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p. 63

AUDIO & HI-FI

43 PROJECT: Audio-video processor
an ELV design

COMPONENTS

55 PROJECT: GAL programmer
Design by M. Nosswitz

COMPUTERS & MICROPROCESSORS

14 PROJECT: Multi-purpose Z80 card - Part 1
Design by A. Rietjens

GENERAL INTEREST

26 Elements of passive electronic components - Part 1
by Steve Knight, B.Sc.
60 New logic symbols
by our technical staff

POWER SUPPLIES & BATTERY CHARGERS

21 PROJECT: Compact mains power supply
Design by A. Robnucker

RADIO, TELEVISION & COMMUNICATIONS

29 The NICAM system
by our Technical Editor
34 PROJECT: NICAM decoder
Design by Rob Krijgsman, PE1CHY
43 PROJECT: Audio-video processor
an ELV design
50 PROJECT: FM tuner - Part 3
Design by H. Reelsen

SCIENCE & TECHNOLOGY

46 Digital short-range radio
by Brian P. McArdle

TEST & MEASUREMENT

24 PROJECT: 1.3 GHz prescaler
Design by P. Esser

MISCELLANEOUS INFORMATION

Electronics scene 11-13; Events 13; Readers’ corner 64;
Letters 64; Switchboard 65; Readers’ services 66;
Terms of Business 68; Index of advertisers 74
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GO-FASTER ELECTRONS

DEMANDS from the world's semiconductor industries have led to a search for a category of new materials in which electrons can travel faster.

Silicon is widely used at present, but increasingly sophisticated modern need, such as real-time speed and image processing, call for other semiconductor materials that have to be made artificially.

At Nottingham University in the English midlands, researchers are using Molecular Beam Epitaxy (MBE)—an advanced technique for growing such materials—and have produced very high-quality samples, including two-dimensional electron gases and resonant tunnelling structures.

In MBE, extremely thin semiconducting layers—of atomic dimensions—are formed in an ultra-high-vacuum environment containing about 1000 billion times fewer atoms per cubic metre than the earth's normal atmosphere. Beams of atoms or molecules evaporate from the surface of heated materials in crucibles and follow straight-line trajectories through the vacuum until they stick to the surface of a substrate or the walls of the vacuum chamber. A continuous liquid nitrogen flow keeps surfaces in the vacuum chamber cold, thus trapping stray gases and helping to maintain the vacuum. Rapid opening and closing of shutters in front of each source allows the composition of the grown layer to be controlled—the growth processes are usually controlled by a computer. Subsequent to the completion of the required structures, various lithographic processes are used to form different devices for experimentation.

The research group, known as NUMBERS (Nottingham University Molecular Beam Epitaxy Research Syndicate), funded by Britain's Science & Engineering Research Council, has many collaborative projects with laboratories in the UK and overseas. Among these is one with the National Physical Laboratory, which has resulted in the development of a quantum hall standard resistor, a sophisticated device fabricated from the semiconductor material produced by MBE. University of Nottingham, Department of Physics, University Park, Nottingham NG7 2RD, England.

**TYPETALK ENABLES DEAF TO USE THE TELEPHONE**

After successful trials, a new national telephone relay service, named Typetalk, which allows deaf and speech-impaired people to communicate by text communication with voice users, has been launched in Britain. The trials were carried out by the Royal National Institute for the Deaf with the financial backing of British Telecom and the National Institute for the Deaf with the financial support of British Telecom, and resulted in a system that is capable of handling 10,000 calls per month.

Typetalk, which has taken over from the Telegraph Exchange for the Deaf in London, has a theoretical capacity of 10,000 customers. Since in Britain there are 200,000 potential users of text services deaf and hearing-impaired people who are unable to use a voice telephone, a second exchange is already being planned.

**NEW MINIATURE CERAMIC DISC CAPACITORS**

The Murata DD003 range of very small ceramic disc capacitors is ideal for applications such as coupling and decoupling in microprocessors. It allows high density population on PCBs and provides engineers with a highly suitable alternative to 100 V plate capacitors. Circuit board layout is made simpler, not only because of the size of the devices, but also the availability of a broad range of capacitance values. Capacitors
with values between 1 pF and 22 000 pF are available in NPO, SL, 2B4 and 2F4 dielectrics. In the 12 V version, the maximum capacitance value is 47 000 pF.

ESD Mercator, South Denes, Great Yarmouth NR30 3PX, England

**BRIEFCASE OSCILLOSCOPES**

**THE PORTABLE oscilloscopes in the LBO -315 range, available from Thurlby-Thandar, provide a fully professional specification in spite of their compact size (230x75x290 mm or 9x3 x11 1/2 in).**

The large 95 mm rectangular CRT has 8x10 divisions (1 div = 6.35 mm) and either 2 kV or 12 kV acceleration potential dependent on model. Models with bandwidths of 60MHz and 20MHz are available. Sensitivity may be varied between 5 mV and 5 V/div.

Thurlby Thandar Ltd, Glebe Road, Huntingdon PE18 7DX, England.

**NEW DATA TERMINAL**

INCREASINGLY since the Second World War, interest in the data modes on the amateur radio bands has been growing; Over the past decade, this interest has shifted from teleprinters to computer-based systems.

A new terminal called DATATERM has now become available, which will handle RTTY, AMTOR, CW, SSTV and FAX (on the last two receive only). The terminal is designed as an expansion card to fit the standard half slot of a PC. Power is derived from the expansion slot, so no external power supply is required. The DATATERM consists of a pair of pre-limiting active filters and a pair of filters later in the signal path, which gives excellent performance under crowded band conditions. There is a fitted AFSK generator on board, as well as CW keying and PTT switching facilities. It is possible to switch filter bandwidths for the reception of amateur and commercial signals.

Melmar Products, 2 Salters Court, Gosforth, Newcastle NE3 5BH, England.

**LASER DIODE DRIVER IC**

AVANTEK, a subsidiary of Hewlett-Packard has introduced a new IC to drive solid-state lasers or light-emitting diodes in digital fiber optic systems operating at speeds of up to 1.5 Gb/s. Designated IDA-07318, this silicon bipolar IC may be used in terminals for data communications such as FDDI (100 Mb/s), FC (800 Mb/s) and serial HIPPI (800 Mb/s); and telecommunications, such as SONET (OC-3 through OC-24). It also has applications in instrumentation and communications requiring high-speed current modulation capabilities.

Avantek, Inc., 481 Cottonwood Drive, Milpitas, CA 95035-7492, USA.

WORLD FIRST AS UK ELECTRONIC FIRM ELIMINATES CFCS

BRITAIN'S Northern Telecom has become the world's first major electronics company to eliminate ozone-depleting chlorofluorocarbons (CFCS) from its operations, years ahead of the mandate set by the international Montreal Protocol agreement which calls for elimination of CFC and other ozone-depleting substances by the year 2000.

The company estimates that the alternative it has developed will prevent nearly 9000 tonnes of CFCs being released by its plants into the atmosphere over the next eight years. The new processes should also save more than $50 million during that time in CFC purchasing costs and other expenses.

Northern Telecom, c/o AB Consultants, 120 Wootton Street, London SE1 8LY, England.

**FILTERED CONNECTORS**

IMPROVED manufacturing techniques and a re-design of ITT Cannon's popular family of EMC filtered connectors has resulted in a new, expanded range, the D*JK. Included in the new family are straight-pin versions for PCB mounting, as well as the widely used right-angle types. The straight-pin versions are available in 9, 15, 25, 37 and 50 way plugs and sockets.

The D*JK versions are fully intermateable and intermountable with standard D subminiature connectors, as well as being compatable with all D-sub accessories. This means a low-cost solution to meeting the mandatory EMC emission requirements can be achieved without design changes, either in internal circuitry or the PCB-connector interface. EMC emission filtering is achieved by high-quality radial ceramic capacitors soldered between machined contacts and the backshell to provide excellent attenuation performance. Grounding fingers on plugs are independently sprung.

**EIGHT-WAY DISTRIBUTION SOCKET**

ON SCREEN WEATHER PICTURES BY TELEPHONE

THE 'Carousel' eight-gang trailing socket from Maplin Electronics has no equal. The socket distribution system is an unusual eight-sided package manufactured to BS5750 Part 2, and complies with BS 1363A (1984) and BS 6500. The 170x170x60 mm unit is made of high-grade styrene for strength, durability and fire resistance.

Maplin Electronics, P.O. Box 3, Rayleigh SS6 8JR, England.

A prototype system of sending weather satellic pictures over telephone lines into homes and offices has been developed and successfully tested by a group of communications engineers at the Polytechnic South West. The system needs only a telephone, access to a personal computer and a modem.

To receive a colour picture of the weather situation as it was some 20 minutes previously, the user logs in and types in the Polytechnic's telephone number. His PC then dials the main computer at the Polytechnic and every 30 minutes, for the price of a 2-minute telephone call, a high-quality, high-resolution, easily-interpreted colour image appears on the computer screen. To end the transmissions, the user logs out.

The picture, beamed from a satellite stationed 32 000 km above the earth, shows the European weather situation. After several transmission have been received, the computer automatically fades between images, showing cloud movement like sophisticated systems do on television weather forecasts.

A pilot public system for south-west England will be launched shortly and commercial aspects for other areas of the UK and overseas will be investigated.

Dr Graham Wade, School of Electronic Communication and Electrical Engineering, Polytechnic South West, Drake's Circus,
NEW LABS TO BOOST SUPERCONDUCTOR RESEARCH

The centre producing the world's highest-temperature superconductor, a material that loses all resistance to electrical current at a temperature of -145 °C (128 K), has just opened purpose-built laboratories for further research in this promising field. Superconductivity is an exciting field that has enormous potential as well as being a challenge for scientists.

The Interdisciplinary Research Centre (IRC) in Superconductivity is a unit of Cambridge University. It concentrates on research into materials that have superconducting properties at temperatures generally greater than 90 K (-183 °C). First discovered in 1986, these can be used in a wide range of applications to eliminate the energy losses caused by electrical resistance.

Superconductivity has now been observed in five fundamentally different types of material. While the IRC plays a central role in researching fundamental material properties as well as in thin films and processing technology, one of its prime aims is to speed up the application of oxide superconductors in industry. One of the IRC's researchers is Nobel Laureate Sir Nevill Mott, whose immense knowledge of solid-state physics and experience is now applied to the wider programme of superconductivity research.

Last year a group of researchers headed by Britain's AECL Technology succeeded in producing superconducting wires tens of metres long, made from ceramic powders, and now aims to produce wires several hundreds of metres long.

University of Cambridge, Old Schools, Cambridge CB2 1TN, England.

MAGNETIC FIELD METER
MINIATURE field-strength meter from Magtronics introduces a new concept in the measurement of magnetic field.

Applications for the new meter include monitoring the strength of permanent magnets, testing loudspeakers and f.c. motors, checking computer disks for stray magnetic fields, or any material incorporating magnetically active components.

Three models cover the ranges 0-0.02T, 0-0.2 T and 0-2 T on a 3½ digit LCD display. A peak hold facility is included for determining maximum field strength.

A polarity indicator with a sensitivity of 0-5 mT is also included in the new range.

Redcliffe Magtronics, 20 Clothier Road, Brislington, Bristol BS4 5PS, England.

AWARD WINNING TRAINING

E & L Instruments has won the Department of Trade and National Westminster Bank 1991 corporate shop floor exporter of the year award, as well as the gratitude of many firms that have been enabled to undertake their training programmes in Information technology without wasting valuable production time and resources.

The company designs systems and makes electronic equipment for business education and training that allows training to be carried out without interfering with other equally important operations. In the photograph, one of the firm's electronics technicians is seen testing modules for its OPTEL (Optical Fibres and Telecommunications) course, which consists of a practical experiments kit, text and an audio-visual tutorial package.

OPTEL meets the needs of information technology at all levels of education and is an ideal foundation course for vocational training. The course offers nine major topics for study, commencing with the history of communications and the electromagnetic spectrum. It then develops the student's knowledge of light sources and light detectors. The final topics provide a detailed appreciation of optical fibre systems and the current assessment of future developments within the context of optical fibres and telecommunications.

Complementing the theoretical side, there are 14 practical experiments in which critic custom-designed modular boards for system learning are used. The student can experience the all-important areas of audio generation and modulation, fibre-optic analogue transmission and reception, analogue-to-digital and digital-to-analogue conversion, pulse transmission and reception, filtering and amplification.

Further information on these, and many other, events may be obtained from the IEE, Savoy Place, London WC2R OBL, Telephone 0171 240 1871, or the IEEE, Savoy Hill House, Savoy Hill, London WC2R OBB, Telephone 071 836 3357.
Although microcontrollers are now firmly established, we are pretty sure that the present Z80 processor card will appeal to many of you. Easy to use in combination with such options as a liquid crystal display and an infra-red remote control, and supported by a BIOS that takes the hassle out of I/O programming, this is the nineties-style way of dealing with an 'evergreen' 8-bit microprocessor.

Design by A. Rietjens

This Z80 card is easy to use for a wide range of applications thanks to its solid base formed by a number of Z80-family ICs. Remarkably, the multi-purpose character of the card is not compromised in any way by the I/O options available. These options provide functions that normally call for the soldering iron to be switched on because you have to build them yourself, not even mentioning the effort that goes into writing suitable control software. The hardware and software proposed here ensures, in the best possible way, that non-used functions do not interfere with the ones that are used, or can be adapted easily for other purposes. An example of this is a PC-XT keyboard that may be used as an input device with the present card.

Block diagram

As may be expected, a system as outlined above requires quite a bit of electronics. To keep you from losing track at this point already, have a look at the block diagram in Fig. 1. The Z80-CPU is used alongside two Z80-PIOs and one Z80-CTC. Together with the memory, these ICs form the heart of the Z80 card, which is completed with the usual I/O decoding and memory addressing logic. The latter supports the use of bank switching, so that up to 128 Kbyte may be addressed.

As you can see in the block diagram, there is no shortage of I/O and interfacing capacity: RS232, a parallel printer and a display are all catered for. The card receives

---

**MAIN SPECIFICATIONS**

Hardware:
- Z80B-CPU running at 5 MHz
- 32 I/O lines, min. 8 and max. 16 for internal use
- 4 timers
- Up to 64 Kbyte RAM and 64 Kbyte ROM or EPROM
- 8-bit A-D/D-A converter
- Standardized RS232 serial interface; all standard baud rates between 50 and 38,400
- Centronics-compatible parallel printer interface
- Two connections for 'universal I/O Interface' extension cards
- On-board watchdog
- Input device: PC/XT keyboard or RC5 infra-red receiver
- Connection for LC display with up to 2x40 characters
- On-board battery backup

Software:
- BIOS available to control and test all card functions
- BIOS is MSX-compatible
- Built-in test routines
Fig. 1. This block diagram clearly shows the structure of the Z80 card. The Z80-CPU is supported by two Z80-PIOs and one Z80-CTC. Together with the memory, these four ICs form the heart of the system.

data either from a terminal via the RS232 link, or more directly via an XT-compatible keyboard or any RC-5 compatible infra-red remote control. Apart from the digital interfaces, the card also offers an analogue interface in the form of an 8-bit ADC/DAC.

Those of you who require even more I/O capacity will be pleased to find two universal buses that carry the (buffered) databus, a select line and two address lines. This extension bus is readily connected to any peripheral device or card that does not require more than four addresses in the I/O range. Examples of cards that can be connected are the relay card for the universal bus (Ref. 1) and the opto interface card for the universal bus, to be published in a future issue.

The Z80 card has a watchdog that serves to signal power supply failures. When such a failure occurs, it arranges for the 'current state of affairs' to be stored in time by issuing a non-maskable interrupt (NMI). It also serves to re-initialize the card by means of a reset after a software crash, and to switch between the battery and the power supply to prevent data loss when the system is switched on and off.

**Memory structure**

To operate the Z80 requires an external memory in the form of an EPROM or a RAM. As shown in the block diagram, the present card offers four memory configurations. The standard system configuration consists of two 16-KByte PROMs and one 32-KByte RAM. One EPROM contains the application program, the other the system BIOS (basic input/output system). The basic software available in the system allows an application program (stored in the first EPROM) to be started automatically. The other memory configurations allow the BIOS to be combined with user software run from a 27128, 27256 or a 27512, with a 64-KByte RAM in parallel. Further information on the exact memory and address allocations for each configuration may be found in Figs. 2a and 2b.

**The Z80 BIOS**

A BIOS is basically a program structure that enables the basic hardware and software to function as intended.
Memory and I/O address decoders

The selection signals of the main components in the Z80 system must be decoded to realize different memory configurations while avoiding conflicts. The configuration chosen and the bank switching information together affect the memory addressing. The configuration is selected before the system is started, and the bank switching is arranged by software. However, provision is made in a hardware to ensure that bank switching is possible only if allowed in the configuration used.

For those of you who do not know what bank switching is all about, the following explanation. Since the Z80 can only address 64 Kbytes of memory, certain provisions must be made if we want it to access a larger memory. In this case, we wish to address 128 Kbyte, divided into blocks of 32 Kbyte each. This means that there are four possibilities to select a total of 64 Kbytes of memory. Hence, two bits are required to implement bank switching — each bit selects two blocks of 32 Kbyte. When switching such a bit, we must take care not to switch off the block currently used by the CPU. Fortunately, the BIOS contains routines that make bank switching smooth and easy, as required, for instance, to access the RAM 'alongside' the BIOS EPROM in configurations 1, 2 and 3. The bank switching information is stored in a latch that can be written to via I/O addressing.

PIOs: parallel I/O

The Z80 card has two PIO (parallel input/output) ICs, each of which contains two 8-bit ports. The PIOs are initialized to bit input/output mode, which means that each bit may be used individually as an input or an output. PIO1 is partly used for internal functions, while PIO2 is available to the user, and is programmed as an input (this can be changed as required via the BIOS).

One port of PIO1 is used for internal signals, and the other to decode the signals supplied by the infra-red receiver. When the IR receiver is not used, the port is, of course, free for other applications. Among the functions of the other port in PIO1 are Centronics handshaking, decoding the PC-XT keyboard, and detecting interrupt signals issued by the RS232 interface. Evidently, these port functions can not be redefined via the BIOS.

CTC: four timers

The Z80-CTC on the present card contains three counters/timers. Of these, timer 3 is used to generate interrupts at 10-ms intervals during which time-dependent functions
can be completed. A software 'hook' is provided to extend this interrupt routine with your own software. A hook is the software equivalent of a road diversion. The system area of the memory contains addresses that are filled with return instructions after the card is switched on. A number of BIOS routines start with a call to one of these addresses. Normally, this address contains a 'return' instruction to the BIOS routine. However, by replacing this return with a call to a user routine, the program can be diverted to a different extension of the BIOS routine, which is thus 'hooked' to the basic one. Five addresses are available for each hook, which is sufficient to place a call and a return. If a jump instruction is used at the hook address, the extension subroutine does not return to the hook and the basic BIOS routine, which is thus not executed. The diskette supplied with this project (order code FcS 1711) contains an example of the use of a hook.

Returning to the functions of the timers, the interrupt routine for Timer 3 counts down the 'on' time of the on-board buzzer, so that the software need not wait for this. Timer 0, Timer 1 and Timer 2 in the CTC are free for your own use.

**Keyboard and IR control**

Parallel input to the Z80 card is furnished either by a PC-XT keyboard or the RC-5 code infra-red receiver described in Ref. 2. The PC keyboard is connected to the board via its curly cord and 5-way DIN plug. The Z80 card automatically detects the parallel input device at power-on. Any type of RC-5 compatible IR transmitter may be used. The push-button with the number '1' on it is defined as the escape (ESC) key, while the other buttons are assigned an ASCII value equal to their code plus 32. The key definitions are stored in RAM, which allows them to be readily changed. The jumper marked REM was originally designed into the circuit to select remote control data tables. Its function has been removed, however, leaving it free for your own programming experiments.

**RS232 and Centronics interfaces**

The RS232 and Centronics interfaces on the Z80 card enable it to communicate with a terminal (or a PC running communication software). The interface gives the Z80 card the function of DCE (data communication equipment) which means, among others, that the Z80 card will only 'do' something via the RS232 if so requested. The software contains routines that allow parts of the memory to be read or written via the RS232 interface. It is also possible to adapt the baud rate and the transmission format. The interface supports all standard baud rates between 50 and 38,400.

