Delta V_{out}}{2} = 0.0045$  $\overline{\Delta V_{in}}$ 

c.m.r.r. improved by trimming R<sub>1</sub>, R<sub>2</sub> for zero balance. Figure of merit is c.m.r.r. or voltage gain for differential inputs divided by voltage gain for common-mode inputs that is 0.007/230.

theoretically zero-a basic advantage of the differential mode

$$V_{\rm BE} \approx \frac{kT}{q} \log \left(\frac{I_{\rm c}}{AT}\right)$$

A is dependent on transistor area. Two transistors on same chip, but different areas, provides a VBE variation given by

$$dV_{\rm BE} = \frac{kT}{q} \log \left(\frac{A_2}{A_1}\right)$$
  
Temperature drift is  
$$\frac{d\Delta V_{\rm BE}}{dt} = \frac{k}{q} \log \left(\frac{A_2}{A_1}\right) = \frac{\Delta V_{\rm BE}}{T}$$

At room temperature (300K) drift is  $3.3\mu V/deg C$  for each mV of initial offset.

#### Further reading

Thermal variation of the emitter-base voltage of bipolar transistors. Widlar, R. J. Proc. IEEE Jan. 1967. Limits of temperature drift in non-compensated d.c. amplifiers. IEEE Journal of Solid-State Circuits, Feb. 1970.

**Cross reference** Set 12, card 10



For identical transistors the collectors are at equal potentials for common-mode signals. Hence they can be considered connected together  $v_{\rm out} \approx \frac{-v_{\rm in}}{R_{\rm E}} \cdot \frac{R_{\rm c}}{2}$ 

Note: If R<sub>E</sub> replaced by a constant-current circuit,  $v_{out}(cm)$  can be  $\ll 1$ .

 $c.m.r.r. = \frac{-g_m \cdot R_c}{R_c/2R_E}$  $= \frac{-g_m R_E}{2}$ 

### **Temperature drift**

For a pair of identical transistors, balanced to provide zero difference between the base-emitter voltages of each, then the temperature drift is

Differential output given by  $v_0 = v_{C_1} - v_{C_2}$  $=-g_{\rm m}v_{\rm in}R_{\rm C}.$ Voltage gain Av  $= v_{\rm o}/v_{\rm in} = -g_{\rm m}R_{\rm C}$ =  $I_{\rm C}R_{\rm C}/26 = -V_{\rm C}/26.$ 



This single-ended input signal

can be considered as the algebraic sum of differential and common-mode signals: i.e.  $v_{in} = v_{in}/2 + v_{in}/2$ 

and  $0 = \frac{v_{in}}{2}$  $v_{in}/2$ common differential

mode mode If c.m.r.r. high, effect of common-mode signals negligible. Hence single-ended input equivalent to differential input.

Set 20: Transistor pairs-

### **Current mirrors**



### **Description (bipolar)**

The current mirror is an extremely useful two-transistor circuit extensively used as an integral part of monolithic operational amplifiers to define one current, the mirror current  $I_M$ , in terms of a reference current  $I_R$ . With identical transistors the base-emitter voltages are identical and if  $Tr_2$  has a high current  $I_R$  and  $I_M$ 

### Typical data

Tr<sub>1</sub>, Tr<sub>2</sub>: part of CA3086 I<sub>R</sub>: 1mA from commercial current generator,  $\pm 0.05\%$ I<sub>M</sub>: measured using 4-digit multimeter. Curves opposite show R<sub>0</sub> and R<sub>D</sub> as functions of +V.

are matched, to a first order. Any matched pair of n-p-n transistors may be used but those on a single chip are preferred; the variable temperature sensitivity of discrete device reduces reliability of the circuit. An important requirement in many applications of the current mirror is a high output resistance. The left-hand graph above shows that the static



(normalised)

1.0

output resistance  $(R_0 = V/I_M)$ increases with V for a given reference current value. However, the right-hand graph shows that whilst the dynamic output resistance  $(R_D = \delta V / \delta I_M)$ also increases with V, the far more rapid rise in  $R_0$  causes the ratio of dynamic to static output resistance to fall rapidly with increasing V. Hence, a compromise must be made between the values of  $R_0$  and  $R_{\rm D}$  to be used with a given value of  $I_{\rm M}$ .

For currents in the microamp range the output resistance of the current mirror can be increased by inserting a negative feedback resistor in the emitter lead of  $Tr_2$ .

For higher current requirements, transistors on the same chip can be connected in parallel to increase the junction areas.



### Description (m.o.s.)

2

Current mirror circuits can also be produced using m.o.s. transistors, the basic form using n-channel devices shown above left. Transistor Tr<sub>1</sub> is diode-connected performing as a transistor with 100%feedback. Thus its drain current is still controlled by its gatesource voltage VGS, i.e.  $I_{\rm D} \approx g_{\rm ss} V_{\rm gs}$  where  $g_{\rm ss}$  is the forward transconductance. Forcing a current IR into this diode-connected transistor causes  $V_{GS}$  to rise until a state of equilibrium is attained when Tr<sub>1</sub> sinks the reference current. The gate-source voltage of Tr<sub>2</sub>

is identical to that of  $Tr_1$ , due to the parallel connections. Hence if both have identical characteristics,  $Tr_2$  is also capable of sinking an identical current, the mirror current  $I_M$ . A reasonably good degree of matching can be obtained between the n-channel devices on a monolithic chip, the mirror current being typically within 10% of the reference current.

A current mirror using monolithic p-channel i.g.f.e.ts, shown centre left, provides better performance than the n-channel type due to the ability to provide much closer matching of the characteristics of p-channel devices. Such a circuit provides an  $I_M/I_R$  ratio which is to a first order independent of  $V_{DS}$ , as shown centre right. The graph above right shows that the normalized ratio of  $I_M/I_R$  is within 1% of its nominal value over a wide range of ambient temperatures and  $I_R$  values.

I<sub>R</sub>(mA)

### Further reading

٩ (m)

Jaeger, R. C. High output resistance current source, *IEEE Journal of Solid-State Circuits*, pp.192-4, August, 1974. RCA Solid State Databook SSD-201B, pp.183-8 and 213-25, 1974. Hart, B. L. Bidirectional Wireless World, vol. 76, 1970, pp.511-4. Hart, B. L. Bidirectional current source, *Electronics Eng.* pp.39-41, July, 1974.

AMBIENT TEMPERATURE (°C)

### **Cross references**

Set 3, card 9. Set 6, card 4. Set 9, card 5. Set 10, cards 1, 3, 7. Set 12, cards 4, 7. Set 15, card 6. Set 16, card 1.

### Complementary switching transistors



#### Circuit description The arrangement of the complementary pair of transistors, with or without the resistor R, or connected between two other points, is a frequently used combination (see cross refs.). With the resistor R as shown, the circuit acts over a portion of its I-V characteristics as a negative resistance, of value -Rapproximately. Graphs obtained are shown for values of 1k and $10k\Omega$ .

### Components Tr<sub>1</sub>: BFR81, Tr<sub>2</sub>: BFR41

 $\mathbf{R}: \mathbf{1}\mathbf{k}, \mathbf{10}\mathbf{k}\boldsymbol{\Omega}$ 

### Performance

Slope in negative resistance region  $0.75k\Omega$  and  $8k\Omega$  for  $R=1k\Omega$  10k $\Omega$  respectively (see graph).

Initially with low I, V may be considered as  $V_{EB_1} - V_R + V_{BE_2}$ . As V<sub>R</sub> is low initially V is the sum of the two exponential emitter-base voltages. As I increases these two voltages tend to 0.6V but V<sub>R</sub> continues to rise and because of the minus sign V starts to fall, at a value slightly less than 1.2V. The fall continues (negative resistance region) until both transistors saturate. At this point we can no longer assume that all the current is passing



through both collectors and R and we are best to view V as being  $V_{\rm EC_1} + V_{\rm R} + V_{\rm CE_3}$ . Since the two transistor voltages are relatively fixed, V then starts rising again. The value of the trough is given by  $V_{\rm EC_1}$ sat +  $V_{\rm BE_2}$ sat  $\approx V_{\rm BE_2}$ sat. The voltage range of the circuit is readily increased by the modification shown opposite with Zener diodes appropriate to the application. These circuits are described as being open-circuit stable, i.e. for any current drive there is a unique voltage. The dual of this is the voltage driven, short-circuit-stable device. An

f.e.t. realization of this is shown with the corresponding characteristic (see Further reading).