**Watchdog and battery backup**

As already mentioned, the watchdog has a number of functions on the Z80 card. To begin with, it ensures the minimum required length of the CPU reset pulse when the card is switched on. In addition, the watchdog monitors the unregulated and the regulated supply voltages, and arranges the switching between the 5-V supply and the battery, and vice versa. The watchdog has an input that continually checks if the card has not 'crashed'. If a crash occurs, the watchdog resets the card. The latter function of the watchdog will be particularly valued with
Fig. 4. Complete circuit diagram of the Z80 card. Among the advantages of using Z80-family components is their downright simple connection.
measurement and control applications that run unattended.

Liquid crystal display

The Z80 card offers the possibility of connecting a liquid crystal display (LCD) of the normal or back-lighted type. Although LCDs with up to 80 characters are supported, the preferred type has 2x40 characters. However, one-line and four-line LCDs may be connected also. Nearly all of these intelligent LCDs are based on the HD44780 display controller from Hitachi, and have basically the same connections, albeit that the pins are sometimes arranged differently.

Circuit description

After a rather lengthy tour along the various functions shown in the block diagram of the Z80 card, it is time to see how these functions take on their practical shape.

The circuit diagram is given in Fig. 4. In the upper left-hand corner of the diagram we find the Z80B-CPU. A 6-MHz processor is used here because the system clock frequency is 5.0688 MHz. This frequency is used by the serial interface circuit to derive the standard baud rate series, starting at 50 bits/s.

The memory and I/O components are seen to the right of the Z80B-CPU and below it respectively. The I/O ICs are, of course, also B-versions because of the system clock frequency. The advantages of using Z80 family I/O components are basically that they are inexpensive, widely available, and extremely easy to implement. Essentially, once all the system and data lines of the components are connected, only the 'select' lines remain. Further, we must give some thought to the priority level assigned to each source that can generate an interrupt. This is arranged via the IEO (interrupt enable output) and the IEE (interrupt enable input) terminals, which are connected into a chain. The priority order is defined as follows: (1) IC7 (PIO1 for internal use); (2) IC6 (P102 for external use); ICs (CTC).

In the basic software, only P011 and the CTC generate interrupts. The PIO interrupts originate from the RS232 port and/or the PC/XT keyboard. The CTC generates an interrupt every 10 ms, during which, among others, the IR keyboard is checked. Without the pull-up resistor at pin 8, the keyboard buffer would be filled with random characters if the IR remote control is not used. Resistor R23 has a similar function, and prevents unwanted interrupts if IC12 is not fitted.

The connection of the EPROM(s) and the RAM(s) is not entirely straightforward because of the different memory configurations that are allowed. Position ICz can hold either a RAM or an EPROM, which requires jumpers JP1 to JP5 to be set accordingly. The circuit diagram shows the jumpers set to the 'EPROM' positions (memory configuration '0').

Position IC1 accommodates one of three
EPROM types: the 27128, 27256 or 27512, so that address lines A14 and A15 need to be given their appropriate level. These lines are controlled by the memory address decoder located in IC9, a GAL Type 16V8. The configuration is determined directly by the setting of connectors CON0 and CON1. These are pre-set to give the memory configuration shown in Fig. 2a, and determine whether or not address lines A14 and A15 are passed.

The GAL also arranges the selection of the three memory components, which, apart from A14 and A15, also depends on the memory configuration and the SEL0 and SEL1 signals furnished by IC9. SEL0 and SEL1 allow you to switch between BANK0 and BANK1 in blocks of 32 KByte (see Fig. 5).

Transistors T1 and T2, and resistors R5 and R6, ensure that the RAMs do not remain selected at power-down, so that their data is available again when the system is switched on. The power supply connections of IC2 and IC3 are connected to pin 1 of IC10, which takes care of the battery backup switch-over function.

The MAX690 watchdog (IC10) switches between 5 V and the battery voltage, USBAT, as soon as the voltage at pin 2 drops below (UBAT-50 mV), or rises above (UBAT+50 mV). Further, watchdog types MAX690 and MAX694 generate a reset if the supply voltage drops below 4.65 V. Those of you who want a wider margin are advised to use the MAX692, which issues a reset at 4.4 V. Finally, the watchdog supplies a defined reset pulse at power-on (MAX690 and MAX692: 50 ms; MAX694: 200 ms). The input voltage to the card is monitored with the aid of R7 and R8. If the voltage at pin 4 of the watchdog drops below 1.3 V, output FPO goes low. Provided the NMI jumper is installed, this low level can trigger a non-maskable interrupt that allows the current situation to be saved before the supply voltage drops below 4.65 V, and IC10 resets the card. This option is supported only by a software hook contained in the NMI.

Apart from monitoring the supply voltages, the watchdog is also capable of checking if the Z80 card is still running. The watchdog timer monitors the WDI input, to which a signal must be applied that changes at least every 1.6 ms. If this signal fails, the watchdog resets the card. The WDI input may be connected to the selection signal of IC7 via jumper PIO1. Since, if the BIOS is used, the CTC generates an interrupt every 10 ms, and the associated subroutine addresses IC7, the presence of the selection signal is a good sign that the card is still running.

A second GAL, IC8, contains the address decoding logic for the I/O circuits. Here, the advantage of a GAL is a drastic reduction of the chip count for an address decoder that allows the I/O ICs to be addressed fully and without ‘image’ areas elsewhere in the memory. An address overview of the I/O components is given in Fig. 6.

The analogue interface is built around an AD7569 8-bit A-D/D-A converter. This IC ranges the handshaking protocol on the serial link, which does away with the need for any software equivalents. The step-up converters contained in the MAX232 ensure RS232 signal levels of +10 V and -10 V, which will work in most, if not all, applications. Connector K11 is wired such that the Z80 card forms a DCE that is readily connected to a DTE (data terminal equipment; a computer in most cases) via a 9-way flat-cable.

At this point we have nearly completed our tour along the main components in the circuit diagram. Connector K12 serves to hook up a PC/XT keyboard. Such keyboards are widely available at very low prices from PC surplus outlets. If you happen to have a PC XT/AT type with automatic switching, the Z80 card ensures that it is set to PC XT mode. The keyboard is reset by software following a hardware reset. This is done via transistor T5.

The contrast of the LCD connected to K10 is controlled via preset R2. A back-light display may be used in low ambient light conditions. Because of the possible need for a back-light supply, and to keep display multiplexing noise away from the processor, the LCD is powered separately by IC12, while the rest of the circuit is powered by IC10. The back-light supply depends on the LCD type used. There are types that require a supply voltage of 5 V (replace R21 with a wire link), and types that require a certain current (in which case R21 must be given an appropriate value). The back-light connections are pins 15 and 16 of connector K9. The jumper marked ‘LCD’ provides a simple way of switching the back-light on and off.

The printer datalines are furnished by latch IC16, while the control signals to and from K9 and K10 are buffered by IC17. This IC also buffers the control signals to the two external bus connections. The databases are buffered by IC14 and IC15. The pinning of connectors K7 and K8 is compatible with the universal bus (Ref. 1).

That concludes the description of the Z80 card as far as its concept is concerned. Next month we will tackle the construction and testing of the card, as well as making use of the associated test software contained in the BIOS EPROM.

References:
COMPACT MAINS POWER SUPPLY

Design by A. RoBrucker

The proposed mains power supply is a versatile unit for use where a stable voltage, medium power and good parameters are required. The quality of its output is easily comparable to that of good-quality commercial products, but it is considerably less expensive than those units.

The circuit

The circuit of the supply is just as compact as its mechanical design: a transformer, a rectifier, two ICs and some additional passive components are all that is required. The design is based on IC1, a 5-pin, voltage regulator Type L200 with integral, presettable current limiting and thermal overload protection.

To ensure correct regulation of the output voltage, the device has an internal 2.75 V reference source. The voltage at pin 4 is compared with the reference voltage to enable the correct level of output voltage, $V_o$, to be set with potential divider $R_3 - R_4 - P_1$. With values as shown in Fig. 1, the output voltage can be set between 5 V and 26 V with $P_1$.

It should be borne in mind that the maximum input voltage to IC1 is 40 V and the maximum difference between its input and output voltage is 32 V. That means that the secondary voltage of the mains transformer must not exceed 24 V. This gives an unregulated voltage input into IC1 of 32 V, resulting in a maximum output voltage of 24 V. To set $P_1$ for an output voltage >24 V would not be a good idea, because the regulation at fairly high output currents would then no longer function correctly, owing to the inadequate voltage reserve. The result would be an unacceptable 100 Hz ripple on the output.

To enable correct functioning of the integral current limiting, the voltage drop across $R_2$, connected between pins 2 and 5 is monitored; when it reaches 450 mV, current limiting begins. The maximum output current, $I_o$, is therefore given by

$$I_o = \frac{450}{R_2} (mA).$$

With $R_2 = 0.22 \Omega$ as in the diagram, the maximum output current is thus 2 A.

Voltage monitoring

The voltage monitoring circuit is based on IC2. When the load draws too high a current, the current limiting in IC1 operates, resulting in a drop in the output voltage. This reduction is detected by IC2, which then causes $D_3$ to light. Even very brief drops in the output voltage are indicated by this LED for a period, $t$, that depends on the capacitance of $C_3$ and may be calculated from

$$t = \frac{1}{2\pi \cdot (R_6 \cdot C_3)^{1/2}}.$$

Virtually all electronic circuits need some sort of power supply, often a high-quality one. For the latter, if an output current of not more than 1.5 A and an output voltage of 5–20 V are required, the supply described here is ideal. Its design is compact: even the transformer and heat sink are housed on the printed-circuit board. It provides overload indication, current limiting and protection against short-circuits.

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Fig. 1. Circuit diagram of the compact mains power supply.
Fig. 2. Printed circuit board for the compact mains power supply.
Transformer

Before the components for the mains supply are bought, it should be decided what level of output voltage is, or will be, required. It would not make much sense to fit a 24 V transformer and then set P1 permanently for 5 V output. The consequent dissipation in IC1 would be high, whereupon load currents of up to 2 A not would be available, since the integral thermal protection circuit would limit the output current. Component values differing from those in the parts list and on the diagram are, therefore, given below for the most usually required output voltages.

- 5 V: 
  - T1 = 8 V secondary; R1 = wire link; D1 is not required; C1 = 16 V.
  - 6 V: 
    - T1 = 8 V secondary; R1 = wire link; D1 is not required; C1 = 16 V.
  - 9 V: 
    - T1 = 9 V secondary; R1 = wire link; D1 is not required; C1 = 25 V.
  - 12 V: 
    - T1 = 12 V secondary; R1 = wire link; D1 is not required; C1 = 25 V.
  - 15 V: 
    - T1 = 15 V secondary; R1 = 220 /; C1 = 25 V.
  - 18 V: 
    - T1 = 18 V secondary; R1 = 330 /; C1 = 40 V.

It should be borne in mind when ordering the transformer that the alternating secondary current should be about 1.4 the desired direct output current. At the same time, the value of R2, the rating of C1 and the current rating of B1 should be considered. Component values and ratings differing from those stated in the parts list or on the circuit diagram for a number of load currents are given below:

- 300 mA: 
  - T1 = 700 mA; R1 = 0.82 /; 1 W; C1 = 1000 /F; B1 = 640C1000.
  - 650 mA: 
    - T1 = 1 A; R1 = 0.68 /; 1 W; C1 = 2200 /F; B1 = 840C1000.
  - 800 mA: 
    - T1 = 1.2 A; R1 = 0.56 /; 1 W; C1 = 2200 /F; B1 = 840C1500.
  - 950 mA: 
    - T1 = 1.5 A; R1 = 0.47 /; 1 W; C1 = 2200 /F; B1 = 840C2200/1500.
  - 1.15 A: 
    - T1 = 1.7 A; R1 = 0.39 /; 1 W; C1 = 2200 /F; B1 = 840C3200/2200.
  - 1.35 A: 
    - T1 = 2 A; R1 = 0.33 /; 5 W; C1 = 4700 /F; B1 = 840C3200/2200.
  - 1.5 A: 
    - T1 = 2.2 A; R1 = 0.27 /; 5 W; C1 = 4700 /F; B1 = 840C3200/2200.

In normal operation, the potential at pin 7 of IC2 is 4.55 V. When that voltage drops for 0.13 s, if a longer period is required, the device must not operate with voltages above 18 V; that is why R1-D1 limits the voltage to 15 V. That voltage results from transformer secondary voltages of 12 V and higher. If, therefore, a transformer is used with a 12 V secondary, D1 is not required; C1 = 25 V.

If there is no output voltage, it is likely that the alternating secondary current should be about 1.4 the desired direct output current. At the same time, the value of R2, the rating of C1 and the current rating of B1 should be considered. Component values and ratings differing from those stated in the parts list or on the circuit diagram for a number of load currents are given below:

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  - 650 mA: 
    - T1 = 1 A; R1 = 0.68 /; 1 W; C1 = 2200 /F; B1 = 840C1000.
  - 800 mA: 
    - T1 = 1.2 A; R1 = 0.56 /; 1 W; C1 = 2200 /F; B1 = 840C1500.
  - 950 mA: 
    - T1 = 1.5 A; R1 = 0.47 /; 1 W; C1 = 2200 /F; B1 = 840C2200/1500.
  - 1.15 A: 
    - T1 = 1.7 A; R1 = 0.39 /; 1 W; C1 = 2200 /F; B1 = 840C3200/2200.
  - 1.35 A: 
    - T1 = 2 A; R1 = 0.33 /; 5 W; C1 = 4700 /F; B1 = 840C3200/2200.
  - 1.5 A: 
    - T1 = 2.2 A; R1 = 0.27 /; 5 W; C1 = 4700 /F; B1 = 840C3200/2200.

When all the components have been selected and bought, construction can be started. Begin with the resistors and terminal strips, followed by the fuse holder, capacitors, diodes, rectifiers and the two ICs. The transformer comes last. Make sure that the ceramic has been removed from the ends of the wires that are inserted into the terminal strips.

Regulator IC1 must be fitted on to a heat sink without insulating washer but with some heat conducting paste. The heat sink will be set to the desired value with the aid of P2.

The voltage monitoring circuit is preset by first turning P2 until D3 just lights and then turning it till D3 just goes out. Because of the time-constant, P2 should be turned very slowly.

If the transformer hums and the fuse blows, either C1 is connected with incorrect polarity or two secondary windings of Tr1 have been connected in anti-parallel instead of parallel.

If there is no output voltage, it is likely that the secondary windings of Tr1 that should have been connected in series are, in fact, linked in anti-series.

Do not connect two of these power supplies in parallel. However, connecting them in series to obtain a higher output voltage is perfectly all right. It is also possible to use two power supplies to construct a symmetrical supply.
1.3 GHz PRESCALER

Design by P. Esser

Not only do the majority of frequency counters found in smaller workshops and laboratories not operate above 10 MHz, but usually they cannot be modified to work at higher frequencies either. To overcome that problem, here is a prescaler that delivers a clean rectangular signal at TTL level at frequencies up to 1.3 GHz and which can be used with virtually any frequency counter.

The prescaler proposed here offers several advantages. Firstly, it increases the measurement range of the frequency counter to which it is linked and, secondly, it makes it possible to use a much shorter cable between counter and instrument on test—see Fig. 1. A disadvantage is, of course, that, to see the selected metering range, you must look at both the counter and the prescaler.

Scaler ICs

A first scaling down of the input signal is effected by a chip specially designed for this purpose. This can be either the Telefunken Type U664B or the Siemens Type SDA4211. Block diagrams of these circuits are shown in Fig. 2.

The U664B was originally developed for use in the frequency synthesizer of a television receiver. Without any additional components, it divides by 64. In the absence of an input signal, it operates in the highest frequency range. Normally, the only external components required are two small capacitors.

The SDA4211 offers two scaling factors: 64 or 256, depending on the potential at pin 5. If that pin is at +5 V, the input signal is divided by 64; when the pin is at earth, scaling is by 256. On the PCB—see Fig. 4—the selection is facilitated by a 3-way terminal strip and a jump link.

The two circuits are fully interchangeable as regards pinout and function, but not, of course, in scaling factor.

Two paths

The measured signal (frequency \( f_s \)) is split into two immediately after the input socket—see Fig. 3. One part is fed to the prescaler proper (lower part of the diagram) via \( C_4 \), while the other is taken to a processing and amplifying section (upper part of the diagram) via \( L_1 \).

Anti-parallel connected diodes \( D_2 \) and \( D_3 \) limit the level of the input signal to not more than \( \pm 700 \text{ mV} \). The signal is then applied to pin 2 of IC3. The symmetrical input of this circuit is connected asymmetrically, since the second input, pin 3, is connected to ground via \( C_{11} \). Jumper \( JP_1 \) is the earlier mentioned scaling selector if the SDA4211 is used. If the U664B is used, the 3-way terminal strip and jump link are not required.

The measured signal (frequency \( f_s \)) is available at pin 6, from where it is applied to potential divider \( R_7 - R_8 - P_1 \). From there it is fed to amplifier \( T_2 \), whose output is applied to the first of three cascaded Type 74LS90 decade counters, IC4, IC5, and IC6.

Each of these counters divides its input signal by 2.5. This somewhat unusual scaling factor comes about as follows. The upper half of the IC divides by 5. For every five input pulses, the \( Q \) output goes high twice; in other words, the \( Q \) output delivers an output pulse for every 2.5 input pulses. The output of the cascaded threesome is thus a signal of frequency \( f_s \times 1000 \).

The other part of the input signal is applied via \( L_1 \) and \( C_2 \) to \( T_2 \), which, connected as a common-emitter circuit, behaves exactly like an inverting opamp. The voltage amplification of the stage is roughly the same as the open-loop amplification of the transistor, but it is dependent on the source impedance. Diode \( D_3 \) limits the negative half of the signal to not more than \(-700 \text{ mV} \).

The output of the stage is taken from the collector of \( T_2 \) and then further amplified in \( T_3 \), which is also connected as a common-emitter circuit. It is then taken from the collector of \( T_3 \) and applied to NAND Schmitt trigger IC1b, which, with the other three NAND gates, ensures clean edges and correct gating of the two signals. When switch \( S_1 \) is open, the original signal \( f_s \) is available at the output; when it is closed, the scaled down signal \( f_s \times 1000 \) is at the output socket.

Construction

Populating the printed-circuit board shown in Fig. 4 is straightforward, but greater care than usual is required around the input socket where surface-mount components are used. Inductor \( L_1 \) must be wound by the constructor. It consists of 2-3 turns enamelled copper wire (dia. 0.4 mm) on a small ferrite core.
The input socket is a BNC type for PCB mounting; this obviates the need of screened cable at the input.

If the SDA4211 is used (IC3), the link at JP1 should connect the +5 V line to pin 5 of IC2. If the U664B is used, the jumper should not be used. Nothing more can go wrong here than the scaling factor.

1N4148

Fig. 3. Circuit diagram of the 1.3 GHz prescaler.

Brief specification
- Two switchable measurement ranges: 1:1000
- Upper frequency limit 1.3 GHz
- Input sensitivity <100 mV
- Compact, economical design
- Power supply 5 V
- Single board construction

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PARTS LIST

Resistors:
- R1, R3 = 1 kΩ
- R2, R4 = 47 kΩ
- R5 = 390 Ω
- R6 = 560 Ω
- R7 = 150 Ω
- R8 = 2.2 kΩ
- R9 = 330 kΩ
- P1 = 10 kΩ preset, horizontal

Capacitors:
- C1 = 10 μF, 16 V
- C2, C3 = 1 μF
- C4 = 120 pF, surface mount
- C5 = 1 nF, surface mount
- C6–C10 = 10 nF
- C11 = 820 pF, surface mount

Semiconductors:
- D1 = 1N4148
- D2, D3 = BAT81, BAT82 or BAT83
- T1, T2 = 2N918
- T3 = BF324
- IC1 = 74LS132
- IC2, IC4, IC5 = 74LS90
- IC3 = U664B or SDA4211

Miscellaneous:
- L1 = see text
- S1 = single-pole on/off switch
- K1 = BNC socket for PCB mounting
- JP1 = 3-way terminal strip
- PCB 914059

Fig. 4. Printed circuit board for the 1.3 GHz prescaler.
If you don't ponder things too deeply, it might appear as if the whole edifice of electronics is supported solely by the transistor and the chip, just as once it was supported by the thermionic valve. But a fat lot of good these dynamic components would be if it weren't for the passive Cinderellas of resistance, capacitance and inductance. In other words, a stick of carbon, two pieces of metal separated by some sort of dielectric and a length of wire fashioned into a helix. Not an impressive picture in terms of the actual make-up of these components, to be sure, but their physical behaviour, when examined carefully, more than makes up for their apparently simple physical construction.

Let us look particularly at the humble inductor, a much neglected component on the electronics scene, and see what properties it possesses to make it just as important as its often more glamourized partners.

When an electric current flows in a wire, one of its most important manifestations is the establishment of a magnetic field in the immediate vicinity of the wire. This field is composed of so-called lines of magnetic force or magnetic flux, which takes the form of concentric circles around the wire that lie both within and outside the conductor. Figure 1 shows the kind of field set up and how the directions of current and lines of force are related. Strictly, of course, the concentric circles are concentric tubes and the field is at its most intense, that is, has its greatest flux density, at the surface of the conductor. This intensity falls off as the distance from the conductor increases, which is indicated on a magnetic field diagram by the variations in line spacing. We notice, too, that lines of force always form closed loops whatever the situation. A magnetic line doesn't suddenly come to a stop with nowhere to go. The direction in which the magnetic force would act on an isolated north pole placed within the field is indicated by the arrowheads: this direction reverses if the current in the conductor is reversed. Keep in mind, of course, that the concept of lines is a conventional fiction; a real field does not have barren spaces between the lines. We simply use them as a convenient way of representing magnetic field densities and nowadays these are not measured in lines (as they once were) but in terms of the electromagnetic effect the field will produce.

If the wire carrying the current is formed into a loop, rather as shown in Fig. 2, the lines of force all pass through the loop in the same direction. A conventional coil of wire, or solenoid as it is usually called, is nothing more than a number of continuous loops, and a current flowing in such a coil establishes the lines of force in a lengthwise direction through the centre of the coil, emerging from the end and completing their circuits through the surrounding medium. Figure 3 shows how the concentric loops merge together to give a resultant field of considerable intensity, rather as the field surrounding a permanent bar magnet (of iron filings memory!) displays its characteristic pattern. Like the bar magnet, too, the field around a current-carrying coil exhibits a north and a south pole; these poles change place when the current in the coil is reversed. The field vanishes when the current flow ceases.

It is the appearance—and the disappearance—of the field established around a current-carrying conductor that determines the whole phenomenon of electrical inductance.

Self inductance

So, what do we mean by inductance, anyway? Well, let us return to Fig. 2. If the magnetic flux caused by the current in a single loop of wire is looked at closely, it is seen that every line must pass through the loop somewhere or, put into other words, every line must 'link' with the loop. The number of lines enclosing the loop in this way are referred to as the flux linkages and in this basic example the number of such linkages must be the same as the number of flux lines.

Now, thinking about the flux associated with the solenoid of Fig. 3, we see that the majority of the lines link with every turn of the coil, although there are those that link with a few of the turns only.

If we consider each turn separately, the total linkages with the coil consist of the summation of the number of linkages linking each
and it is now evident that this total is much greater than the number of flux lines. Flux linkage, then, is simply the flux multiplied by some constant that itself depends upon the shape, the closeness of the turns and the physical dimensions of the circuit, that is, the opportunities that exist for the flux to link with as much of the circuit as possible.

Since the total flux is proportional to the current, so also will be the total flux linkages; hence, we can say that the ratio of flux-linkage to current is constant for any given circuit arrangement. This constant is the self-inductance of the circuit and is symbolized by the letter \( L \).

Now, a bold statement of this sort is likely to leave the majority of us a bit frustrated. So, to get our teeth into the problem let us have a brief look at the work of Michael Faraday and see what effect the electromagnetic field associated with a solenoid has on its surroundings.

**Mutual inductance**

Faraday used a circuit rather like that shown in Fig. 4. Here we have the coil of Fig. 3 in a circuit consisting of a switch \( S \) and a source of e.m.f., battery \( B \). We call this coil the primary winding. In close proximity to this primary winding there is another coil, the terminals of which are connected to a current sensing meter, \( G \). This coil is the secondary winding. Faraday noticed that as long as a steady current was maintained in the primary winding, that is, as long as the field surrounding this coil remained constant, there was no deflection on the meter. However, when the switch \( S \) was opened or closed, there was a transient deflection, a momentary flick of the pointer that returned quickly to zero as the field either established itself to a constant value (switch closed) or collapsed from a value to zero (switch open). Faraday deduced that whenever flux linkages between the two circuits were changing, a current flowed in the circuit of the secondary winding, and this in turn implied that an e.m.f. must be acting on it. On top of this, the meter deflections were in opposite directions according as switch \( S \) was being closed or opened; that is, the current flowed (and the e.m.f. acted) in opposite sense according as the flux linkages between the coils were increasing or decreasing.

This little bit of electrical history, easy to appreciate in retrospect, illustrates the phenomenon of electromagnetic induction: the current in the secondary winding is an induced current and the e.m.f. producing it is an induced e.m.f. A changing field always sets up an e.m.f. in any conductor situated within the field.

On this basis, the question naturally arises as to whether any corresponding effect occurs when the magnetic field associated with the secondary winding, set up by the induced current, interacts with the primary winding. We have seen that the direction of the induced e.m.f. depends on whether the change in the mutual flux linkages is increasing or decreasing. The current induced in the secondary winding will set up a magnetic field of its own and this will provide linkages with the primary winding. These linkages may either increase or decrease the total linkage between the coils according to the direction of the secondary current. Now, it is found that when the mutual flux linkages caused by the primary current are increasing (\( S \) being closed), the secondary current flows in the direction which reduces the total mutual linkages. Conversely, when the current in the primary winding is falling (\( S \) being opened), the direction of the secondary current is such that the mutual linkage is increased. We can interpret this as an attempt by the current in the secondary coil to keep the flux linking constant, whether the current in the primary is increasing or decreasing.

So, what we have in effect is a kind of electrical inertia; both a rise and a fall in the circuit current are resisted by the appearance of an induced e.m.f. that works to maintain the existing conditions.

What about a single coil on its own? Does the same thing happen there? Consider the primary coil of Fig. 4 to be on its own. As soon as \( S \) is closed, the magnetic field builds up and sweeps outwards from the turns of the coil; in doing this, it must induce a voltage in any conductors situated within the field and that includes the coil which is producing the changing field. Hence, as the linkages are increasing in the coil after the switch is closed, the induced e.m.f. acts so as to oppose the increase; that is, it opposes the e.m.f. of the battery which is endeavouring to send current through the coil. This effect is known as the back-e.m.f. of self-induction: as one of my teachers put it many years ago, all inductors have suicidal tendencies. Thus, the current in an inductive circuit does not assume instantaneously the value it would have if this were calculated by Ohm's law, that is, on the basis of the resistance of the inductor. This must not be confused with inductance.

Similarly, when the switch is opened, on an Ohm's law calculation, the current should fall immediately to zero, but this does not happen. As the current falls, the magnetic energy stored in the field must decrease proportionally, and an e.m.f. is induced in the coil that this time tends to prevent the reduction in flux linkages. The manner in which the current changes in the circuit for both switch on and switch off is illustrated in Fig. 5, which also shows the corresponding changes in a circuit containing pure resistance only.

**Conservation of energy**

The effects of inductance accord with the principles of the conservation of energy. Nature never gives us something for nothing: every bit of energy in a system has to be accounted for in one way or another. In the coil, the magnetic energy is caused by the current and that current in turn is derived from the chemical energy stored in the battery. While the current is steady, the magnetic field and hence the magnetic energy stored in the field will remain constant and the only energy drawn from the battery is that of heat energy, which is dissipated in the resistance of the coil. If the battery voltage is increased, the current, in accordance with Ohm's law, should increase in proportion. This involves an increase in magnetic energy, but while this transformation from chemical to magnetic energy is taking place, all the chemical energy cannot change immediately into heat energy. Thus, the current cannot change immediately to a new level as dictated by Ohm's law, that is, on the grounds that all the energy supplied by the battery is converted to heat. The back-e.m.f. is consequently the effect that limits the current to a smaller value; once the magnetic field is established at its new level, the current reaches its greatest value and the back-e.m.f. vanishes.

In the same way, when the current is switched off, it should, in accordance with Ohm's law, fall immediately to zero, since no more energy is being supplied from the battery. But, as the current falls, the energy stored in the field must also fall proportionally, that is, it must be transformed into another form. This transformation is actually to heat energy in the circuit conductors and takes the form of...
an e.m.f. that tends to maintain the current flow and permit the conversion to heat to take place.

It should be obvious that the greater the time rate at which the current changes, the more rapid the corresponding changes in electromagnetic energy must occur and the greater the induced e.m.f. will be. This is embodied in Faraday's law which states that the induced e.m.f. is equal to the product of inductance and rate of change of current. The unit of inductance is the henry, H, named after Joseph Henry (1797-1878), the American physicist and pioneer of electromagnetism, who was a contemporary of Faraday. Sub-units are the millihenry, mH, and the microhenry, µH.

A circuit has a self-inductance of 1 H when the current through it changes at the rate of 1 A sec⁻¹ and causes an induced e.m.f. of 1 V.

**Types of inductance**

Broadly speaking, inductances as found in general electronics may be classified as (a) coils that have very large inductance values in the minimum of space and reasonably large current-carrying capacity — these types usually have iron cores and take the forms of power transformer or choke, or armature and field coil in motors and generators; (b) coils used in circuits of radio and television receivers, small transmitters and general amplifier systems. The latter types generally consist of relatively small windings wound on non-ferrous supports such as paxolin or polystyrene tubing or, in some cases, as self-supporting coils in air. They have inductances of at most a few hundred millihenries and normally carry only small radio-frequency currents.

In the case of air-cored inductors, the inductance for a given configuration will be constant since the field is set up in a non-ferrous region which cannot be 'overloaded' or saturated with magnetism. For coils wound on iron cores, the inductance is roughly constant for small values of direct current, but beyond a certain point, the iron saturates and there is no further increase in the field intensity with an increase in the magnetising force. The inductance of iron-cored chokes is usually stated at a specified level of current, for instance, 10 H at 100 mA.

The other commonplace coil with a closed iron core is the low-frequency transformer. This may use the conventional laminated stack of iron or be wound on a toroid or 'anchor' ring circuit.

As in the case of Faraday's experiment, transformers depend for their action on mutual inductance. Two coils are wound in close proximity on a common core as shown in Fig. 6. If a current flows in the primary winding, some of the lines of flux established in the core will link with the turns of the secondary winding. The actual linkages between the coils will depend on the current in the primary winding and the positions of the two coils relative to each other. Because of the closed iron core, the linkages are very high and the field is confined closely to the iron circuit. The number of linkages with the secondary winding caused by the current in the primary winding is thus a mutual function of the two circuits.

Circuits of this sort have a mutual inductance of 1 H if the e.m.f. induced in one of them is 1 V when the current in the other is changing at the rate of 1 A sec⁻¹.

Iron cores act as conducting paths as far as the magnetic fields set up in the coils are concerned; hence, an e.m.f. is induced in the core material of a choke or transformer whenever it is operating from an alternating current, i.e., supply. This causes random (eddy) currents to circulate within the core, which in turn creates heating and, therefore, a loss of efficiency. Such currents are reduced by laminating the cores and restricting the currents to small and isolated regions of the core.

In the years leading up to the Second World War, the tuning of circuits at radio frequencies was carried out almost universally by fixed inductors and variable capacitors. Owing to eddy current and other losses, it was then not practicable to use iron-cored inductors at high frequencies, although audio frequency range transformers were available with very thin 'radio metal' laminations and sectionalized windings. Low-loss cores, the so-called ferrites or 'dust-iron' cores, have since been developed and are now commonplace as tuning slugs, which make it possible not only to adjust the inductance, but also to lead to very compact forms of r.f. tuning coil.

**Winding inductors**

Unless there is an inductance bridge available, winding an inductor to a specified value of inductance is a bit of a hit-and-miss affair. Inductance is proportional to the square of the number of turns, \( N^2 \), so that, if the number of turns is doubled, the inductance increases four-fold. Other factors that affect the inductance are the length and diameter of the winding, the spacing of the turns, whether the coil is a single-layer or multi-layer, and sectionalized core material.

Provided the coil is air-cored, it is not difficult to estimate with reasonable accuracy the inductance of a coil that is close-wound in a single or multiple layer. This is where the curves of Fig. 7 come in. These graphs take care of the factors relating to coil length and radius as well as the depth of the winding or the diameter of the wire. Figure 8 shows these measurements made with respect to a single-layer coil at (a) and a multi-layered one at (b).

To find the inductance of such coils, first of all work out the ratio \( R : (L + D) \), where \( R \) is the radius of the coil, \( L \) is the length of the winding, and \( D \) is the diameter of the wire (single layer) or the depth of the winding (multi-layer). Whether you use metric or imperial measures is not important, since we are only seeking a ratio; nevertheless, metric type, and along the horizontal SHAPE axis of the graph and read off the corresponding value for the MULTIPLYING FACTOR \( Y \) on the vertical axis. The inductance can then be calculated from the formula

\[
L = N^2 \frac{R}{Y} (\mu H)
\]

Next month's instalment will deal with the iron-cored transformer.
THE NICAM SYSTEM

Stereo TV sound has finally come of age with the progressive introduction over the past few years of a digital system called NICAM. This article aims at providing a background to the operation of the NICAM (near-instantaneously companded audio multiplex) system, which is now in use in most of the UK, Scandinavia, Belgium and Spain. NICAM-728, with subversions for PAL systems B/G and I, is also recommended by the EBU as the system for multi-channel sound transmission with terrestrial television. It has been adopted for use in several countries, including the UK, and now forms part of a draft CCIR recommendation.

By J. Buiting, technical editor.

WHEN we talk about different television standards, the discussion is usually about different ways of conveying the picture to the viewer. Up to ten years ago, the sound was taken for granted, which is remarkable because the stereo age was well under way at that time. Following a German initiative, some European countries introduced stereo TV sound based on an auxiliary subcarrier above the main (mono) FM carrier. Although this works, the NICAM system offers superior sound quality at a roughly equal bandwidth requirement. Originally developed by the BBC, the NICAM-728 specification has been formally approved by the Department of Trade and Industry as the United Kingdom standard for two-channel digital sound with terrestrial television broadcasts.

A brief history of stereo TV sound

Since 1979, a number of stereo TV sound systems have been introduced that were aimed at downward compatibility with the existing mono sound systems. Among the requirements for the new sound systems were:

- minimum interference and crosstalk between the channels;
- quality of existing (main) mono channel must not be affected;
- equipment to upgrade transmitters and receivers must remain as simple as possible.

The need of maintaining downward compatibility, as well as the limited bandwidth available for the new sound systems, have forced the designers of analogue stereo TV sound systems to drop some of their target specifications, and agree on certain compromises that reduce the quality that could have been achieved in theory. Analogue stereo sound systems can be made downward compatible in two ways:

- by modifying the audio signal before it is modulated on to the carrier (single-carrier principle);
- by adding a second sound carrier just above or below the existing (mono) sound carrier (dual-carrier principle).

In both cases, a decoder matrix is required to separate the left and right channels, and produce the stereo sound image. Some systems also require de-emphasis and/or de-companding to improve the signal-to-noise ratio and the dynamic range.

The dual-carrier system is basically analogue, and offers quite reasonable sound quality. However, in this day and age of digital sound, it is not surprising that alternatives have been sought, based on the technology already familiar from CD players and the sound transmission standard developed for the MAC system. In particular the channel separation offered by NICAM is much higher than that achieved by any form of analogue dual-carrier system. Overall, the sound quality of a NICAM broadcast is so close to that of a compact disk that it is hard to tell the difference by just listening.

NICAM-728 digital sound transmission

Strictly speaking, the NICAM-728 system should be classified as a dual-carrier system, because a second sound signal is introduced in the baseband spectrum (see Fig. 1). The spectrum shown is for PAL system I as used in the UK, with the main sound carrier at 6.0 MHz above the vision carrier, and a total channel bandwidth of about 8 MHz. Most other European countries use PAL system B or G, where the main sound channel is at +5.5 MHz, and the channel bandwidth is about 7 MHz.

The NICAM signal is recovered from a QPSK (quadrature phase shift keying) spectrum with a bandwidth of about 600 kHz. The centre frequency of this

Fig. 1. The frequency band occupied by the NICAM-728 digital sound signal in relation to the picture and mono (analogue FM) sound signal components in the TV baseband.
The following parameters apply:

- Pseudo-random sequence generator used previously to the multiplex frame. The FAW is restrictions as regards the baseband bandwidth. The transmitted bit stream is scrambled for descrambling in the receiver. Figure 4 shows the general layout of the scrambler.

Energy dispersal scrambling

The transmitted bit stream is scrambled for spectrum-shaping purposes (remember the restrictions as regards the baseband bandwidth). The scrambling operates synchronously to the multiplexer frame. The FAW is not scrambled, and used to synchronise the pseudo-random sequence generator used for scrambling in the receiver. Figure 4 shows the general layout of the scrambler. The following parameters apply:

- The bit that follows the FAW is the first scrambled bit, and is added modulo-two to the first bit of the pseudo-random sequence;
- The bit that precedes the FAW is the last scrambled bit;
- Scrambling takes place immediately after interleaving (and descrambling is therefore prior to de-interleaving in the receiver);
- The pseudo-random sequence is defined by a generator polynomial $x^2 + x + 1$ and an initialisation word ('seed') 11111111.

Thus, with reference to Fig. 2, the sequence starts:

```
0000 0111 1011 1110 0010
```

FAW and control information block

The FAW is 01001110, which is a series of bits transmitted continuously without gaps. One frame is transmitted every millisecond, so the overall bit-rate is 725 Kbit/s, whence the system designation NICAM-728.

The 720 bits that follow the frame alignment word (FAW) have a structure that closely resembles that of the first-level protected, companded sound signal blocks in the systems of the MAC family. After the control bits and the additional data bits follows a block of 704 interleaved sound data bits. The interleaving pattern relocates data bits which are adjacent in the frame structure of Fig. 2 to positions at least 16 clock periods apart in the transmitted data stream.

Sound multiplex and sound coding methods

To understand how the NICAM system works, we will take a look at the structure of the serial data stream at the transmitter side.

Frame structure and bit interleaving

As shown in Figs. 2 and 3, the data consists of 728-bit frames which are transmitted continuously without gaps. One frame is transmitted every millisecond, so the overall bit-rate is 725 Kbit/s, whence the system designation NICAM-728.

The 728 bits that follow the frame alignment word (FAW) have a structure that closely resembles that of the first-level protected, companded sound signal blocks in the systems of the MAC family. After the control bits and the additional data bits follows a block of 704 interleaved sound data bits. The interleaving pattern relocates data bits which are adjacent in the frame structure of Fig. 2 to positions at least 16 clock periods apart in the transmitted data stream.

Thus, with reference to Fig. 2, the sequence starts:

```
0000 0111 1011 1110 0010
```

FAW and control information block

The FAW is 01001110, which is a series of bits transmitted in that order. The control information converged to the receiver consists of a frame flag bit, $C_0$, three application control bits, $C_1$, $C_2$, and $C_3$, and a reserve sound switching flag, $C_4$ (see Fig. 3). The frame flag bit, $C_0$, is set to '1' for eight successive frames, and to '0' for the next eight frames. The frames are numbered within the sequence as follows: the first frame (Frame 16) of the sequence is the last of the eight frames in which $C_0=0$. Hence, the last frame (Frame 16) of the sequence is the last of the eight frames in which $C_0=0$. This frame sequence is used to synchronise changes in the type of information being carried in the channel.

The function of the three application control bits, $C_1$, $C_2$, and $C_3$, is to define the current application of the last 704 bits in each frame, which may be used to convey either sound samples or data. The available options are shown in Table 1. When a change to a new application is required, these control bits change (to define the new application) on Frame 1 of the last 16-frame sequence of the current application. The 704-bit sound/data blocks change to the new application on Frame 1 of the following 16-frame sequence.

The reserve sound switching flag, $C_4$, contained in the control information block is used to switch back to the output of the conventional FM demodulator when the digital sound decoding system fails. This is, of course, acceptable only if the FM sound channel carries the same programme as the failing digital channel. The means to
Table 1. Applications of 704-bit sound/data blocks.