Set 20: Transistor pairs—6





#### **Circuit description**

The circuit shown overleaf, with  $R = \infty$ , is the basis of simulated thyristors, silicon controlled switches and simulated unijunction transistors. The action, with  $R = \infty$ , can be deduced from the V-I graph shown, bearing in mind that the negative resistance slope is now  $-\infty$ . Alternatively, the circuit above can be considered as follows. Initially Tr<sub>1</sub> and Tr<sub>2</sub> are in the non-conducting state. A positive voltage of about 0.7V on the base of Tr<sub>2</sub> causes that transistor to conduct and lower its collector voltage. This causes

Tr<sub>1</sub> to conduct and providing Tr<sub>1</sub> and Tr<sub>2</sub> produce sufficient current through R<sub>2</sub> and R<sub>1</sub> respectively to keep the transistors conducting, then the action is self-sustaining with a voltage of approximately one  $V_{BE}$  across the circuit. The circuit can thus be used as a switch, triggered by a suitable voltage at either base. With  $R_1 = R_2 = 1 k \Omega$  a value of 1.3mA for I was found to be the minimum which would maintain conduction. This value is as expected as the current I will split fairly evenly between  $Tr_1$  and  $R_1$  and  $Tr_2$ and R<sub>2</sub>. Since approximately

0.65V is required at the base of each transistor to maintain conduction, it will be provided by 0.65mA in each path. For high-speed switching it is important to prevent the transistors from saturating. Addition of the anti-saturation diodes shown prevents the collector voltages from dropping below V<sub>BE</sub>. The base lead diodes are not essential. Increasing R1 increases the trigger voltage necessary and reduces the effect of trigger point transients. Transients in the supply line give rise to false triggering due to rate effect. This can be reduced as R<sub>2</sub> is reduced although this increases the holding current necessary.

#### Further reading

G.E. Transistor Manual. Sharma, S. M. Currentcontrolled negative resistance circuit. *Int. J. Electronics*, 1974, vol. 37, pp.209-18. Negative resistance shown in dual f.e.t. device, *Electronics*, April 18, 1974, p.5E. Cross references Set 10, card 5. Set 8, card 3. Set 6, card 9. Set 3, card 4.

### Set 20: Transistor pairs—7

### **Complementary emitter-follower**



**Circuit description** The basic circuit comprises a complementary n-p-n/p-n-p pair operating under class-C bias, and is the basis for many audio power/amplifiers. When a positive-going input signal exceeds about 0.7V, transistor Tr<sub>1</sub> will turn on, increasing its collector current which develops a voltage across the load R<sub>2</sub>, but with a voltage gain of less than unity. At the same time Tr<sub>2</sub> is biased off. Similarly, a negative-going input signal turns Tr<sub>2</sub> on and Tr<sub>1</sub> off, and the circuit thus provides bidirectional currents through the emitter load. The base-emitter diode characteristic is non-linear at low voltages. resulting in cross-over distortion (approximately  $2V_{BE}$ ) across the load. The resulting distortion on a sine wave input is shown in Fig. 1. Without an additional bias network, the effect of this distortion can be minimized by ensuring that  $V_{\rm S} \gg V_{\rm BE}$ . Using both positive and negative power supplies permits operation down to zero frequency. For a single supply a capacitor is required in series with the load, to provide the base and collector currents of Tr<sub>2</sub> during negative-going input signals.

### **Component** changes

 $R_2$ : Range from 100 to 1k $\Omega$ . Variation in gain minimal. Frequency: Up to 30kHz, little difference from lower frequency operation. Variation of mean current with input level in Fig. 2. R<sub>1</sub>: Chosen to suppress parasitic oscillations. Alternative may be to keep interconnections very short to minimize series inductance.

### Typical data

 $V_s: \pm 6V$ T<sub>1</sub>: BFR41, T<sub>2</sub>: BFR81  $R_1$ : 330 $\Omega$ ,  $R_2$ : 1k $\Omega$ Signal frequency: 1kHz Fundamental/3rd harmonic output/input shown on graphs

Effect of bias (Figs. 5 & 6) **R**<sub>B</sub> varied until transistors just on the point of conduction. Graph shows 3rd harmonic distortion optimized by R<sub>B</sub>. The above is a basic method of biasing to ensure class B or AB operation. Diodes D<sub>1</sub>, D<sub>3</sub> may be chosen to achieve temperature independence of quiescent current. Where increased current gain is needed for high power outputs, the Darlington pair of Fig. 8 (a) and compound emitter followers of (b) and (c) are useful, but may provide more difficult biasing problems.