<table>
<thead>
<tr>
<th>Application control bits</th>
<th>Contents of 704-bit sound/data block</th>
</tr>
</thead>
<tbody>
<tr>
<td>C₁  C₂  C₃ *</td>
<td>stereo signal comprising alternate A-channel and B-channel samples</td>
</tr>
<tr>
<td>0  0  0</td>
<td>two independent mono sound signals (designated M₁ and M₂) transmitted in alternate frames</td>
</tr>
<tr>
<td>0  0  0</td>
<td>one mono signal and one 352-kBit/s transparent data channel transmitted in alternate frames</td>
</tr>
<tr>
<td>0  0  0</td>
<td>one 704-kBit/s transparent data channel</td>
</tr>
</tbody>
</table>

* C₃ = 1 provides for signalling additional sound or data coding options. When C₃ = 0, decoders not equipped for these additional options should provide no sound output.

Table 2. Coding/protection range selection.

<table>
<thead>
<tr>
<th>Coding range</th>
<th>Protection range</th>
<th>Scale factor value</th>
</tr>
</thead>
<tbody>
<tr>
<td>1st</td>
<td>1st</td>
<td>R₂</td>
</tr>
<tr>
<td>2nd</td>
<td>2nd</td>
<td>R₁</td>
</tr>
<tr>
<td>3rd</td>
<td>3rd</td>
<td>R₃</td>
</tr>
<tr>
<td>4th</td>
<td>4th</td>
<td>0</td>
</tr>
<tr>
<td>5th</td>
<td>5th</td>
<td>0</td>
</tr>
<tr>
<td>5th</td>
<td>7th</td>
<td>0</td>
</tr>
<tr>
<td>5th</td>
<td>7th</td>
<td>0</td>
</tr>
</tbody>
</table>

Applications of 704-bit sound/data blocks.

Inhibit such switching is incorporated in the control information. Control bit C₄ is set to '1' when the FM channel carries the same sound programme as the digital stereo signal or the digital mono signal (where two digital mono signals are transmitted, this refers to the M₁ signal only). When the FM channel is not carrying the same programme as the digital sound channel, C₄ is set to '0'. In this state, it can be used to prevent switching to the FM sound. Finally, C₅ has no meaning in the case of data transmission.

Additional data and the sound/data block

Data bits AD₀ to AD₁₀ (see Fig. 3) are reserved for future applications yet to be defined.

The last 704 bits in any frame form a block of either sound or data (the two types of information are not mixed within one frame). One frame contains 64 sound samples (D₁ to D₆₄). The structures of a stereo sound frame and a mono sound frame are shown in Figs. 3a and 3b respectively.

In stereo mode (AC: C₁=C₂=C₃=0), the odd-numbered samples convey the A-channel, and the even-numbered samples the B-channel. Thus, 32 samples of each channel are transmitted in every frame.

If two independent mono sound channels, M₁ and M₂, are transmitted (AC: C₁=0; C₂=1; C₃=0), M₁ is transmitted in odd-numbered frames, and M₂ in even-numbered frames.

If one mono sound channel is transmitted (AC: C₁=1; C₂=0; C₃=0), it is contained in odd-numbered frames, and data are transmitted in even-numbered frames.

Thus, for mono sound signals, each frame with sound information in it contains 64 consecutive sound samples, which will span two complete companding blocks, shown as blocks n and (n+1) in Fig. 3. No format has yet been defined for data information.

Sound signals

Sound signals are sampled at 32 kHz, and coded initially with a resolution of 14 bits per sample. Near-instantaneous companding is used to reduce the number of bits per sample from 14 to 10, and one parity bit is added to each 10-bit sample word for error detection and scale-factor signalling purposes.

The companding process forms the 14-bit digital samples corresponding to each of the sound signals into blocks of 32. All of the samples in each 1-ms block are subsequently coded, using a 10-bit 2's complement code, to an accuracy determined by the magnitude of the largest sample in the block, and a scale factor code is formed to convey the degree of compression to the receiver. Figure 5 illustrates the coding of companded sound signals.

Prior to compression, a pre-emphasis to CCITT recommendation J17 (Ref. 2) is applied to the sound signals, either by using analogue pre-emphasis networks before digitisation, or by using digital filters with the digital signals.

For stereo transmissions, the signals of the left and right sound channels are sampled simultaneously. The Channel-A samples convey the left-hand (L) sound signal, and the Channel-B samples the right-hand (R) sound signal.

One parity bit is added to each 10-bit sound sample to check the six most-significant bits for the presence of errors. The parity group so formed is even (i.e., the module-2 sum of the six protected sample bits and the parity bit equals 0). Subsequently, the parity bits are modified to signal the 3-bit scale factor word associated with each sound signal block.

In addition to signalling the coding range, the scale factor signals seven protection ranges. This information may be used in the receiver to provide extra protection for the most significant bits of the samples. Table 2 shows the coding ranges and protection ranges associated with each 3-bit scale factor word. The five coding ranges indicate the degree of compression to which the block of samples has been subjected for the near-instantaneous companding process. The 3-bit scale factor R₂-R₁-R₀ associated with each 32-sample sound block is conveyed by modification of the parity bits (see Fig. 5).

When a stereo sound signal is being transmitted, FE₁ (facteur échelle; scale factor) is the scale-factor word R₂A-R₁A-R₀A associated with the 'A' samples, and FE₂ the scale-factor word R₂B-R₁B-R₀B associated with the 'B' samples. If P₀ is the parity bit of the iₚ sample, this is modified

Fig. 4. Pseudo-random sequence generator (PRSG) for spectrum shaping (energy dispersal scrambling).
to \( P' \), by modulo-2 addition of one bit of one of the scale-factor words according to the following relationship:

\[
P'_i = P_i \oplus R_{2A} \text{ for } i = 1, 7, 13, 19, 25, 31, 37, 43, 49
\]

\[
P'_i = P_i \oplus R_{4A} \text{ for } i = 3, 9, 15, 21, 27, 33, 39, 45, 51
\]

\[
P'_i = P_i \oplus R_{1A} \text{ for } i = 5, 11, 17, 23, 29, 35, 41, 47, 53
\]

\[
P'_i = P_i \oplus R_{3A} \text{ for } i = 2, 8, 14, 20, 26, 32, 38, 44, 50
\]

\[
P'_i = P_i \oplus R_{1B} \text{ for } i = 4, 10, 16, 22, 28, 34, 40, 46, 52
\]

\[
P'_i = P_i \oplus R_{3B} \text{ for } i = 6, 12, 18, 24, 30, 36, 42, 48, 54
\]

When a mono signal is being sent, FE1 is the scale-factor word \( R_{2A} \) associated with the first block of 32 samples in the frame, and FE2 is the scale-factor word \( R_{2A+1} \) associated with the second block of 32 samples in the frame. As in the case of stereo sound, the parity bit of the first sample, \( P_i \), is modified to \( P'_i \) by the scale-factor coding range information stored for the purpose of error concealment.

The control information described in Section 6.2.3 of Ref. 1 (Chapter 3, Part 3) is not used. However, other information could be transmitted by the same means, i.e., two information bits such that one modifies samples 55 to 59, and the other samples 60 to 64. NICAM receivers should be designed to take account of this facility.

### Modulation parameters

The characteristics of the AM vision (vestigial sideband) and FM sound are defined in the UK specification for PAL system transmissions (Ref. 3), with the exception that the FM sound carrier power is 10 dB down with respect to the vision carrier, instead of 7 dB. In the case of PAL system B/G transmissions, the definitions given in CCIR Report 624-3 apply.

The NICAM signal in the baseband is classified as differentially encoded quadrature phase shift keying (DQPSK or 4-phase DQPSK). This is a four-state phase modulation system in which each change of state conveys two data bits. The input data stream at the modulator is differentially encoded. This is done in two steps: (1) serial to two-bit parallel conversion, and (2) coding of the transmitted phase changes. The amounts of the phase changes are given to the four possible values of the input bit pairs \((A_n, B_n)\) are shown in Table 3.

Table 3. DQPSK carrier state changes.

<table>
<thead>
<tr>
<th>Input bit-pair</th>
<th>Amount by which the carrier changes phase</th>
</tr>
</thead>
<tbody>
<tr>
<td>( A_n )</td>
<td>( B_n )</td>
</tr>
<tr>
<td>0</td>
<td>180°</td>
</tr>
<tr>
<td>0</td>
<td>270°</td>
</tr>
<tr>
<td>1</td>
<td>90°</td>
</tr>
<tr>
<td>1</td>
<td>0°</td>
</tr>
</tbody>
</table>

Thus, the carrier phase can be at one of four rest-states which are spaced at intervals of 90° apart (Fig. 6a). An input bit-pair will shift the carrier phase into a different rest-state by the amount of phase change assigned to that particular value of bit-pair. The transmitted phase changes and resulting carrier rest-states for the input bit-pair sequence 00, 01, 11 and 01 are illustrated in Fig. 6b. In the receiver, the transmitted datastream may be unambiguously recovered by determining the
phase-changes between one bit-pair and the next.

It was already mentioned that spectrum-shaping techniques are applied to keep the bandwidth of the NICAM signal in the baseband within limits. For best performance in the presence of random noise, the amplitude-frequency response of data spectrum-shaping filters at the receiver should be identical to that at the transmitter. The target amplitude frequency response, $H_T(f)$, is given by

$$H_T(f) = \begin{cases} \cos \frac{\pi f}{2} & \text{if } 0 \leq f \leq \frac{1}{t_s} \\ 0 & \text{if } f > \frac{1}{t_s} \end{cases}$$

where $t_s = \frac{1}{364,000}$ s

and the filter has a constant group delay for all frequencies $< 1/t_s$. The filter made on the basis of the above transfer characteristic has a 100% cosine roll-off (for PAL systems B and G a filter with 40% cosine roll-off is required).

In the UK, the NICAM subcarrier is located at 6.552 MHz above the frequency of the vision carrier (see Fig. 1). This frequency is obtained by multiplying the transmitted bit-rate (728 Kbit/s) by 9. In countries where PAL system B or G is used, the subcarrier frequency is 5.850 MHz.

**NICAM decoder concepts**

Among the IC manufacturers that have developed NICAM processors for use in commercial-grade receivers are ITT Semiconductors of Germany, and Micronas, Inc. of Finland. A decoder based on ICs from the latter manufacturer is described elsewhere in this issue.

ITT Semiconductors have integrated their NICAM processors, the MSP2400 and MSP2410, into the Digit 2000 TV system. Figure 7 shows the block diagram of the ITT approach. Apart from the MSP2400 or MSP2410, two additional ICs are required, the AMU2481 and the ACP2371. Remarkably, the MSP2400 has a digital filter to extract the NICAM information from the baseband spectrum (0 to 9 MHz). This is in contrast to the Micronas circuit (Fig. 8), which uses a conventional L-C bandpass filter tuned to 5.84 MHz (PAL system B/G) or 6.552 MHz (PAL system I). The ITT circuit has a number of interesting options such as multistandard sound processing and automatic standard recognition and switching. The configuration as shown in Fig. 7 is capable of handling mono FM, stereo FM (the German dual-carrier system) and all NICAM modes (a special version of the ACP2371 is available for satellite TV sound). The disadvantage of the ITT circuit is, however, that it can not work without control software, and this is where the Micronas system has the edge on the ITT system: it can work 'stand alone', and offers an operational way of computer control.

**Sources:**

1. NICAM-728: specification for two additional digital sound channels with System-I television.

**References:**

2. CCITT Red Book, Volume III. Fascicle III.4: Transmission of sound-programme and television signals, recommendation J.17 'Pre-emphasis used on sound-programme circuits'.
The decoder described here is aimed at the experienced radio and TV enthusiast who wants to upgrade an existing TV set or video recorder with NICAM digital stereo sound. Suitable for PAL TV systems 'I' (UK) and 'B/G' (Scandinavia, Belgium, Spain and others), the decoder is a compact and simple to control circuit that can either be built as a set-top extension, or incorporated into a TV set.

**Design by Rob Krijgsman PE1CHY**

This decoder is based on a NICAM chip set developed by Micronas Inc. of Finland. The set consists of the MAS7A101 QPSK demodulator, the MAS7D102 NICAM decoder, and the MAS7A103 dual D-A converter. The chip set allows two high-quality audio channels (stereo or dual-language mode) to be recovered from a NICAM signal at 5.85 MHz or 6.552 MHz (if broadcast, and depending on the PAL system used) in the TV baseband spectrum. All that is needed to be compatible with either of the two PAL systems is to fit the correct input filter, a jumper and a quartz crystal for the demodulator clock.

**Three ICs**

As shown by the block diagram in Fig. 1, the upper part (say, above 5 MHz) of the TV baseband spectrum is first filtered to extract the NICAM signal centred around 5.85 MHz (system B/G) or 6.552 MHz (system I). The insertion loss of the band-pass filter is compensated by an amplifier.

**MAS7A101 QPSK demodulator**

The NICAM signal is applied to the MAS7A101 QPSK demodulator IC. This is a pretty complex integrated circuit, whose internal architecture is given in Fig. 2. The QPSK signal at the input is buffered before it is applied to a multiplier circuit which consists of analogue switches. The switches are opened and closed by a signal derived from a phase-controlled quartz crystal oscillator.

The crystal frequency equals four times the NICAM subcarrier frequency, i.e.,

\[
5.850 \times 4 = 23.400 \text{ MHz}
\]

for PAL systems B and G, or

\[
6.552 \times 4 = 26.208 \text{ MHz}
\]

for PAL system I. The quartz oscillator is locked to the received NICAM signal by means of a PLL. The demodulated signal is taken through a switchable low-pass filter, and subsequently split into two.

One signal is sent to a second PLL which serves to recover the 728-kHz NICAM bit clock from the demodulated signal. The crystal-controlled VCO in this PLL operates at eight times the NICAM bit clock, or 5.824 MHz. This VCO also provides the central clock signal for the other ICs in the decoder.

The other demodulated signal is sent to a slicer circuit where it is converted into a bi-

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**Front Cover Project**

Unsullered Video

QPSK Demodulator

MAGAZINE

Dual DAC

MAS7A103

Audio Outputs

Mode Control

Mode Display

System I = UK
System B/G = Scandinavia

Fig. 1. Block diagram of the decoder.
ary digital signal. The recovered clock signal and the binary signal are available at the corresponding outputs of the MAS7A101.

MAS7D102 NICAM decoder
The MAS7D102 NICAM decoder (Fig. 3) uses the recovered NICAM bit clock to tackle the decoding proper of the bitstream supplied by the QPSK demodulator. The decoding process involves quite a lot: descrambling, de-interleaving, error detection and correction, and reconstruction of the original 14-bit sound samples in both channels. The MAS7D102 can be programmed or wired to supply digital output signals suitable for one of three different bus systems: the I2S-bus (Philips), the S-bus (ITT), or the DAC-bus (Toshiba). Many functions of the IC can be controlled either via an I2C link, or by means of external hardware. The latter option is exploited here, and has the advantage of obviating a microcontroller and a dedicated control program.

With reference to the IC architecture shown in Fig. 3, it is seen that the digital signal supplied by the QPSK demodulator is split into two. One signal is fed to a synchronisation logic section where the FAW (frame alignment word) is detected and extracted. The FAW is never scrambled. The other copy of the digital signal is sent to the descrambler circuit, which serves to counteract the energy dispersal (spectrum-shaping) scrambling applied at the transmitter. When the decoder chip is first switched on, it uses the standard descrambling initialisation word ‘11111111’, which enables reception of non-encrypted NICAM broadcasts. External hardware is required to be able to change the initialisation word (or 'seed') 'on the fly' when the system is used for reception of Pay-TV transmissions using encrypted NICAM audio.

Returning to the operation of the MAS7D102, the control information bits C1-C4-C5-C6 are extracted from the datastream. These bits enable the receiver to determine the type of programme material: i.e., dual-language or stereo. The decoded control bits are available in an I2C register as well as on an output port. The latter allows a simple display to be connected that indicates the receiver mode. The sound samples are fed to the de-interleaver, and from there to the error detection/correction circuit. Finally, they are de-companded to their original 14-bit resolution, and fed to the output of the IC according to the selected signal format (I2S-bus, S-bus or DAC-bus). The format selection is effected via the I2C bus, or via logic levels applied to the configuration (CONFIGx) pins, which in addition allow you to select between mono-A or mono-B during dual-language broadcasts. The functions of all registers contained in the MAS7D102, and the configuration options that can be set in hardware, are given on page 41.

MAS7A103 dual DAC
This IC converts the 14-bit sound samples furnished by the decoder into two analogue audio signals. Since the output datastream of the decoder IC is multiplexed, the first task of the DAC is to extract and separate the information that belongs with each channel. Next, the two digital signals are converted into analogue ones by R-2R ladder networks. These supply output currents rather than voltages, so that two external opamps are required to obtain audio signals that can be fed to an amplifier. Before that can be done, however, the audio signals need to be taken through a 15-kHz low-pass filter to remove the residue of the 32-kHz sampling signal.
Fig. 4. Circuit diagram of the NICAM decoder.
This filter takes us back to the block diagram in Fig. 1, with the final remark that J17 de-emphasis is applied on the audio signals.

**Practical circuit**

After studying some of the background theory on NICAM (to be found elsewhere in this issue), and having acquired samples and datasheets of the NICAM chip set, the author set out to work, and was able to design and build a simple NICAM decoder. 

A practical circuit was designed and built, which was tested with the aid of NICAM broadcasts received from the Belgian national TV station BRT (these broadcasts were experimental at the time, and are currently regular). The BRT transmits NICAM-728 according to PAL B/G standard, with the PAL TV system used in your country.

The final result is an uncluttered circuit shown in Fig. 4. The unfiltered video signal, taken from a suitable point in the TV tuner (more about this further on), is applied to the input of a four-section bandpass filter tuned to 23.400 MHz (note that different types are required for systems B/G and system I). Its insertion loss lies between 8 dB and 10 dB. This is compensated by amplifier IC1, whose gain can be set as required with the aid of preset P1 to give a signal level of 200 to 800 mVp-p at the input of the QPSK demodulator, IC2.

As indicated in the diagram, the frequency of quartz crystal X1 is determined by the PAL TV system used in your country. Depending on the characteristics of the crystals used in positions X1 and X2, the exact values of C2-C6 and C11-C15 may have to be changed from those shown in the circuit diagram. Given that the quartz crystals probably have to be cut to order (the frequencies being non-standard as far as we have been able to find out), some experimenting may be required to obtain the correct oscillator frequencies.

The QPSK demodulator, IC2, supplies the recovered 728-kHz bit clock, the digital NICAM signal, and the 5.824-MHz system clock to the decoder, IC3. An R-C network, R16-C21, resets the demodulator and the decoder ICs at power-on.

Mode selection is effected with configuration bits config1 and config2. The available options are mono-2, mono-1 and mono-1/2 (dual language mode). The logic bit combinations required for these settings are supplied by IC6, IC9 and three push-buttons, S1, S2 and S3. The combination of these parts forms a kind of three-position flip-flop with a built-in latch function, a debounce circuit and an indication (on five LEDs). Capacitor C25 ensures that the 'mono-2' mode is automatically selected at power-on.

Diodes D12, D13 and D14 provide the required logic levels at the CONFIG inputs of the decoder IC. LEDs D10 and D11 indicate the currently transmitted mode: dual-language (mono-1/2) or stereo. This indication can not be changed by pressing the MODE switches.

Like the QPSK demodulator IC, the NICAM decoder, IC3, is used in a standard application circuit as suggested by the manufacturer. Similarly, few surprises are found in the link to the dual DAC, IC4, and the subsequent two-stage opamp-based current-to-voltage converters/amplifiers. It will be noted, though, that the opamps work from a symmetrical (+12 V/-15 V) supply. The gain of IC6 and IC7 in the right (R) output channel is set such that the loss introduced by the 15-kHz low-pass filter, FC2, is overcome whilst ensuring an audio output level that is compatible with other equipment driving a amplifier 'line' input. The same goes, of course, for the corresponding components in the left (L) channel. The low-pass filters are, again, ready-made pre-aligned modules from Toko. Here, we are dealing with two A288ULV-500SN three-section L-C filters (the designer apologises for the type numbers). Finally, the J17 de-emphasis networks in the right and left audio channels are formed by R35-C40 and R37-C41 respectively. The outputs of the NICAM decoder are capable of driving amplifier 'line' inputs.