Line driver (Fig. 7) Driving 50- $\Omega$  cable with line capacitance slows pulse edges.

Fig. 5

Fig. 6

Fig. 7



Improved using complementary pair. Typical fall time  $\leq 100$ ns but dependent on simulated cable capacitance ( $R_0$ : 50 $\Omega$ , C: 3nF, t<sub>F</sub>: 75ns).

### Further reading

Williams, P. Voltage following, Wireless World, Sept., 1968. Williams, G. E. Practical transistor circuit design and analysis, McGraw-Hill 1973. Markus, T. Electronic circuits manual, McGraw-Hill.

**Cross references** Set 7, cards 1, 3. Set 20, card 1.



### Set 20: Transistor pairs—8

### **CMOS** circuits



### Description

The c.m.o.s. inverter shown above comprises a p-type and n-type enhancement mode m.o.s. transistor on the same chip.  $V_{DD} - V_{SS}$  may be in the range 3 to 15V. The pair is useful in digital circuits because of well-defined threshold that  $V_{in}$  must exceed before the device turns on.

n-type: V<sub>GS</sub> positive for ON p-type: V<sub>GS</sub> negative for ON Where the output has to sink or source current, the pair can be envisaged as the series switches. Fig. 1 is a source condition, obtained for  $V_{in} =$  $V_{ss}$  and  $V_{out} = V_{DD} - I_d R_o$ where R<sub>o</sub> is the output resistance of p-transistor in the on state. For  $V_{in} = V_{DD}$ , the n-type sinks current for Fig. 2;  $V_{out} = I_d R_o$ .

Basic i.c. package CD4007 contains one inverter and two complementary pairs with the drains unconnected. This permits a variety of interconnections: n-types are paralleled to increase sink current capability, Fig. 3. p-types are paralleled to increase source current capability, Fig. 4. Fig. 5 is a dual bi-directional transmission gate where the two outputs and input may be interchanged for two inputs and one output. Note that the position of the transistors in

the middle pair have been reversed. The CD4049 package contains

six inverters with current drive capability almost an order of magnitude greater than basic package. Parallel connection for increasing current sinking indicates typical current sharing for d.c. condition, Fig. 6.



**CD4016** i.c. package contains four analogue switches (transmission gates). Each switch comprises an inverter and a parallel pair of n- and pchannel transistors, Fig. 7. For a threshold of approximately 2V in each

control

Fig. 7

R<sub>2</sub>3

Fig. 10

vell

channel and control voltage 10V, the gate control voltage of the p-m.o.s. is 0V. If  $V_{in}$ > 8V, the n-channel will be open, but the gate-source voltage of the p-m.o.s. is -8V (greater than threshold) and signal is switched through.

Fig. 9

Fig. 8

Vnn

٥Vse



A.C. amplifier (Fig. 9) Best linearity and voltage swing if  $V_{out} = V_{DD}/2$  and this is provided by the resistive connections (Figs. 9 & 10). Supply drain  $\approx 2\text{mA}$  for 10V supply.

#### Further reading

RCA Solid-State Databook Series SSD—203A 1973. Design Ideas with COS/MOS New Electronics, April 30, 1974. I.C. op-amp has CMOS output Electronics, Sept. 19, 1974. Fitchen, F. C. and Ellerbruch, V. G. Linear operation of the MOSFFT complementing pair, IEEE Journal of Solid-State Circuits, Dec., 1971.

Cross references Set 8, card 1. Set 10, card 7. Set 11, card 5.

Set 12, card 1.

### **Triples & mixed pairs**

The principle of using transistors in pairs can often be usefully extended to the use of devices in triples, the resulting equivalent transistor having a current gain equal to the product of those of the individual transistors. In all such arrangements, two of which are shown above, the equivalent transistor has the same "polarity" as the input transistor.

The three n-p-n transistors in Fig. 1 act as a compound emitter-follower having a current gain of approximately  $h_{te}^{a}$ , for large  $h_{te}$ , and is useful for driving currents of several amps from a driver stage delivering less than a milliamp. The complementary transistor triple in Fig. 2 is useful in voltage regulators as a low-dissipation series element,

the  $V_{BE}$  and  $V_{CE}$  values of the equivalent transistor being the lowest possible for a triple. In this arrangement the temperature coefficient of only one transistor affects the output. The operating current in the input transistor will be very small, reducing its current gain. This effect can remove much of the benefit of using a triple instead of a pair. By using resistors as shown in Fig. 3 the operating currents in the earlier stages are increased and stabilized.

Thinking of the various pairs of devices discussed on other cards as elementary building blocks can prove a powerful method of developing more complicated circuit functions. For example, a long-tailed pair  $Tr_1$ ,  $Tr_2$  and  $Tr_3$  may be used to drive a current mirror  $Tr_4$ and  $Tr_5$  as shown in Fig. 4 to produce a waveform generator which provides a symmetrical

Typical values. Supply  $\pm 10V$ , V<sub>R</sub> 0V, R<sub>1</sub> 5k $\Omega$ , R<sub>2</sub> 2·2k $\Omega$ , R<sub>8</sub> 4·7k $\Omega$ , C<sub>1</sub> 33nF, D<sub>1</sub> 1N914, D<sub>2</sub> 1N5234, D<sub>3</sub>, D<sub>4</sub> HP5082 – 2800, Tr<sub>1</sub>, Tr<sub>2</sub>, Tr<sub>3</sub> 2N3546, Tr<sub>4</sub>, Tr<sub>5</sub>  $\frac{1}{2} \times$  SL301-A. triangular output when driven by input pulses. This circuit uses the long-tailed pair to



switch a defined current into two paths; which combined with a closely-matched current mirror permits the capacitor to be charged and discharged at the same rate when the charging current is varied. This concept of mixing elementary pairs of devices can be developed to build a single-supply operational amplifier using bipolar and m.o.s. transistor pairs, as shown in Fig. 5, having a unity-gain bandwidth of about 10MHz. The operational amplifier, which requires two CA3600E and one CA3046 packages, has three stages. The differential input stage uses two p-channel m.o.s. transistors Tr<sub>3</sub>, Tr<sub>4</sub>, the second stage uses an n-p-n bipolar transistor  $Tr_7$  and the output stage is a complementary m.o.s. transistor pair Tr<sub>8</sub>, Tr<sub>8</sub>. The zener network, using two diode-connected transistors  $D_1$ ,  $D_2$  of the CA3046, feeds a p-channel current mirror Tr<sub>1</sub>,  $Tr_{2}$  that establishes a 400 $\mu$ A constant current in the input stage. This differential-input amplifier is loaded by four resistors,  $R_1$  to  $R_4$ , and a

bipolar current mirror, Tr5, Tr6, to provide optimum balance, any voltage offset being nulled with the potentiometer  $R_{s}$ . The current in the second stage, determined by  $R_{\delta}$ , is adjusted to equal the 400- $\mu$ A first-stage current to provide similar negative and positive slew rates. The output stage is biased as a class-A amplifier by  $R_5$  and may be driven to within a few millivolts of the ground rail. The overall voltage gain varies inversely with the load resistance which. as with a monolithic operational amplifier, would have a value

of about  $2k\Omega$ . Compensation requires inclusion of the feedback capacitor  $C_1$ , with  $C_2$ added when using the operational amplifier as a unity-gain follower. In this configuration,  $R_9$  and  $C_3$ should be added to avoid the possibility of latch up and D<sub>a</sub> and D<sub>4</sub> added to the inputs to prevent negative-going input signals exceeding about 700mV which could also cause latch up. Typical values are: V+15V;  $\begin{array}{l} R_{1}, R_{2}, R_{3}, R_{4} \ 200\Omega \pm 1\%; \\ R_{5} \ 20k\Omega \pm 1\%; \ R_{6} \ 11k\Omega \pm 1\%; \end{array}$  $R_{7}$  7.5k $\Omega \pm 1\%$ ;  $R_{8}$  10k $\Omega$ ;  $R_{9} 1k\Omega; C_{1} 39pF; C_{2} 300pF;$ C<sub>3</sub> 150pF; D<sub>3</sub>, D<sub>4</sub> 1N914.