**Construction**

First, cut the printed circuit board (Fig. 5) into three to separate the power supply...
Fig. 5. Track layout and component mounting plan of the single-sided PCB for the NICAM decoder.
board, the decoder board and the keyboard. The population of these boards is entirely straightforward, and should not present problems. It is recommended to use IC sockets. The voltage regulators are bolted straight to the power supply board, and do not need heat-sinks. The fuse is fitted in a holder with a protective plastic cap. On the decoder board, the section with the blue (or red) core in the QPSK bandpass filter, Fl, is at the side of the NICAM decoder chip, IC2. The keyboard/display section of the printed circuit board has on it three Digitast press-keys with a built-in LED. The front panel of the enclosure for the NICAM decoder (if used) must be cut and drilled to allow the push-buttons and the two LEDs to the right of the board to protrude—more about this further on.

For an initial test, the completed boards are interconnected. Switch on, and check the presence of the correct supply voltages at a number of points. Press the keys and see if the associated LEDs light. If this works all right, stop, and start thinking very hard about

**Finding the input signal**

The present NICAM decoder is intended as an upgrade for existing TV sets, set-top TV tuners or video recorders. In nearly all cases, this equipment will have to be opened or modified to find or create a point where the NICAM signal can be 'tapped' and fed to the decoder. The following points should be taken into account:

1. Opening your TV set or VCR in most cases voids your warranty on this equipment.
2. The chassis of most older TV sets is connected directly to the mains. Never work on such a TV set without using an isolating transformer.
3. Make sure you have the service documentation (or at least a circuit diagram) of the equipment.

The intrepid among you should be looking for a point at the input of the sound demodulator where a signal is available that contains as little video information as possible. In most cases, the input signal of the main FM demodulator (5.5 MHz for system-B/G, or 6.0 MHz for system-I) is taken through a ceramic band-pass filter to suppress the components in the video spectrum. In general, it is best to tap the signal ahead of this filter. The minimum level of the signal to be fed to the NICAM decoder is about 50 mV. In all cases, the load presented by the input of the NICAM decoder should be as small as possible. This may require an emitter follower to be fitted as discussed below.

A little more complex, but certainly more convenient as far as the filtering is concerned, is a TV set or VCR with a so-called quasi-parallel sound demodulator system. The designer used his HR-S5000E video recorder from JVC to supply the NICAM signal. After studying the service documentation that came with the VCR, it...
was decided to try the output signal of an emitter follower located between the 'sound IF' output and the input of the 5.74 MHz ceramic filter fitted for the German 'dual-language' demodulator. The search for this emitter follower, Q11, was complicated by the fact that it happened to 'reside' between three pretty large circuit diagrams. Figure 6 shows essentially what has been added to the VCR: one resistor, a coupling capacitor and a 'phono' socket do a perfect job.

As already mentioned, an emitter follower may have to be used to prevent the input signal of the sound demodulator disappearing when the NICAM decoder is connected. One of the circuits shown in Fig. 7 will be adequate. The first, Fig. 7a, may be used when the signal is superimposed on a d.c. level between 0.3 and 0.7 times the supply voltage. The other, Fig. 7b, has an input coupling capacitor, and is used in all other cases. Remember, you are dealing with signals of 5 MHz and higher here, so keep component wires as short as possible.

Demodulator input level
Switch on the NICAM decoder, and tune the

Fig. 7. Emitter followers for NICAM signals on a relatively small d.c. component (7a), and NICAM signals on a d.c. component so large that a.c. coupling is required (7b).

Testing
The input impedance of the NICAM decoder is fairly high: about 900 Ω. This means that conventional coax cable with an impedance of 50 Ω or 75 Ω can not be used unless its length remains below 50 cm or so. Longer cables of either type will cause reflections and serious mismatches, resulting in attenuation of the NICAM subcarrier. If you can not go round the use of a relatively long, low-impedance, coax cable between the TV set and the NICAM decoder, be sure to fit a terminating resistor across socket Kt. This resistor prevents reflection and high-frequency loss to some extent. When a 50-Ω cable is used, fit a 52.9-Ω terminating resistor, and change R1 into 444 Ω. Similarly, when a 75-Ω cable is used, terminate it with 81.8 Ω, and change R1 into 431 Ω. In some cases, ordinary screened cable as used with audio equipment, or car radio coax cable (if you can get it), is the best alternative. In any case, do not fit BNC or similar low-impedance RF sockets at the TV side and the decoder input. On the prototype we used an insulated 'phono' (RCA-style) socket for chassis mounting. An insulated socket is required to prevent an earth loop between the analogue and digital ground rails.
The bus format can be selected either by applying logic levels to pins Config4 and Config3, or by programming control bits Config4 and Config3 via the I²C microprocessor interface.

<table>
<thead>
<tr>
<th>Config 4</th>
<th>Config3</th>
<th>DAC bus format</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>0</td>
<td>High-Z</td>
</tr>
<tr>
<td>0</td>
<td>1</td>
<td>S-bus</td>
</tr>
<tr>
<td>1</td>
<td>0</td>
<td>I²S bus</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>Toshiba DAC bus</td>
</tr>
</tbody>
</table>

The pins Stereo, Mono1 and Mono2 are active-low outputs that indicate the current NICAM transmission mode.

<table>
<thead>
<tr>
<th>Stereo</th>
<th>Mono1</th>
<th>Mono2</th>
<th>Type of transmission</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>1</td>
<td>1</td>
<td>Stereo signal</td>
</tr>
<tr>
<td>1</td>
<td>0</td>
<td>1</td>
<td>Dual language transmission</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>0</td>
<td>One mono sound channel and 352 Kbit/s data channel</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>1</td>
<td>No sound signal. Transparent 704 Kbit/s data transmission, or no NICAM encoded transmission</td>
</tr>
</tbody>
</table>

During dual-language transmissions, the main language selection is controlled by input pins Config2 and Config1.

<table>
<thead>
<tr>
<th>Config2</th>
<th>Config1</th>
<th>DAC bus</th>
<th>Sound sample order</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>0</td>
<td>I²S/Toshiba</td>
<td>M1 M1</td>
</tr>
<tr>
<td>0</td>
<td>1</td>
<td>I²S/Toshiba</td>
<td>M1 M2</td>
</tr>
<tr>
<td>1</td>
<td>0</td>
<td>I²S/Toshiba</td>
<td>M2 M1</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>ITT</td>
<td>M1 M1 M1 M1 M1</td>
</tr>
<tr>
<td>0</td>
<td>0</td>
<td>ITT</td>
<td>M1 M1 M2 M2</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>ITT</td>
<td>M2 M2 M2 M2</td>
</tr>
</tbody>
</table>

The decoder has two addresses on the I²C bus. Address 4E (hex) is for writing to the decoder, and address 4F (hex) for reading from the decoder. There are three status registers (read) and three control registers (write) that can be accessed. The three control registers can be addressed individually by the two most significant bits of each control word. The three status registers can be addressed as a complete set only.

<table>
<thead>
<tr>
<th>Control register</th>
<th>D7</th>
<th>D6</th>
<th>D5</th>
<th>D4</th>
<th>D3</th>
<th>D2</th>
<th>D1</th>
<th>D0</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>0</td>
<td>1</td>
<td>0</td>
<td>1</td>
<td>MuteS</td>
<td>MuteA</td>
<td></td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>0</td>
<td>1</td>
<td>1</td>
<td>0</td>
<td>Config4</td>
<td>Config3</td>
<td></td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>1</td>
<td>Da</td>
<td>Db</td>
<td>Dc</td>
<td>Pa1</td>
<td>Pa0</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

- Test1 and Test2 are reserved for test purposes, and must be set low.
- The MuteS control bit mutes sound output. Active high.
- The MuteA control bit mutes sound output and resets the synchronisation of the decoder completely. Active high.
- The Reset control bit resets the decoder completely. Active high.

The function of the Da, Db and Dc control bits is to define external ports Pa, Pb and Pc as inputs or outputs, as shown below.

<table>
<thead>
<tr>
<th>Da</th>
<th>Db</th>
<th>Dc</th>
<th>Pa1</th>
<th>Pa0</th>
<th>Pb0</th>
<th>Pc0</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>1</td>
<td>0</td>
<td>out</td>
<td>out</td>
<td>out</td>
<td>out</td>
</tr>
<tr>
<td>0</td>
<td>1</td>
<td>1</td>
<td>out</td>
<td>out</td>
<td>out</td>
<td>in</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>0</td>
<td>out</td>
<td>out</td>
<td>in</td>
<td>out</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>1</td>
<td>in</td>
<td>in</td>
<td>out</td>
<td>in</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>0</td>
<td>in</td>
<td>in</td>
<td>out</td>
<td>in</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>1</td>
<td>in</td>
<td>in</td>
<td>in</td>
<td>in</td>
</tr>
</tbody>
</table>

The status registers of the MAS7D102 have the following structure:

<table>
<thead>
<tr>
<th>Status register</th>
<th>D7</th>
<th>D6</th>
<th>D5</th>
<th>D4</th>
<th>D3</th>
<th>D2</th>
<th>D1</th>
<th>D0</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Osn</td>
<td>C1</td>
<td>C10</td>
<td>C4</td>
<td>C3</td>
<td>C2</td>
<td>C1</td>
<td>C0</td>
</tr>
<tr>
<td>2</td>
<td>Ser10</td>
<td>Ser9</td>
<td>Mute</td>
<td>Test1</td>
<td>Pa1</td>
<td>Pa0</td>
<td>PB0</td>
<td>Pc0</td>
</tr>
<tr>
<td>3</td>
<td>Ser8</td>
<td>Ser7</td>
<td>Ser6</td>
<td>Ser5</td>
<td>Ser4</td>
<td>Ser3</td>
<td>Ser2</td>
<td>Ser1</td>
</tr>
</tbody>
</table>

- The Osn status bit goes high when the decoder is not synchronised.
- C10 and C1 are the two Ci bits extracted from each NICAM frame.
- C4-C0 are the C bits associated with the current NICAM transmission, and they indicate the mode as shown below.

<table>
<thead>
<tr>
<th>C1</th>
<th>C2</th>
<th>C3</th>
<th>NICAM transmission mode</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>0</td>
<td>0</td>
<td>Stereo transmission</td>
</tr>
<tr>
<td>0</td>
<td>1</td>
<td>0</td>
<td>Dual language transmission</td>
</tr>
<tr>
<td>1</td>
<td>0</td>
<td>0</td>
<td>One mono channel plus data transmission</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>0</td>
<td>One 704 Kbit/s data channel</td>
</tr>
</tbody>
</table>

- C0 is the FrameFlag bit that indicates the super frame pattern of the NICAM transmission.
- C4 is the Reserve Sound Switching flag, which goes high when the FM mono signal carries the same programme as the digital stereo signal.
- The Mute status bit goes high to indicate that the decoder has been muted for some reason.

- TestS is a test status indication bit reserved for test purposes.
- Pa1 and Pa0, Pb0 and Pc0 indicate the status of the corresponding external pins, when they are configured as input ports.
- The Ser10-Ser1 bits show the value contained in the sample error counter. This counter is incremented whenever an erroneous sample is detected. The control processor can read the error count at suitable time intervals, and take decisions depending on the error rate.

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ELEKTOR ELECTRONICS MAY 1992
TV set or the VCR to a station transmitting NICAM sound. Use an oscilloscope to check the signal level at pin 3 of the QPSK demodulator IC. The level should be between 200 mV_pp and 800 mV_pp. If necessary, adjust preset P1 to achieve a level of about 500 mV_pp.

QPSK demodulator PLL adjustment
Connect the scope to pin 11 of the QPSK demodulator IC. You should see a so-called 'eyes' waveform (which may be very difficult for the scope to trigger on). Adjust P1 so that the tops of the waveform are just below the supply voltage; i.e., they are just not clipped. This gives a signal level of about 5 V_pp. Move on to pin 7 of IC2. This supplies the error voltage of the demodulator PLL. It is a fairly 'messy' signal superimposed on a direct voltage, which will look like a broad band on the scope. Tune to a non-NICAM station, and back to the NICAM station again, to see how the PLL responds by locking on to the NICAM signal. For best performance of the PLL, the d.c. component in the error signal should be at about half the supply voltage, i.e., 2.5 V. When it is too close to either 0 V or +5 V, change the crystal matching capacitors, C1 and C3, until the centre of the band is at about 2.5 V. Increase the capacitor values (to 22 pF or 27 pF) when the d.c. component is too low, and decrease them (to 15 pF or 12 pF) when the d.c. component is too high. Try to get as close to 2.5 V as you can. The exact oscillator frequency will be very difficult to measure at pin 5 of IC2 because the impedance is high locally. This means that any capacitive load, however small, formed by a test probe will detune the crystal oscillator to some extent.

Clock recovery PLL adjustment
The 5.824-MHz PLL for the NICAM clock signal recovery is adjusted in a similar manner to the QPSK PLL as discussed above. Connect the scope to pin 18 of IC7, and check that the error voltage has a d.c. component of about 2.5 V. If not, change the values of C14 and C15. It will be found that this error voltage is much 'cleaner' than the one used for controlling the first PLL. If the second PLL frequency is correct, pin 23 of the demodulator IC supplies a clock signal of 728 kHz, which is easily measured with a frequency meter.

That completes the adjustment of the NICAM decoder. If you have not already done so, connect a stereo amplifier to the outputs and enjoy the programme!

Fig. 9. Completed printed circuit boards: main decoder board, keyboard and PSU.

Finishing touch
Some of you may want to fit the decoder permanently inside a TV set, while others may want to use it as a self-contained unit.

The prototype of the decoder was housed in an aluminium enclosure Type 55205 from Telei. The decoder and supply boards were fitted on a perspex plate that could be slid horizontally into the railings provided along the inside of the front and rear panels.

The keyboard PCB and the mains switch are fitted on to the front panel, for which a ready-made self-adhesive foil is available. This foil is used as a template to determine the locations of the holes to be cut in the front panel. A jig-saw is used to cut the rectangular clearances for the mains switch and the three push-buttons.

The keyboard PCB is mounted on four screws of which the (countersunk) heads are glued to the inside of the front panel. Plastic stand-offs are used to fit the PCB at the right distance behind the front panel.

The rear panel is drilled to hold the mains socket, the NICAM input socket and the two audio output sockets.

Conclusion
The NICAM decoder described here has been in use for some time now, and provides excellent stereo sound on broadcasts received from BRT1 and BRT2. Regrettably, the unit could not be tested in the UK, although suitable components (a 6.552-MHz QPSK bandpass filter and a 26.208 MHz quartz crystal) were available.

Although the construction and adjustment of the unit are fairly simple, finding a suitable input signal may be daunting if you have little experience in TV and VCR technology. We feel, therefore, that it is fair to warn beginners not to undertake this project until a dedicated TV tuner is available, which will be described in a future issue of Elektor Electronics.

Postscript for advanced users
As already mentioned, the MAS7D102 NICAM decoder has optional I2C control, which may be used to access most of the internal registers. The SDA and SCL inputs of this IC are TTL-compatible, and may be connected to an I2C bus via appropriate interfaces. If you have a PC available fitted with an I2C interface (Ref. 1), you may use the information given in the MAS7D102 inset to implement software control on the NICAM decoder.

The descrambler on board the MAS7D102 can be loaded with a descrambling key other than the standard 'seed' used for non-encrypted NICAM broadcasts. Changes to the scrambling keys must occur synchronously at the transmitter and the receiver(s). The NICAM decoder IC provides a serial data input, Dsdata (pin 6), and a clock input, Dsclk (pin 5) to access an internal shift register. This register contains the descrambler key that is loaded in parallel into the descrambler one per frame. The shift register contents can be updated at any time with a maximum clock rate of 5 MHz. The time interval between the falling edge of the Ngate signal (pin 39) and the rising edge of the Agate signal (pin 38) is not allowed for descrambler key updating. During this interval, Dsclk (pin 5) must be held static. Output signals C0 (pin 35), Agate and Ngate may be useful for synchronisation purposes.

Happy listening!

The co-operation of Mr. Matti Antman of Micronas Inc., Mr. Peter de Vroome of Arco-bel b.v., and our photo model Miss Diny Erven is gratefully acknowledged.

Reference:

ELEKTOR ELECTRONICS MAY 1992
An ELV design

The audio-video processor is best compared to the control amplifier in an audio system. That, too, gives a choice of input signals. Where required, the signal standard may be changed. The quality of the signal can be modified in a manner comparable to tone control. Finally, the signal is output in a number of ways (as in tape and line outputs).

The design of the processor can be seen in the block diagram in Fig. 1. In the first instance, this diagram will be analysed stage by stage; this will at the same time give a sort of user instruction. The technical aspects will follow naturally when the various circuit diagrams are discussed.

Inputs and outputs

At the rear of the processor are no fewer than 16 different connectors—see Fig. 2. Which of these are required in any given situation depends on the type of termination of the relevant cable and on the available signal.

Video 1/6 is an input with dual function via a SCART* connector (also called Euroconnector). When used as Video 1 input, it is fed with a CVBS (Chroma, Video, Blanking, Synchronization) signal, which is sometimes just called composite video signal.

When used as Video 6 input, it accepts signals from an S-VHS recorder or camera. These signals consist of two components: chroma for colour information and VBS (black-and-white) for brightness.

The audio signals associated with these inputs are also applied via the Euroconnector.

Video 2 is a SCART* input for CVBS and RGB (Red, Green, Blue) signals; selection of either is effected by a slide switch. When the switch is in position CVBS, it is possible, with the aid of a computer and a genlock, to mix CVBS and RGB signals. It is planned to publish the design of a genlock later this year.

Video 3 is a BNC input for a CVBS signal. The associated audio signal is input via the two audio sockets next to the BNC socket.

Video 4 is a second SCART* socket that can also be used to input a CVBS signal. Moreover, it provides CVBS and RGB output signals. This arrangement is intended primarily for the standards conversion. To that end, the sockets should be linked to the SCART socket on the television receiver. The tuner of the receiver will then provide the CVBS signal, whether NTSC, SECAM or PAL, to the processor. The processor translates this signal into an RGB signal and sends it to the TV receiver. At the same time, the blanking line is switched to arrange the TV receiver displaying the signal at the RGB inputs, that is, the converted signal, instead of that from its integral tuner. There is only one but not all TV receivers, particularly older models, have RGB inputs on the SCART socket.

Video 5 is a mini DIN socket for inputting S-VHS signals. The associated audio signals are fed to the processor via audio sockets.

Video 7 is a mini DIN socket for outputting S-VHS signals. Again, the associated audio signals are output via standard audio sockets.

Audio 1-8 are the audio inputs and outputs associated with the correspondingly numbered video inputs and outputs.

The Type AVP300 audio-video processor is a multi-standard equipment that can be used almost anywhere in the world. It can translate between the three television standards (PAL, NTSC, and SECAM), hop from one type of signal to another (S-VHS, Hi-8, RGB, CVBS), and enables audio and video signals to be modified: video signals as regards colour saturation, contrast, brightness and the balance between red, green and blue, and audio signal(s) in respect of tone, balance and volume. Moreover, it has a (limited) facility for mixing signals.
Audio 11 is a headphone output via which the signals before and after the control amplifier can be heard.

Port is an 8-way DIN socket, via which external equipment can be used to control various functions of the processor. It will enable, for instance, the inputting of yet-to-be-developed video effects in the future without any further work on the processor.

**Standards conversion**

The processor can handle video signals of the following standards: PAL, SECAM, NTSC 3.58 MHz and NTSC 4.43 MHz. It is able to recognize these standards automatically and modify the chroma-VBS separation filter accordingly. Which standard is recognized is indicated by LEDs. Selection of positive or negative video signals is effected manually by a switch at the rear of the processor.

Also at the rear panel are the switches for setting the standard of the output signal, which is PAL or NTSC. In the case of NTSC, a further selection must be made of the colour carrier (3.58 MHz or 4.43 MHz). In the case of PAL, the switch must be set to 4.43 MHz.

Furthermore, the processor can be used to transform a conventional TV receiver into a multi-standard model (for relevant connections, see under Video 4).

**Formats conversion**

Apart from video standards, the processor can convert each of the signal formats S-VHS, RGB and CVBS to either of the other two. This is largely a matter of choosing the correct input and output connectors. Note that the CVBS/RGB-in switch (next to the Video 2 input) must be set to CVBS if an S-VHS or CVBS signal is input, irrespective of to which socket.

**S-VHS to RGB:** S-VHS signals can be input via the Video 5 or Video 6 sockets; the input switch must be set accordingly. The RGB signal is available at output Video 4: the CVBS/RGB-out slide switch must be set to RGB.

**S-VHS to CVBS:** S-VHS signals are input via the Video 5 or Video 6 sockets. The output may be taken from Video 4 or Video 8; both of these may be used simultaneously. The associated switch near these outputs must be set to CVBS.