#### **Further reading**

Williams, P. Voltage Following, Wireless World, vol. 74, 1968, pp. 295-8.
Nowicki, J. R. Compound transistor connections, *Electronic Engineering*, September, 1971, p.63.
Burwen, E. High-gain triple Darlington has low saturation voltage, *Electronics*, Oct. 3, 1974.



### Set 20: Transistor pairs-10

### Pot pourri

## n-n-p/p-n-p simulation at high voltage

Optical couplers are used to provide fast high voltage switching with a simulated complementary pair. Transistors  $Tr_1$  and  $Tr_2$  are high-voltage n-p-n types:  $Tr_1$  is on when  $Tr_2$ is off and vice versa. With  $Tr_1$ off, C charges to  $V_Z$ , storing charge. This is used to turn  $Tr_1$ on fast when the optical coupler operates. Optical coupler provides the

polarity inversion when  $Tr_4$ and  $Tr_5$  are driven by 5V pulses.

Tr<sub>3</sub>, Tr<sub>4</sub>, Tr<sub>5</sub>: 2N2222 R<sub>1</sub>: 270kΩ (10W), R<sub>3</sub>: 1kΩ R<sub>3</sub>: 100Ω, R<sub>4</sub>: 47Ω R<sub>5</sub>

R<sub>6</sub>: 680Ω, C: 10nF

D<sub>1</sub>: 6.8V zener, OC1 : Monsanto MCO1

Rise and fall times of around  $2\mu$ s are claimed for components used.

### Unijunction from bipolar pair



Tr<sub>1</sub>: 2N4126, Tr<sub>2</sub>: 2N4124 R<sub>1</sub>, R<sub>2</sub>:  $4.7k\Omega$ , V: +3 to 25V Because the collector current of Tr<sub>1</sub> is the base current of Tr<sub>2</sub> and vice versa, then any change of current provides a positive feedback action. B<sub>1</sub>, B<sub>2</sub> and E are the equivalent unijunction terminals. When E is open-circuit

 $V_{\rm x} = R_2 V/(R_1 + R_2).$ 

When the potential at E exceeds  $V_{BB}$  and  $V_X$ , both transistors saturate, and  $R_a$  is short-circuited. This condition is maintained for a potential at E down to  $V_{EB} + V_{CESA}$  of  $Tr_2$ . Switching speed depends on maximum frequency of of operation of the transistors. For  $Tr_1$ ,  $Tr_2$  silicon drift of  $V_{EB} < 3mV/deg C$ .



#### Shunt-series d.c. feedback pair



Current fed back via  $R_f$  is proportional to  $Tr_2$  collector current, and thus the circuit provides current-shunt negative feedback, which primarily controls the overall current-gain.

Effective current gain is  $A_i/(1+\beta A_i)$  where  $A_i$  is the current gain of the two stages, with  $R_t$  connected between  $Tr_1$  base and ground.  $\beta \approx R_E/R_t$ . For  $\beta < 0.1$ , input resistance is  $R_i/(1+\beta A_i)$  where  $R_i$  is input resistance without feedback.

Voltage gain  $\approx \frac{R_t}{R_E} \cdot \frac{R_2}{R_S}$ 

where  $R_s$  is source resistance. Voltage gain can be increased by by-passing  $R_E$ , or by-passing the mid-point of the feedback resistor  $R_f$ . Series-shunt d.c. feedback pair



In this connection, the voltage gain is now mainly affected by the feedback, and the input resistance is increased. Voltage gain is  $A_v/(1+\beta A_v) \approx 1/\beta$  if  $A_v$  is large.  $A_v$  is the product of the gains of each stage, with  $R_F$  connected from  $Tr_2$ collector to ground, and  $\beta = R_E/R_F$ . Input resistance is  $R_1(1+\beta A_v)$ , where  $R_1$  is effective input resistance of the first stage. Note: Biasing networks for those CE-CE amplifiers are not shown.

### Common emitter pairs with composite transistors



Tr<sub>1</sub>, Tr<sub>2</sub>: 2N4355 V<sub>cc</sub>: 15 to 30V Voltage gain  $\approx$  100 Cut-off frequency  $\approx$  10kHz Input resistance > 200k $\Omega$ By sacrificing voltage gain, a high input impedance can be obtained with the circuit shown. below.