**RGB to S-VHS:** RGB signals are input via Video 2. The corresponding RGB/RGB switch must be set to RGB. If this switch is in position CVBS, it may be changed over to RGB by applying a voltage of 1–3 V to pin 16 (blanking). The S-VHS signal may be taken from Video 7 or Video 8 (not simultaneously). The switch associated with Video 8 must be set to S-VHS.

**RGB to CVBS:** RGB signals are input via Video 2. The CVBS signal may be taken from Video 4 or Video 8; both of these may be used simultaneously. The switch at both the outputs must be set to CVBS.

**CVBS to S-VHS:** CVBS signals may be input via Video 1, Video 2 or Video 3. The converted signal is available at Video 7 or Video 8.

**CVBS to RGB:** CVBS signals may be input via Video 1, Video 2 or Video 3. The converted signal is available at output Video 4; the associated switch must be set to RGB.

**Quality of converted formats**

Retention of quality during the conversion from one format to another is ensured by special stages similar to those found in modern television receivers. However, in the case of conversion of CVBS signals, there may be a slight loss of quality. This is because the stripping of the chroma information from the signal tends to be troublesome; this, together with the limited bandwidth of standard VHS recorders, is the reason that the chroma and VBS components of the signal are kept separated in S-VHS recorders. In the conversion to a CVBS signal, there is no loss of quality. This assumes, of course, that the TV receiver and recorder connected to the processor are of good quality.

**Controls**

A close look at the block diagram in Fig. 2 shows that the processor resembles a modern multi-standard television receiver less the RF and CRT sections. The video section is the largest and most interesting part of the processor. Input signals follow two paths to the video colour controller. RGB signals from Video 2 are fed directly to this stage, but S-VHS and
CVBS signals must be decoded first. Switching between the decoded signals and RGB signals is carried out electronically in the decoder. The control signal for that is the RGB blanking signal. If the CVBS/RGB switch is set to position RGB, the output signals of the decoder relating to S-VHS and CVBS components are switched off by the controller. This is why it was emphasized earlier on that this switch must be set to CVBS.

S-VHS and CVBS signals are first applied to an input selector, an electronic switch that is operated by pressing button MODE on the processor. S-VHS signals are taken from the switch directly to the chroma-VBS separation filter. CVBS signals are first passed through a switchable inverter to enable positive as well as negative video signals to be processed. Normally, the switch is set to negative.

Readers may well ask why S-VHS signals are applied to the chroma-VBS separation filter, since these components are already separated in this format. That is, of course, so and with S-VHS signals the filter therefore serves as a buffer only. If, however, a CVBS signal is applied, the filter separates the sub-carrier and colour signal from the black-and-white (VBS) signal. The filter has two settings: one for PAL, SECAM and NTSC with a colour sub-carrier of 4.43 MHz and the other for NTSC with a colour sub-carrier of 3.58 MHz. The setting is determined by the multi-standard decoder as soon as this stage has detected which TV standard is used. At the same time, the type of standard is indicated by LEDs.

Following the filter, the chroma and VBS signals follow separate paths. To begin with, the VBS signal is applied to the sync pulse and sandcastle pulse generator, which derives new sync pulses from it. Separate line and field sync pulses, as well as sandcastle pulses, are fed to the Video 2 input. Sandcastle pulses indicate the various stages of the line, field and blanking pulses by voltage levels.

The sandcastle pulse is also applied to the video colour decoder, the multi-standard decoder and the CCD (Charge Coupled Device) delay line to ensure that these stages perform their functions at the right moment.

The sync pulse and sandcastle pulse generator also provides a composite sync (BS or Blanking/Synchronization) signal. This is used to re-render the picture signal, after it has been processed, into a VBS or CVBS signal, and also serves as sync signal for the RGB output.

The black-and-white signal is applied not only to the sync pulse and sandcastle pulse generator, but also to the video colour controller. In its path there is a delay line that prevents it from arriving too early at the controller: the chroma signal is also delayed during decoding.

The chroma signal is fed to the multi-standard decoder, where it is demodulated (stripped of the 4.43 MHz or 3.58 MHz sub-carrier) and split into two colour-difference signals. To give the correct colour to the colour-difference signals, the decoder needs a reference, and this is provided by the NTSC phase preset. That control is not required with PAL or SECAM, because signals in those formats already contain a reference. With NTSC signals, the preset needs to be adjusted until the colours appear natural or as natural as possible.

From the multi-standard decoder, the two colour-difference signals are applied to a CCD delay line. There, the incoming picture line is compared with the previous picture line stored in the delay line. This arrangement enables the removal of any errors in the colour-difference signals.

The input signals have then been processed to the stage where they can be applied to the video colour controller—the heart of the audio video processor. Although the colour information has not been completely decoded at this stage, the video colour controller correctly converts the colour-difference signals to RGB (red, green and blue) and provides a new sound for the present picture. CVBS or CVBS is required, a modulator is needed: in Fig. 1, this is called PAL/NTSC encoder. That name already indicates that both PAL and NTSC signals can be provided. The standard of the output signals is determined with a switch. If that is set to NTSC, a second switch allows setting either of two frequencies for the chroma sub-carrier.

The audio section is arranged in a similar manner, but contains far fewer components. Audio signals are also input via switches, which are operated together with those for the video signals. In that way, the audio signal at the master potentiometer is always associated with the present video signal.

There are also two inputs for independent audio signals that can be mixed with the original sound. If the master potentiometer is turned off, the audio signals at these inputs can be used as a new sound for the present picture. CVBS or CVBS is required.

The control amplifier also has controls for high and low tones and balance, as well as a mono/stereo selector. Audio signals can be listened to with headphones. These signals may be taken either from the input or from the output of the control amplifier: selection is by means of a switch. This enables the audio input to be monitored with the fader 'off'.

Next month's instalment will describe the circuit of the video section.
1. Introduction

An interim standard for Digital Short Range Radio (DSRR) has been approved recently by the European Telecommunications Standards Institute (ETSI). The institute was established by the European Community to assist with the harmonization of equipment specification and frequency allocations in member countries. The DSRR interim standard is one of its first radio specifications. It will go forward for public enquiry and, in due course, become a standard applicable in every community state. There could be modifications, but the interim standard will probably remain the basis of the final specification. After the standard has been finalized, no modifications can be introduced by individual states. In time, DSRR should become a true European radiocommunications system.

DSRR is a major advancement in business radio and should reduce many of the present problems. There are explained further in Section 2. The interim standard has a number of requirements in regard to control signalling and protocols which, at present, are not in general use. Because of developments in digital signal processing, these should not create major problems for manufacturers in the various countries. However, it also proposes the use of two frequency bands of 888-890 MHz and 933-935 MHz. Known as the Low Band (LB) and the High Band (HB) respectively. The significance of two separate bands is explained in Section 3. This could result in difficulties, because EC countries are still far away from true frequency harmonization. Since the EC is committed to harmonization and the removal of trade barriers, it is considering a directive to all member states that, in order to reduce any problems in the introduction of DSRR, priority should be given to DSRR over other services in the proposed bands. At present, DSRR appears to be on target for wide use within a few years.

2. PMR channels

The channels are usually assigned for dual frequency operation, which means that a unit transmits and receives on different frequencies as in Fig. 1. If a base station has \( f_t \) and \( f_r \) as its transmit and receive frequencies, a mobile station operating to the base station must have \( f_t \) and \( f_r \) as its transmit and receive frequencies respectively. In the United Kingdom, equipment for this service must conform to the Performance Specification MPT1326. This is a fundamental specification and DSRR units must meet a similar specification—see Appendix 5. The difference is the method of operation and the use of the channels.

In the PMR (Private Mobile Radio) service, the main barrier to further development is the method of using the radio frequency spectrum. In assigning channels to users, there is insufficient spectrum for every user to have his own channel. That means that a number of users must share the same channel. This, in turn, causes a nuisance effect and loss of confidentiality since every message on a channel is detected by every receiver irrespective of the intended destination. Tone control, like EEA or ZVEI, can be used to minimize this effect. At the start of a message a transmitter sends a specific sequence of tones. Only the intended receiver can identify the sequence and is activated from the point of the operator. If an incorrect sequence, that is, a sequence for a different receiver, is detected, the receiver remains deactivated. Unfortunately, this process does not solve the problem of channel sharing. A common occurrence is that two users wish to use the channel at the same time. Therefore, each user must monitor the channel to ascertain that it is free before he can commence operation. Queues often occur despite the fact that many channels are regularly idle. DSRR should reduce this problem by making more efficient use of the spectrum. It should also improve user confidentiality without the need of secrecy operations, that is, encryption procedures.

3. DSRR frequency bands

There are two bands of 2 MHz in which channels are assigned with 25kHz channel spacing as follows:

<table>
<thead>
<tr>
<th>Channel</th>
<th>High band</th>
<th>Low band</th>
</tr>
</thead>
<tbody>
<tr>
<td>01</td>
<td>933.025 MHz</td>
<td>888.025 MHz</td>
</tr>
<tr>
<td>02</td>
<td>933.050 MHz</td>
<td>888.050 MHz</td>
</tr>
<tr>
<td>03</td>
<td>933.075 MHz</td>
<td>888.075 MHz</td>
</tr>
<tr>
<td>...</td>
<td>...</td>
<td>...</td>
</tr>
<tr>
<td>77</td>
<td>934.925 MHz</td>
<td>889.925 MHz</td>
</tr>
<tr>
<td>78</td>
<td>934.950 MHz</td>
<td>889.950 MHz</td>
</tr>
<tr>
<td>79</td>
<td>934.975 MHz</td>
<td>889.975 MHz</td>
</tr>
</tbody>
</table>

Channels 01 and 79 are used as control channels for selective signalling (SSC) and cannot be used to transmit or receive voice or data. The traffic channels are 02 to 78 inclusive: the usual type of operation illustrated in Fig. 2. Two units, A and B, operate to each other through a repeater or master unit. This is dual-frequency operation with repeaters and master units transmitting in the High Band and receiving in the Low Band. Units (mobiles) transmit in the Low Band and receive in the High Band. If A and B want to transmit directly to each other, single frequency operation would be used. In this method, the units transmit and receive on the same channel in the same band: High Band.

In operation, a unit will scan and identify a free traffic channel which is subsequently used for voice or data. The number of the
The two control channels, 01 and 79, will be excluded. This is a simplified description and the reader should consult the official standard for a proper explanation. For operation through repeaters or master units, the procedure must be varied. However, the entire operation is a major change from the present method for PMR. The digital signal processing as in the SSC and ACK control signals are also transmitted and received in this mode.

Consider an example where user A wishes to contact user B in single-frequency operation. The main steps can be summarized by the following procedure, in which it has been assumed that A has just been switched on and is in the standby mode.

i) A enters call set-up mode. The traffic channels are scanned to find a free channel. If no channel is available, the unit will return to standby mode.

ii) A transmits an SSC on the control channel to B. This code includes identifiers for A and B, number of proposed traffic channels, and so on. The format of the code words is discussed in Section 5. A switches to receive mode on the traffic channel.

iii) If a traffic channel is identified as busy, A switches to the traffic channel specified in the SSC from A. On receipt of the ACK from B, A switches to the traffic channel. If no ACK or another SSC different from ii) is received by A, a set of retry procedures is implemented.

iv) A transmits the call set-up SSC as in ii) on the traffic channel to B. B, sends an ACK and both units enter communication mode. If no voice or data is transmitted after 10 seconds, or if the traffic channel has become busy between ii) and iv), or the SSC or ACK is not received, both units revert to standby mode.

v) In communication mode, all transmissions are preceded by a full SSC. There is a limit of three minutes to the time in communication mode in order to avoid congestion. In addition, if no valid voice or data is received after a certain period (5 seconds, or 10 if the unit has just entered the mode), both units return to the standby mode.

This is a simplified description and the reader should consult the official standard for a proper explanation. For operation through repeaters or master units, the procedure must be varied. However, the entire operation is a major change from the present method for PMR. The digital signal processing as in the SSC and ACK control signals is central to the system.

### 5. Selective signalling code

The selective signalling code (SSC) is a block of 568 bits which is fundamental to DSRR operation in all mode. It is sub-divided into the following blocks:

a) **Preamble of 256 bits** of 1010...etc. (bit reversal) for bit synchronization of the decoder in the receiver.

b) **Frame synchronization of 16 bits** to establish code word framing in the decoder of the receiver.

c) **Code word of 88 bits**.

b) and c) are transmitted three times as in Fig. 3.

The code word in c) is sub-divided further as follows.

<table>
<thead>
<tr>
<th>SSC number</th>
<th>Traffic channel code</th>
<th>First call code</th>
<th>Command code</th>
<th>Reserved</th>
<th>Code word counter</th>
<th>Manufacturer's code</th>
<th>Second call code</th>
<th>Cyclic redundancy check</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>7</td>
<td>24</td>
<td>4</td>
<td>2</td>
<td>2</td>
<td>8</td>
<td>24</td>
<td>16</td>
</tr>
</tbody>
</table>

The 16 check bits are generated from the 72 other bits by a (88, 72) cyclic code which has as its generator polynomial

\[ x^{16} + x^{11} + x^5 + x^4 + x^3 + x^2 + x + 1 \]

This should be a factor of \( x^{88} + 1 \) as per the mathematical conditions for cyclic codes. Refer to Appendix 1 for the various factors and other information. The code is supposed to have a Minimum Distance of 6 which will permit detection of up to five errors per word. This is the minimum number if differences in any two code words of 88 bits and is explained further in Appendix 2. The encoding operation can be summarized in the following steps:

1. Code word \( (d_{72} d_{71} d_{70} \ldots d_3 d_2 d_1) \) before the check bits are generated.
2. Code word

\[
\begin{array}{c|c}
88 & 87 & 86 & 19 & 18 & 17 & 16 & 15 & 14 & 3 & 2 & 1 \\
\hline
d_{72} d_{71} & d_{70} & \ldots & d_3 & d_2 & d_1 & \ast & \ast & \ast & \ast & \ast & \ast \\
\end{array}
\]

Data bits | Check bits

3. Code word can be written as a polynomial over \( GF(2) \) as follows:

\[ d_{72} x^{72} + d_{71} x^{71} + \ldots + d_3 x^3 + d_2 x^2 + d_1 x + 1 \]

with the check bits taken to be zero in each of the terms from \( x^{15} \) to \( x^0 \).

4. Polynomial in 3) is divided by the generator polynomial and the remainder, which is a polynomial of degree 15 with terms from \( x^{15} \) down to \( x^0 \), is added back to the original polynomial in 3) to produce the new revised polynomial of the final code word. The * in each location from 1 to 16 has now been replaced by a '1' or '0' as appropriate.

5. The final check bit corresponding to the coefficient of \( x^0 \) is inverted in order to give protection against misframing in the decoding operation.

The format for the ACK signal is exactly the same as for the SSC, but there are some variations. For example, the First Call Code for the SSC becomes the Second Call Code for the corresponding ACK signal.

### 6. Speech codec

Each speech frame of 20 ms is encoded into 76 speech parameters and this in turn is processed as a block of 260 bits. It should be noted that these bits are not of equal value. However, there is no need to examine this point. The codec is similar to that for the GSM.
The 34 significant bits, known as Class 1 bits, from the 260 bit block are expanded to 39 by the addition of five parity bits. A cyclic code with generator polynomial \((x^5 + x^2 + 1)\) produces the additional bits. The procedure is exactly the same as in Section 5 with the exception that the block size is 39. The generator polynomial should be a factor of \((x^{39} + 1)\) as per the mathematical requirements. Refer to Appendix 1 for the various factors and additional information.

The check bits represent the lower powers of the new polynomial for \(z^4\) down to \(z^0\).

### 7. Summary

The overall situation can be summarized in the following points:

1. **DSRR** is more efficient in the use of spectrum since High Band and Low Band require a total of 4 MHz.
2. **DSRR** provides a higher level of confidentiality and protection for the average user. In business radio, each user is licensed for a specific channel that must be shared with other users. A channel could be unintentionally or deliberately blocked by an unmodulated carrier. The Co-channel Rejection Test in MPT 1326 (which was not in the previous specification MPT 1301) does reduce, but not eliminate, this problem.
3. **DSRR** has scope for further development and expansion. In business radio, a reduction in the channel spacing to 6/4 kHz would provide more channels, but would not be considered an improvement.
4. **DSRR** has no requirement for an encryption facility as in the GSM. The method of encryption for the GSM has not been made public, but is believed to be a stream cryptosystem. A pseudo-random binary sequence is generated by an arrangement of shift registers and applied to the sequence of data bits in an addition modulo 2 operation. For the DSRR, there is no reason that an encryption device, such as a scrambler, cannot be added, but this is not required to meet the standard.
5. The encoding operation for data other than voice transmission is left to the manufacturer. The decoding operation for control signals, voice, and data is also left to the manufacturers.

### Appendix 1

The following factors were obtained using MATHEMATICA by Stephen Wolfram.

\[ (x^{38} + 1) = (x + 1)^8 (x^{10} + x^9 + x^8 + x^7 + x^6 + x^5 + x^4 + x^2 + x + 1)^8. \]

\[ (x^{16} + x^{14} + x^{12} + x^{11} + x^9 + x^8 + x^7 + x^6 + x^4 + x^3 + x^2 + x + 1)^8 + (x + 1)^2 (x^2 + x^6 + x^5 + x^4 + x^3 + x^2 + x + 1). \]

\[ (x^{39} + 1) = (x + 1) (x^2 + x + 1) (x^{12} + x^{10} + x^9 + x^8 + x^7 + x^5 + x^2 + x + 1) (x^{12} + x^{10} + x^9 + x^8 + x^7 + x^6 + x^5 + x^4 + x^2 + x + 1) (x^{12} + x^{11} + x^{10} + x^9 + x^8 + x^7 + x^6 + x^5 + x^3 + x^2 + x + 1). \]

\(x^5 + a^2 + 1\) has no real factors.

The interim standard does not include an analysis of the generator polynomials and, consequently, the choice of the factors in each of the two cases is not known.

### Appendix 2

Consider a simple 3-bit word that has an additional bit for even parity. Any two code words differ in at least one position in order to have distinct code word. When the parity bit on the right in the table is included, the minimum variation becomes 3. To correct a single error per word, the minimum variation would have to be 3. This is known as the Minimum Distance. A typical example is the (7, 4) Hamming Code, which has four data and three check bits. A code of Minimum Distance 3 would be able to detect up to two errors per word or be capable of correcting up to one error per word.
Appendix 3.
Permutation 1

<table>
<thead>
<tr>
<th>Position</th>
<th>New position</th>
<th>Notation in DSRR Standard</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>18</td>
<td>p(0)</td>
</tr>
<tr>
<td>2</td>
<td>19</td>
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<td>3</td>
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<td>p(2)</td>
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<tr>
<td>4</td>
<td>21</td>
<td>p(3)</td>
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<td>5</td>
<td>22</td>
<td>p(4)</td>
</tr>
<tr>
<td>6</td>
<td>1</td>
<td>d(0)</td>
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<tr>
<td>7</td>
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<td>43</td>
<td>d(37)</td>
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Appendix 4.
Permutation 2

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<th>Position</th>
<th>k</th>
<th>16(k mod 20)</th>
<th>Integer (k/20)</th>
<th>New position</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>1</td>
</tr>
<tr>
<td>2</td>
<td>13</td>
<td>4</td>
<td>1</td>
<td>17</td>
</tr>
<tr>
<td>3</td>
<td>2</td>
<td>6</td>
<td>2</td>
<td>33</td>
</tr>
<tr>
<td>4</td>
<td>3</td>
<td>8</td>
<td>1</td>
<td>33</td>
</tr>
<tr>
<td>5</td>
<td>19</td>
<td>32</td>
<td>0</td>
<td>305</td>
</tr>
<tr>
<td>6</td>
<td>20</td>
<td>1</td>
<td>1</td>
<td>2</td>
</tr>
<tr>
<td>7</td>
<td>21</td>
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<td>1</td>
<td>18</td>
</tr>
<tr>
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<td>304</td>
<td>0</td>
<td>306</td>
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<tr>
<td>9</td>
<td>39</td>
<td>0</td>
<td>1</td>
<td>3</td>
</tr>
<tr>
<td>10</td>
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<td>1</td>
<td>19</td>
</tr>
<tr>
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<td>41</td>
<td>2</td>
<td>2</td>
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<tr>
<td>12</td>
<td>42</td>
<td>319</td>
<td>13</td>
<td>304</td>
</tr>
</tbody>
</table>

Appendix 5.
Radio frequency characteristics

**Transmitter**
- Frequency error: ±2.5 kHz (maximum)
- Carrier power: 4 watts (maximum)
- Adjacent channel power: ≥70 dB below carrier power or 0.2 µW (maximum)
- Spurious emissions: 0.25 µW (maximum)
- Intermodulation attenuation: 40 dB for any component

**Receiver**
- Sensitivity: 6 dB µV e.m.f. (maximum) for a bit error rate of 10⁻²
- Co-channel rejection: -18 dB (minimum)
- Adjacent channel selectivity: 50 dB (minimum)
- Intermodulation response: 55 dB (minimum)
- Spurious response rejection: 60 dB (minimum)
- Blocking: 84 dB (minimum)
- Spurious radiations: 2 nW (maximum)

For the actual methods of measurement, the reader should consult the standard.
The greater part of this month's instalment is devoted to the operation of the microprocessor-controlled synthesizer used to tune the RF board, and to program and memorize station presets.