Tr<sub>1</sub>, Tr<sub>2</sub>: 2N4355 V<sub>CC</sub>: +10 to 30V Voltage gain  $\approx$  10 Input impedance: 5M $\Omega$ 

Further reading for circuits 1-6 1. Mitchell, P. & Robbins, K. Simulating an npn/pnp pair for high voltage switching. *Electronics*, Aug. 22, 1974. 2. Shyne, N. A. Bipolar pair simulates unijunction, *Electronics*, Jan. 24, 1974. 3, 5, 6. Cowles, L. G. Transistor circuit design, Prentice-Hall, 1972. 4. Millman & Halkias, Electronic Devices and

Electronic Devices and Circuits, McGraw-Hill, 1967, p.502.

Cross references Set 20, cards 6, 7. Set 12, card 9.

1. The common-base and common-collector pair used in the d.c. feedback pair configuration has a low input impedance, low output impedance and wide bandwidth. Using high voltage transistors it is suitable as a pulse-amplifier to produce very fast positive edges even into capacitive loads (a complementary class B output stage would be needed to make the falling edge equally fast). It lacks d.c. stability but can be combined with a high speed op-amp as shown to give an output slew rate claimed in



the reference to be  $200V/\mu s$ and full power output of 60V peak to peak at 1MHz. The overall feedback brings the output stage within the loop reducing the offset and drift to that of the op-amp. The latter provides current drive

2. A recent paper has considered the complex behaviour of apparently simple m.o.s. transistors. If the substrate connection is available, then for a p-channel enhancement mode f.e.t. that substrate also acts as the base of a p-n-p transistors with source and drain representing the emitter and collector respectively. To a first-order approximation the devices are



shown to be independent when conducting with the total current being equal to the sum of the individual currents. The m.o.s. action is controlled by a square-law transconductance equation, while the bipolar transistor has a collector current proportional to the base current to a first order of magnitude (low current gain in this example). Thus the waveform in a load resistor connected to a negative supply could be varied by applying an input signal separately or into the c.b. stage but does not need to supply a large voltage swing reducing the demands on its slew rate. The reference describes other more complex circuits capable of extending the small-signal bandwidth to 20MHz and slew-rate to 1200V/ $\mu$ s or even higher using special amplifiers. Precautions on layout and decoupling are given.

### Reference

Bunin, H. Low-cost v.c.o. driver amplifiers really perform if designed right, *EDN*, 1974, Oct. 5, pp. 51-5.

jointly to the gate and substrate terminals. This parallel combination should also be amenable to other waveform processing methods by combining feedback networks with it.

### Reference

Hart, B. L. and Barker, R. W. J. First-order theory of m.o.s.f.e.t. hybrid-mode operation, *Int.J. Electronics*, 1975, vol. 38, pp. 625-30.

3. This circuit is not strictly a two-transistor arrangement but illustrates how a circuit can be partitioned to reveal familiar pairs. It is a wide band precision rectifier that uses a high output impedance rather than negative feedback to overcome the diode non-linearities. A Darlington pair composed of Tr<sub>2</sub>, Tr<sub>3</sub> is modified to remove the collector of Tr<sub>2</sub> from the circuit output, by returning it to ground. This leaves Tr<sub>3</sub> providing a low-capacitance constant-current sink. D.c. negative feedback defines the operating points (with negligible loading because



the current gain is high allowing high-value resistors. The  $43\Omega$  resistor defines the circuit transconductance and Schottky diodes maintain

the bandwidth with minimum p.d. and hence non-linearity at low levels. Where large electrolytics are needed to extend low frequency response (20Hz is quoted in reference) then parallel tantalum capacitors are added to maintain low impedance at high frequencies. The source has to be capable of carrying the base current of  $Tr_1$ . The reference indicates the upper cut-off frequency as 80MHz with good accuracy to 10MHz, and good linearity and repeatability for 200mV r.m.s. full-scale.

### Reference

Knott, K. F. Sensitive wideband linear a.c.-d.c. converter, *Proc.IEE*, 1975, vol. 122, pp. 249-52. notes

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circuit designs

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