When it comes to designing a tuning system for a high-quality FM receiver, we are faced with the choice between two evils: synthesizer tuning or variable capacitor tuning. The first is forever and a day tied up with the problem of digital noise, the second with the problem of component availability, complex adjustment and repeatability. Bear in mind that a single tuning capacitor will not do here; what we require is a type with, say, six synchronized sections. Moreover, implementing a station preset facility on a tuner with a multi-section synchronized tuning capacitor is something that (we fear) is best left to instrument engineers. Also, most of you will not like the noise such a preset produces when a station is selected: it rather throw up memories of radios and TV sets used in the sixties and early seventies.

So, a synthesizer it will be. Although this does require special integrated circuits, these are by no means as costly and difficult to obtain as a six-way synchronized tuning capacitor. Since the tuner module used on the RF board is tuned by variable capacitance diodes (varicaps), there are no capacitors or inductors to adjust. What's more, the synthesizer concept proposed here can work without any adjustment whatsoever. As to tuning noise and phase jitter, this will not pose limitations in modern synthesizer concepts, by virtue of the high working frequency of today's synthesizer ICs, and the use of high-performance phase detectors.

A possibly more serious problem, particularly in home-made receivers, is formed by the noise generated by the digital components in a synthesizer. Given the fact that the digital noise level is determined to a large extent by the number and length of printed-circuit board tracks, the advantages of a microcontroller (with on-chip ROM, RAM and interfaces) over a system with external (E)PROM and peripheral ICs are fairly obvious. The only problem with mask-programmed microcontrollers is that they are produced in large quantities only. Hence, an alternative is used here in the form of an 80C32 microcontroller combined with an external EPROM that contains the control software. This has two advantages: first, the 80C32 is a low-power, inexpensive, and easily programmable device (witness our assembler course). Second, the external EPROM allows you to make changes to the control software should you so desire.

This leaves us with the problem of digital noise generated by the synthesizer. Unfortunately, such noise is hard to suppress completely, even when all the rules of RF screening are strictly observed. The solution to the problem is fairly drastic: switch off the source of the interference, i.e., the microcontroller, when it is not needed. This can be done with impunity because the microcontroller is active for very short periods as it changes or stores frequencies. During normal reception, the microcontroller is switched to the 'sleep' state, which also disables its clock oscillator.

Functions

Although a single-chip solution is not feasible because of the above aspects, the circuit of the synthesizer (Fig. 11) is fairly simple. The main components of the synthesizer are a control loop amplifier (opamp IC1, a type TL082), a prescaler (IC2, a Type SP893), and a synthesizer proper (IC4, a Type NJ6821). All other ICs in the circuit form part

ELEKTOR ELECTRONICS MAY 1992
of the microcontroller system, and serve to control the synthesizer, which offers a microcontroller interface via its pins 9 to 17. These pins convey the tuning frequency from the microcontroller to the synthesizer. This information is presented in binary form as eight datawords of four bits each. The function of the microcontroller is to gather all the data required to build these datawords, and also to store and display them.

One source for the tuning data gathered by the microcontroller is a 15-key keypad, which will be described in next month’s installment. This keypad is located on the single-sided ‘controls’ PCB, and is constantly scanned for activity. It has 10 numeric keys for direct entry of a frequency or a station preset number, as well as ENTER, STORE and EXECUTE keys to enter, store and call up station frequencies. The other two keys, UP and DOWN, allow the receiver to be tuned up or down in 50-kHz steps.

The station frequency and preset number are indicated on a large, bright, 7-digit, LED display that is also accommodated on the 'controls' board.

A single bisectional 8-bit port is used to scan the keys and drive the displays. This function requires an output current of 5 mA to be supplied at the logic ‘high’ as well as the logic 'low' level. Figure 11 shows that this port is formed by a 74HC245 (IC405). The address decoder of the display is actuated by the IOW signal. The display driving function has priority over the keyboard scanning function. Hence, if a key is pressed while the display is being updated, the 1-kΩ resistors in the keyboard matrix function as pull-up resistors only.

**Enter the 80C32 SBC**

The microcontroller system used here is basically a stripped-down version of the 80C32 single-board computer (Ref. 1) used for writing and testing the control software for the present FM tuner. The 80C32 is the ROM-less, CMOS, version of the 8052. One reason for using the 80C32 here is that it can be switched to a 'sleep' mode, which is necessary to prevent digital noise in the RF sections of the receiver. The external EPROM, IC403, is a 32-KByte type divided into two 16-KByte memory areas starting at 0000H and 8000H.

The static CMOS RAM, IC401, is a Type 6264LP-2. When the receiver is switched off, the RAM is powered by a lithium battery, which ensures that stored frequencies remain intact for at least 10 years. The software allows you to enter up to 99 presets, which is more than the maximum number of stations that can be contained by the entire FM band, even if a 250-kHz raster were used.

To make sure that the RAM is timely disabled when the supply voltage drops, RAM input CS2 (pin 26) is held at a slightly lower voltage with the aid of resistors R51 and R402. This prevents the microcontroller writing random data into the RAM when the receiver is switched off.

IC406 decodes the address ranges for the RAM, and address 0E00H for the I/O port, IC405. The NOR gates contained in IC407 form the address decoder for the EPROM. Circuit IC403 demultiplexes the lower 8 address bits and the data signals.

The inverters contained in IC408 allow a serial interface (RS232) to be connected, if this can work with a voltage swing of 5 V. The interface may be modified as follows when it is to be connected to a device that works with ±12 V swings: fit a 3.3-kΩ resistor in series with the Rx line. Together with the 2.7-kΩ resistor in array R420, this forms a
voltage divider that changes +12 V to +5 V. To convert the negative (-12 V) level at the Rx input into (approximately) 0 V, connect a diode Type 1N4148 between the Rx line and ground. The cathode goes to Rx, the anode to ground. The two interrupt inputs, INTO and LNT1, respond to logic 'high' signals, provided, of course, that interrupts are enabled (in software).

The processor is reset by applying a logic 'low' level to the RESET input of the synthesizer board. R-C network R418 -C413 supplies the reset pulse at power-on.

The control software for the synthesizer is all machine code, and fast enough to scan the keyboard as a part of the normal program. Hence, an interrupt request is not required when a key is pressed.

The synthesizer

The heart of the synthesizer is formed by the NJ8821, a CMOS IC from Plessey. This IC is marked by a high-performance, high-frequency, phase detector, and low current consumption (3.5 mA typ.). In a PLL circuit, this IC allows excellent phase stability and low noise to be achieved.

Each PLL-based synthesizer consists of four basic components, whose function and materialization are discussed below.

Voltage-controlled oscillator (VCO)

This is contained in the FD12 tuner module on the RF board. Its output frequency is supplied to the prescaler in the synthesizer via the UOSC socket on the module.

Reference oscillator

As shown in Fig. 2, this is contained in the NJ8821. The oscillator works with an external 2-MHz quartz crystal, which is connected to the OSC IN and OSC OUT pins of the IC. The crystal operates in parallel-resonance mode, and is 'flanked' by the usual parallel capacitors to ground.

Reference counter and prescaler

The reference frequency, fo, is derived from the reference oscillator frequency by a programmable 11-bit counter (divisor), whose divisor can be set between 6 and 4,094 in steps of 2. A binary -2 scaler is connected to the output of the reference counter to achieve an output signal with a 50% mark/space ratio. This means that the total divisor is two times the programmed divisor.

The oscillator signal, which is to be compared to the reference signal, is supplied by the VCO in the FD12 tuner, and led to the synthesizer input, FIN (pin 4), via a Plessey Type SP8795 prescaler. The prescaler output is divided by two programmable dividers in the NJ8821: a 7-bit divider, 'A', and a 10-bit divider, 'M'. Both are connected to the prescaler via the 'MC' signal at pin 18. The 'MC' signal is low at the start of a counter cycle, and remains low until counter 'A' has completed one cycle. Next, 'MC' goes high, and remains high until counter 'M' has completed its cycle. Next, both counters are reset. The resulting divide ratio is

\[ M \times N + A \]

where \( N \) and \( N + 1 \) are the divisors of the prescaler controlled by the 'MC' signal.

The dividers that can be programmed are 0 to 127 for counter 'A', and 3 to 1,023 for counter 'M'. For this so-called modulo-2 counting principle to operate correctly, divider 'M' must always be greater than divider 'A'.

Port 1 of the 80C32 is used to program the
Fig. 16. Track layout (mirror image) and component overlay of the PCB for the power supply unit.

ELEKTOR ELECTRONICS MAY 1992
such that the divider output frequency, \( f_d \), equals the reference frequency, \( f_r/2 \) (behind the binary scaler), given the desired oscillator frequency after prescaling at pin FIN.

**Phase comparator**

The \( f_s/2 \) and \( f_r \) signals are compared in a phase comparator, which is actually a two-stage phase detector. The first stage is a digital phase/frequency detector, which enables the PLL to lock fast by supplying a 'cause' error signal that can signal one of three conditions: output PDB supplies positive-going pulses when the tunner oscillator frequency is too high (\( f_s > f_r/2 \)); negative-going pulses when the tunner oscillator frequency is too low (\( f_s < f_r/2 \)); or no signal (PDB switched to high impedance) when the two frequencies are equal or sufficiently close. In the latter case, the MOSFET at the LOCK output (pin 3) starts to conduct. Consequently, transistor Ti conducts also, and causes a LED to light, indicating that the PLL is locked. The inverted 'lock' signal may also be used to mute the receiver when the PLL is out of lock.

As soon as the PDB output of the digital phase comparator is at high impedance, an analogue sample-and-hold starts to work on the fine tuning. Its output signal, PDA (pin 1) is at about half the supply voltage when the PLL is locked. Starting at that level, the voltage rises when the phase of \( f_s/2 \) lags that of \( f_r \) and drops when \( f_s \) lags \( f_r/2 \). The linear range of the output voltage is determined by an external resistor, \( R_{417} \), connected to the RB terminal (pin 19). The RB output controls the amplification of the sample-and-hold comparator. The value of \( R_{417} \) is determined by the reference frequency used. The 'hold' capacitor is connected between the CH terminal (pin 20) and ground.

The output voltages of the two phase comparators are added by two resistors at the inverting input of the control loop amplifier, opamp IC12. The opamp is wired in a so-called pi-configuration, and its non-inverting input is held at a well-decoupled level of \( \pm 2.5 \) V, i.e., half the supply voltage. The speed of the control loop and the suppression of high-frequency components in the comparator output signals are determined by the time constant formed by the feedback circuit, and a passive low-pass filter, \( R_{417} - C_{412} \), at the output of the opamp. The opamp output signal is the synthesizer-generated tuning voltage for the FD12 tuner module on the RF board.

**Prescaler**

It will be recalled that the FD12 tuner module is basically a superheterodyne receiver, in which the local oscillator (= VCO) frequency is 10.7 MHz higher than the receive frequency. This means that the VCO frequency range must be 98.7 MHz to 118.7 MHz to cover the entire FM band (88 MHz to 108 MHz). However, the synthesizer IC can handle frequencies up to about 15 MHz only, whence the need for a prescaler that reduces the VCO frequency to under 10 MHz.

Figures 14 and 15 show the block diagram and the pinning respectively of the SP8795 prescaler, whose input frequency range extends from 20 MHz to about 225 MHz. This prescaler is marked by high sensitivity (200 mVpp) and low current consumption (approx. 5 mA). An on-board voltage regulator enables the IC to be used with supply voltages between 6.5 V and 9.5 V. Alternatively, 5-V operation is possible by not using the on-board regulator. This option is used in the present circuit: pins 7 and 8 are connected direct to the supply voltage (pin 2).

A special feature of the SP8795 when used in combination with the NJ8821 is the modulo-2 divider, which can switch between dividers N and \( N + 1 \), where \( N \) equals 32, and is selected by a logic 'high' level at pin 1. When a 'low' level is applied, the IC divides by 33. Here, this means that the frequency range of the synthesizer input signal is about 3 MHz to 3.7 MHz.

**Power supply board**

The power supply circuit discussed last month (Fig. 10) is constructed on the printed circuit board shown in Fig. 16. The fuses on this board are inserted into the mains lines, which requires plastic caps to be fitted on the holders. A double-pole mains on/off switch is connected between the mains inputs of the PSU board and the appliance socket on the rear panel of the tuner case.

To prevent confusion, note again that the +32-V output voltage at connector K3 of the PSU unit powers the entire RF board. This means that it is not required to connect separate wires from the PSU to the 'A' and 'B' ST (tuning voltage) pins of the RF board. Instead, interconnect these two pins at the RF board, and run a single wire from K3 on the PSU board to the 'A' pin on the RF board. The 5-V output of the PSU board is used to power the synthesizer circuit. □

Next month we will tackle the construction of the synthesizer, and discuss the operation of the 'controls' board.
'Grab yer gal and hit the floor' is sure to take on a completely new meaning before long. Now while gals in the more traditional sense of the word are often pretty difficult to control, let alone to be forced into 'tailor-made' behaviour (which adds considerably to their charm), the electronic versions we are dealing with here (identified by three capital letters, GAL) are admittedly less exciting, but much more easy-going. The GAL programmer described in this article offers everything needed to burn complex logic functions into today's most popular GALs. The software used to control the programmer is menu-driven, and can be run on all IBM PCs and compatibles.

Design by M. Nosswitz

Following last month's introductory article on features and functions of GALs (general array logic) we now take a more practical look at things with the description of a powerful, low-cost, GAL programmer for use with PCs.

The advantages of GALs over discrete logic circuits are significant. At reasonable cost, you obtain a piece of programmable logic that can be erased, too! Apart from their remarkable flexibility, GALs offer the possibility to 'stamp' them electronically with an identification code, as well as to protect them from being read out (and copied). Further, GALs are pretty fast, A-versions achieving propagation delays of the order of 10 ns only.

The operation of the programmer described here is strictly controlled via the Centronics port of an MS-DOS compatible computer running the software developed for the programmer. The control software was developed with the aid of Turbo Pascal 6.0, and is capable of programming GAL Types 16V6, 20V8, 16V8A, and 20V8A. The control software is available ready-programmed, and comes on a diskette supplied through the Readers Services.

The programmer and the computer communicate via the Centronics port, using a serial format to exchange data and commands. Remarkably, only five lines are required to handle all functions. At the programmer side, a shift register is used to convert the serial data into parallel. Despite this converter, the total circuit of the programmer is not too complex.

The hardware

Just like almost any other electrically programmable component, a GAL needs a programming voltage that is higher than the normal supply voltage. As shown in the circuit diagram (Fig. 1), this has been taken into account in the design of the power supply of the programmer, which caters for the normal board supply voltage of 5 V as well as for an auxiliary voltage of 16.5 V. Preset Pt serves to adjust the latter voltage accurately. Fortu
nately, the current consumption of the programmer is low, so that a small (4-VA) mains transformer may be used, while the two regulators can make do without heat-sinks. The control signals needed for shift registers IC4 and IC5 are supplied via three of the eight data lines on the Centronics interface. Dataline D0 carries the serial data, while D1 and D3 supply the shift register clock and strobe signal respectively. Pull-up resistors R2, R3 and R4 ensure correct signal levels during the data exchange. The two shift registers are connected in series via the serial output QS (pin 9 of IC4) and the data input (pin 2 of IC5). The three-state outputs of IC4 and IC5 are always active because the OE (output enable) inputs are tied to +5 V. The 16 databits at the parallel outputs Q0-Q8 of IC4 and IC5 determine the operation of the rest of the circuit.

One half of a dual 2-of-4 decoder, IC36, switches the supply voltage via transistors T5 and T7. The other half, IC3a, controls the presence of the programming voltage at the respective pins of the GAL socket, via transistor pairs T2-T4 and T3-T5. This is possible only if the decoder outputs have been enabled beforehand by a low level at the ENABLE (E) inputs.

Resistors R20-R25 reduce the short-circuit currents at the register and GAL outputs to safe values when these are switched to the read mode. R1, R9 and R29 are the pull-down resistors needed for the programming mode. Diode D2 protects output Q8 of IC4 against the programming voltage.

The computer can read the GAL data matrix via Centronics handshaking line ACK. This requires a selection operation via the select (SLCT) signal. By virtue of SLCT, the software is capable of checking if the programmer hardware is connected, and if a GAL is fitted. If desired, this function may be switched off by modifying the file GAL.CFG (Fig. 6). If hardware checking is not required, the SLCT line may be omitted.

The GAL is fitted into a zero-insertion force (ZIF) socket. Finally, LED D9 lights...
Fig. 2. Track layout (mirror image) and component mounting plan of the single-sided PCB for the GAL programmer.
when the programmer is active, and Di when the GAL receives its supply voltage. It is recommended to insert the GAL only when Di is out.

Software development for GALs

To begin with, use any ASCII-compatible word processor to produce an equations file that describes the desired function of the GAL. 'GALDEMO.EQN' (Fig. 3) contained on the disk supplied for this project is such a file, and may serve as an example. Basically, variables are assigned to the inputs and outputs, and the logic function is described by a Boolean equation. If desired, an 8-bit identification code ('signature') can be burned into the chip.

Next, run a check on the program syntax. This requires an auxiliary program such as 'EQN2JED' included in the Opal Junior™ GAL programming software package from National Semiconductor (this package is supplied free of charge with your GAL programmer software, how's that?). When no errors are detected, EQN2JED generates the

The software

After starting the control program, GAL651AE.EXE, the screen shows the start and end indicators. Further, the screen graphics indicate three blocks: the JEDEC file, the matrix memory and the GAL's hardware environment. The command names are shown in between the blocks, and can be selected by typing the highlighted letter. Command abbreviations may also be used. An error 'beep' sounds when you enter a non-existing command.

The GAL TYPE command allows you to select the device type to be handled. This selection must be completed before the GAL is inserted into the ZIF socket on the programmer board. While executing the GAL commands, the software automatically checks if the right GAL type is being used.

The READ command is used to transfer the JEDEC file into the matrix memory. After entering 'Y' and a return, the listing is displayed in the JEDEC file in the selected subdirectory. Alternatively, you may enter the full path and file name. After requesting a file list (for instance, A:*), the screen shows all JEDEC files found. The desired file is selected by moving to it using the PageUp and PageDown keys and the arrow keys. The return key activates the selected command, which then operates on the selected file.

The use of the WRITE command is similar to that of the READ command described above. An 'overwrite?' alert is shown if you save a file under a name that is already in use in the selected (sub-)directory. All file names are automatically saved with the 'JED' extension appended.

Selecting PROGRAM from the menu

documentation file GALDEMO1.DOC (Fig. 5), and the associated JEDEC file GALDEMO1.JED (Fig. 4). The JEDEC file produced with the aid of EQN2JED contains all information on the cells contained in the GAL, and serves to actually program the device.

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Selecting PROGRAM from the menu

...
Fig. 4. JEDEC output file produced by the EQN2JED utility from National Semiconductor.

Fig. 5. Example of a documentation file produced by EQN2JED.

The COPY PROTECT command may be used to actuate the copy protection ('security bit') in the GAL. When the security bit is set, it is impossible to read anything from the GAL except the identification code and the GAL configuration.

The BULK ERASE command may also be used on its own, i.e., as part of a programming sequence, to clear the contents of a GAL. This obviates the need for an erase operation before programming.

OUTPUT TO LPT1: directs the matrix contents to a printer connected to the LPT1 output of the PC. Needless to say that hard copy of the matrix contents may be very useful for documentation purposes.

The SHOW/EDIT command allows you to examine, on the screen, the content of the selected location. If a specific cell is selected, the cursor starts to flash at this location. At the same time, the identification code, if given in the source code, is shown in hexadecimal as well as ASCII notation.

The F10 key takes you to the edit mode, which makes it possible to change the cell contents with the aid of the cursor keys. The 'Fill' function may be used to fill a block in the GAL with ones or zeroes.

Selecting the Program, Verify or Load commands causes the number of programmed cells, the GAL manufacturer, and the checksum to be shown in the GAL symbol on the screen.

The function of the End command will be obvious.

Finally, any command entered may be terminated with the aid of the ESC (escape) key.

GAL.CFG

(user number)

$378   Address of Centronics port for GAL data exchange
2000   Error display time in ms
2000   Error beep frequency in Hz
2000   Beep length in ms
1   Check if GAL hardware accessible
0   Switch for general RESSEP (EPROM, PROM, GAL)
yes=1
no=0
1   Basic GAL type selection
16V8=1, 20V8=2, 16V8A=3, 20V8A=4
1   Background colour
5   Foreground colour
5   Background colour
5   Foreground colour
5   Background colour
5   Foreground colour
9   Background colour
8   Background colour
8   Background colour
8   Line distance to upper paper edge (printing)
8   Line distance between page 1 and page 2 (printing)
8   Empty lines after page 2 (printing)
8   Character distance from left-hand paper edge (printing)

Fig. 6. The control program for the GAL programmer reads 25 parameters from a configuration file called GAL.CFG. This file can be produced or edited with any simple word processor, such as EDLIN or the one in PCTools. All 25 parameters must be present in the order shown here. The meaning of the values is apparent from the comment in each line.
NEW LOGIC SYMBOLS
by our technical staff

In the late 1960s, the International Electrotechnical Committee, IEC, set up a working group to devise symbols for binary logic. The work of this group culminated in the early 1980s in IEC Publication 617: Graphical Symbols for Diagrams. Part 12: Binary Logic Elements. This standard, published in 1983, went into general circulation in early 1984. It takes into account the use of computer-aided drafting equipment and all symbols are designed on a grid.

The symbology contained in the standard provides that each symbol has one meaning only; rules are given how these symbols can be united to form the most complex logic functions. In a way, it may be compared to the higher programming languages: these, too, contain symbols (letters, ciphers, punctuation marks) that have one meaning: syntax is used to unite them into complete computer programs.

In spite of the standard having been published in 1983, even today many manufacturers feel obliged to give the 'old' symbol alongside the new IEC symbol. The reason for this is that, although the new symbols are far more informative than the previous ones, they have to be learned like a new language. And it remains true that most of us are conservative: we don't like change.

Nevertheless, we feel that the time has come to start using the new symbols in our drawings. Most technical schools and colleges, as well as the semiconductor industry, have been using them for years. The reason that we have been slower than usual in adopting a newer and better technique is a very practical one. Our design department has been investing in the acquisition and further development of computers and software that are able to provide the desired quality of graphics output—note that a 'normal' CAD system is not suitable. However, these systems have recently come 'on stream' and the department can now start using electronic means of reproducing the new symbols.

Do not think, though, that we are changing overnight: the move to using the new symbols will be a gradual one and will probably take until the end of the year. To prepare you for the change, a number of symbols for the basic functions, as well as a few more complex ones, are shown in the illustrations. We will publish the illustrations again in a number of future editions.

An important point to bear in mind is that we are not adopting the new IEC symbols automatically in all cases: several are far larger than the previous symbols and their use might mean that the relevant circuit diagrams would become rather unwieldy for reproduction in the magazine. In such cases, we may choose a simplified symbol that is more akin to the old one. This will, of course, be made absolutely clear in the text, on the diagram or in the caption to the diagram.

Do not be put off by a first sight of the new symbols: they may seem complicated at first, but they offer excellent facilities for specifying designs without requiring precise forms of implementation. At the same time, they retain a precise specification of the required logic functions with the minimum amount of support documentation.

An excellent book for learning to understand the new symbols is A practical introduction to the new logic symbols by Ian Kampell. ISBN 0 408 01461 X, published by The Butterworth Group, Borough Green, Sevenoaks TN15 8PH, England.

Table 1 (opposite page). Qualifying symbols and symbols used inside the outline.

<table>
<thead>
<tr>
<th>TABLE 1</th>
<th>Dependency notation</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>The letter x in this table is used to denote an identifying number; substitution of an appropriate number is required in normal usage.</td>
</tr>
<tr>
<td>Ax</td>
<td>ADDRESS dependency</td>
</tr>
<tr>
<td>Cx</td>
<td>CONTROL dependency</td>
</tr>
<tr>
<td>Enx</td>
<td>ENABLE dependency</td>
</tr>
<tr>
<td>Gx</td>
<td>AND dependency</td>
</tr>
<tr>
<td>Mx</td>
<td>MODE dependency</td>
</tr>
<tr>
<td>Nx</td>
<td>NEGATE dependency</td>
</tr>
<tr>
<td>Rx</td>
<td>RESET dependency</td>
</tr>
<tr>
<td>Sx</td>
<td>SET dependency</td>
</tr>
<tr>
<td>Vx</td>
<td>OR dependency</td>
</tr>
<tr>
<td>Zx</td>
<td>INTERCONNECTION</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>TABLE 2</th>
<th>General qualifying symbols</th>
</tr>
</thead>
<tbody>
<tr>
<td>&amp;</td>
<td>AND gate or function</td>
</tr>
<tr>
<td>≥I</td>
<td>OR gate or function</td>
</tr>
<tr>
<td>=I</td>
<td>XOR gate or function</td>
</tr>
<tr>
<td>x</td>
<td>Logic identity</td>
</tr>
<tr>
<td>2k</td>
<td>The single input must be active (i.e., non-inverting buffer)</td>
</tr>
<tr>
<td>2k+1</td>
<td>An odd number of inputs must be active</td>
</tr>
<tr>
<td>X/Y</td>
<td>Coder or code converter (e.g., DEC/BCD, BIN/7-SEG)</td>
</tr>
<tr>
<td>MUX</td>
<td>Multiplexer</td>
</tr>
<tr>
<td>DMUX</td>
<td>Demultiplexer</td>
</tr>
<tr>
<td>Σ</td>
<td>Adder</td>
</tr>
<tr>
<td>P-Q</td>
<td>Subtractor</td>
</tr>
<tr>
<td>CPG</td>
<td>Look-ahead carry generator</td>
</tr>
<tr>
<td>π</td>
<td>Multiplier</td>
</tr>
<tr>
<td>COMP</td>
<td>Comparator</td>
</tr>
<tr>
<td>ALU</td>
<td>Arithmetic logic unit</td>
</tr>
<tr>
<td>Ln</td>
<td>Retriggerable nonstable</td>
</tr>
<tr>
<td>jn</td>
<td>Non-retriggerable nonstable</td>
</tr>
<tr>
<td>0</td>
<td>Astable, general symbol</td>
</tr>
<tr>
<td>1</td>
<td>Astable, synchronously</td>
</tr>
<tr>
<td>R</td>
<td>Asynchronous starting</td>
</tr>
<tr>
<td>R</td>
<td>Asynchronous starting, stopping after completion of last pulse</td>
</tr>
<tr>
<td>SRGm</td>
<td>Shift register where substitution for m specifies number of bits</td>
</tr>
<tr>
<td>CTRm</td>
<td>Counter where substitution for m specifies number of bits</td>
</tr>
<tr>
<td>CTRDIVm</td>
<td>Counter/divider where substitution for m specifies cycle length</td>
</tr>
<tr>
<td>RCTRm</td>
<td>Asynchronous counter where substitution for m specifies cycle length</td>
</tr>
<tr>
<td>FIFO</td>
<td>First-in first-out memory</td>
</tr>
<tr>
<td>CT=m</td>
<td>If output: active if the counter state of the register content is m</td>
</tr>
<tr>
<td>I=0</td>
<td>Element is reset at power-up</td>
</tr>
<tr>
<td>I=1</td>
<td>Element is set at power-up</td>
</tr>
<tr>
<td>RAM</td>
<td>Random-access memory</td>
</tr>
<tr>
<td>DRAM</td>
<td>Dynamic random-access memory</td>
</tr>
</tbody>
</table>

Table 3 (opposite page). Qualifying symbols and symbols used inside the outline.
<table>
<thead>
<tr>
<th>Symbol</th>
<th>Description</th>
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<tr>
<td><img src="image1.png" alt="Symbol" /></td>
<td>General logic symbol</td>
</tr>
<tr>
<td><img src="image2.png" alt="Symbol" /></td>
<td>Logic with common control</td>
</tr>
<tr>
<td><img src="image3.png" alt="Symbol" /></td>
<td>Logic with common output</td>
</tr>
<tr>
<td><img src="image4.png" alt="Symbol" /></td>
<td>Polarity indicator at input</td>
</tr>
<tr>
<td><img src="image5.png" alt="Symbol" /></td>
<td>Polarity indicator at output</td>
</tr>
<tr>
<td><img src="image6.png" alt="Symbol" /></td>
<td>Dynamic input (edge triggered)</td>
</tr>
<tr>
<td><img src="image7.png" alt="Symbol" /></td>
<td>Dynamic input</td>
</tr>
<tr>
<td><img src="image8.png" alt="Symbol" /></td>
<td>Internal connection</td>
</tr>
<tr>
<td><img src="image9.png" alt="Symbol" /></td>
<td>Internal connection with negation</td>
</tr>
<tr>
<td><img src="image10.png" alt="Symbol" /></td>
<td>Internal connection with dynamic character</td>
</tr>
<tr>
<td><img src="image11.png" alt="Symbol" /></td>
<td>Internal connection with negation and dynamic character</td>
</tr>
<tr>
<td><img src="image12.png" alt="Symbol" /></td>
<td>Internal (virtual) input</td>
</tr>
<tr>
<td><img src="image13.png" alt="Symbol" /></td>
<td>Internal (virtual) output</td>
</tr>
<tr>
<td><img src="image14.png" alt="Symbol" /></td>
<td>Output of a pulse-controlled element</td>
</tr>
<tr>
<td><img src="image15.png" alt="Symbol" /></td>
<td>E threshold (input with hysteresis)</td>
</tr>
<tr>
<td><img src="image16.png" alt="Symbol" /></td>
<td>Open-circuit H-type output</td>
</tr>
<tr>
<td><img src="image17.png" alt="Symbol" /></td>
<td>Passive pull-down output (similar to 19 but containing internal pull-down resistor)</td>
</tr>
<tr>
<td><img src="image18.png" alt="Symbol" /></td>
<td>Passive pull-up output (similar to 20 but containing internal pull-up resistor)</td>
</tr>
<tr>
<td><img src="image19.png" alt="Symbol" /></td>
<td>3-state output</td>
</tr>
</tbody>
</table>

### Logic with common control
- ![Symbol](image20.png) J input
- ![Symbol](image21.png) K input

### Logic with common output
- ![Symbol](image22.png) R (reset) input
- ![Symbol](image23.png) T (set) input
- ![Symbol](image24.png) T (toggle) input
- ![Symbol](image25.png) Shift right (down)
- ![Symbol](image26.png) Shift left (up)
- ![Symbol](image27.png) Count up
- ![Symbol](image28.png) Count down
- ![Symbol](image29.png) Compare output of an associative memory
- ![Symbol](image30.png) Bit grouping symbol for inputs
- ![Symbol](image31.png) Bit grouping symbol for outputs
- ![Symbol](image32.png) Operand (P) input
- ![Symbol](image33.png) Operand (Q) input
- ![Symbol](image34.png) Greater-than input
- ![Symbol](image35.png) Less-than input
- ![Symbol](image36.png) Equal input
- ![Symbol](image37.png) Greater-than output of a comparator
- ![Symbol](image38.png) Less-than output of a comparator
- ![Symbol](image39.png) Equal output of a comparator

### Dynamic input
- ![Symbol](image40.png) Borrow-in input
- ![Symbol](image41.png) Borrow-generate input
- ![Symbol](image42.png) Borrow-propagate input
- ![Symbol](image43.png) Borrow-out output (e.g., ripple)
- ![Symbol](image44.png) Borrow-propagate output
- ![Symbol](image45.png) Carry-in input
- ![Symbol](image46.png) Carry-generate input
- ![Symbol](image47.png) Carry-propagate input
- ![Symbol](image48.png) Carry-propagate output
- ![Symbol](image49.png) Carry-out output
- ![Symbol](image50.png) Carry-propagate output

### Internal connection
- ![Symbol](image51.png) Content input
- ![Symbol](image52.png) Content output
- ![Symbol](image53.png) Line grouping symbol for inputs
- ![Symbol](image54.png) Line grouping symbol for outputs
- ![Symbol](image55.png) Non-logic input
- ![Symbol](image56.png) Non-logic output
- ![Symbol](image57.png) Input for analogue signals
- ![Symbol](image58.png) Analogue output
- ![Symbol](image59.png) Digital output
Fig. 1. A number of examples drawn according to IEC standard No. 617.
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</tr>
<tr>
<td>Australia &amp; New Zealand</td>
<td>£42.00</td>
</tr>
<tr>
<td>Far East &amp; South Pacific regions</td>
<td>£42.00</td>
</tr>
</tbody>
</table>

To order your subscription, please use the order form on page 67.
LETTERS

ELECTRONIC CONTROL OF ASTRONOMICAL INSTRUMENTS

Dear Editor—In response to the letter from Mr S.J. Dearden (Readers' letters, February), I can let him know that Mr Terry Brooks of Cambridge University Press is writing a book on CCD cameras with PCBs available.

G. Glenn, Cambridge

Dear Editor—We are writing in response to your enquirer in 'Readers Corner' in your February issue concerning an accurate telescope control system operated by a remotely located microcomputer. We have developed such a system from your 'Universal stepper motor controller' (January 1987). On a single PCB we have incorporated the capability to drive two stepper motors to control the two axes of the telescope. The board also incorporates a temperature controlled oscillator to enable the telescope to track the sky, limit switches and hall effect positioning sensors. The compact unit also houses the power supplies for the system. Software has been developed to run the system on an IBM PC. We have high-quality documentation generated by a CAD system plus complete technical information. A similar system has also been used in the department for an X-Y translation table for astronomical use.

John Cutts, Head of Electronic Workshop Department of Physics The University of Queensland Brisbane, Australia

We thank both Mr Glenn and Mr Cutts for their information and interest. The information and drawings supplied by Mr Cutts have already been forwarded to Mr Dearden.

Editor

FOUR-TERMINAL NETWORKS

Dear Editor—Enjoyed reading Mr Knight's article (October & November, 1991), but have some comments.

1. Para. 2 of 'Finding the attenuation' (p. 48, October) should have read as follows.

'A desired value of $R_0$ can be obtained with numerous combinations of $R_1$ and $R_2$, depending on the value of the attenuation. $N$. Looking at Fig. 7, for instance, both sections shown have a characteristic resistance $R_0 = 30 \Omega$ (check this on for yourself!), but the network on the right will provide a greater degree of attenuation than the one on the left. A desired value of $N$ can be obtained with numerous combinations of $R_1$ and $R_2$, depending on the $R_0$ chosen. If, for instance, in Fig. 8, $R_1 = 20 \Omega$, $R_2 = 12.5 \Omega$ and $N = 14 \text{ dB}$, $R_1 = R_2 = 30 \Omega$ and $R_1 = R_2$ becomes $120 \Omega$.

2. Many attenuator designs are tabulated in the literature for various values of $N$ for a specific value of $R_0$. For instance, if $N = 20 \text{ dB}$ and $R_0 = 500 \Omega$, $R_1 = 409.1 \Omega$ and $R_2 = 101.0 \Omega$ for the symmetrical T attenuator. For any new $R_0$, requiring $20 \text{ dB}$, $R_1 = R_2(R_0/500)$ and $R_2 = R_1(R_0/500)$.

3. There should have been a bibliography and/or references of related material listed for interested parties, since I surmise that your readers range from hobbyists to technicians to engineers.

Joseph A. Lutowski, USA

Steve Knight has commented:

'1. I think Mr Lutowski has simply rephrased the paragraph without affecting its content. The object was to show that different attenuations can be obtained for identical values of characteristic resistance $R_0$. The actual attenuations were not stated, but this was not necessary at this point: a basic fact was being stated.'

'2. I agree that such tabulated scales are to be found in a number of reference books and are useful to a designer. There was obviously no space in a short article to provide a bibliography and/or references of related material listed for interested parties, since I surmise that your readers range from hobbyists to technicians to engineers.'

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a list. The two expressions Mr Lutowski gives for any alternative \( R_o \) were not included for the same reason."

"3. It is always a problem providing bibliographies, particularly for short articles, and again there is the matter of space. A list of British references might not have been suitable for the many overseas readers, while a list of interest to Mr Lutowski might not have been so to non-American readers. For those reasons, I decided to omit any references from the article."

"I am pleased to read that Mr Lutowski enjoyed these two articles; they were presented in as simple a manner as possible in an effort to make clear a subject that is often a difficult one for hobbyists and technicians to grasp."

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LETTERS

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COMPONENTS

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BOOKS

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BUNDLES

Elektor Electronics Binder | £6.95 |

FRONT PANELS

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ELEKTOR ELECTRONICS MAY 1992
Omni-Pro II - The Next Generation

When you get a new product, what are your main concerns? Freedom from frustration is certainly one important consideration, for your time is valuable. You will want a product which is reliable and sophisticated, yet simple to use, with clearly written documentation. You will be looking for a high standard of technical support and regular upgrades for the product.

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Well, for instance, the interface is non-invasive to the computer's parallel port, which is speed-limited, and probably connected to your printer. A dedicated plug-in half card performs fast data transfers.

The software is a professional package in full colour that will run in only 400K of RAM. What's more it will run on any PC/AT or compatible - even the latest 486 machines. That's because Omni-Pro II has its own independent clock - some programmers rely on the computer's clock and won't work with faster machines.

Ground pins are connected by relays - not by logic outputs. Some vendors won't approve programmers which don't ground pins in this way.

The 40-pin Textool socket can be changed without even having to remove the cover. A complete range of PLCC adapters is available.

Truly Universal

Omni-Pro II has universal pin-drivers which will accommodate a very wide selection of parts. You can program BIPOLARS, PROMS, E/EEPROMS, PALS, GALS, EPLAS, PEELS, E/EPLDS and MICRO-CONTROLLERS. The latest FLASH EPROMS are supported too. The list has over 1250 devices already and substantial numbers of new devices will be added FREE every quarter.

We have optimised programming speeds, using algorithms like Quickpulse, Flashrite and TI Snap and have already gained parts approval from TI, NS and ICT. We provide fast downloading of files in any standard format: Intel Hex, Motorola, Tek Hex, HP/4000ABS or Binary. You can also send JEDEC files from all popular PLD compilers and JEDEC standard vector testing is supported; a full array of test condition codes can be generated.

Remember - you get a 30 day money-back guarantee, FREE quarterly software updates and FREE technical support - as much as you need. Phone now for a free Demo Disk and up-to-date Device-List.

Omni-Pro II comes with a FREE copy of NS's superb Open Programmable Architecture Language - OPAL Junior.

Omni-Pro II - complete ..... £495

Gang-of-eight Programmer ..... £395

This production programmer from Dataman can handle all 25 and 27 pin EPROMs up to 512K bits. Programs eight copies from a master EPROM, or from an object file. The G8 offers fast programming methods and three user-selectable programming voltages. G8 is clearly designed for the busy workshop being supplied, as standard, in a high quality steel case.

Software Development from £195

Dataman's Software Development Environment, SDE, comprises a two-window Editor, Macro Assembler, Linker, Librarian, Serial Comms and intelligent Make facility. The latter reassembles selectively just those files you have edited, links them and downloads the object-code to your Emulator or Programmer. It's quick and painless. If the assembler finds a mistake it puts you back in the Editor at the right place to fix it.

The Multi-Processor version supports all common micros - please ask for list. The Disassembler makes source files out of object code - from a ROM, for example. SDE is not copy-protected.

SDE Multi-processors & Disassemblers ..... £695
SDE Multi-processors (includes 82 micros) ..... £395
SDE Single-processor ..... £195

OPAL from NS ................... £297

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Simply hold the gun right over the EPROM's window and squeeze the trigger. The strobe eraser puts out intense UV light at the right wavelength (253.7nm). Erase EPROMS on the bench, in the PCB or in the programmer.

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