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### Digital Capacitance Meter

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<thead>
<tr>
<th>Model</th>
<th>Description</th>
<th>Price</th>
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<tbody>
<tr>
<td>CM-1550B</td>
<td>9 Ranges 1'-30.000ufd 5% basic acry, Zero control w/ Case, Big 1&quot; Display</td>
<td>$58.95</td>
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### Digital LCR Meter

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<tr>
<td>LC-1801</td>
<td>Measurments: Coils 1uH-200H, Caps 1pf-200uf, Res 01-20M</td>
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### Multimeter with Capacitance & Transistor Tester

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<tr>
<th>Model</th>
<th>Description</th>
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<tbody>
<tr>
<td>CM-1500B</td>
<td>555</td>
<td>$2,595.95</td>
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### Quad Power Supply

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<th>Model</th>
<th>Description</th>
<th>Price</th>
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<tr>
<td>XP-580</td>
<td>2-20V @ 2A, 12V @ 1A, 5V @ 2A, -5V @ 5A, Fully regulated and short circuit protected</td>
<td>$69.95</td>
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### Triple Power Supply

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<thead>
<tr>
<th>Model</th>
<th>Description</th>
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<tbody>
<tr>
<td>XP-620</td>
<td>Assembled 75, Kit $50, 2 to 15V @ 1A, (or 4 to 30V @ 1A), and 5V @ 3A, All the desired features for doing experiments, Features short circuit protection, all supplies</td>
<td>$26.95</td>
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### AM/FM Transistor Radio Kit with Training Course

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<tr>
<th>Model</th>
<th>Description</th>
<th>Price</th>
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<tbody>
<tr>
<td>AM/FM 108</td>
<td>14 Transistors + 5 Diodes, Makes a great school project</td>
<td>$27.95</td>
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### True RMS 4 1/2 Digit Multimeter

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<thead>
<tr>
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<th>Description</th>
<th>Price</th>
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</thead>
<tbody>
<tr>
<td>M-700T</td>
<td>0.5% DC Accuracy, 1% Resolution with Freq. Counter, Data Hold</td>
<td>$135</td>
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### Digital - Section

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- Two no bounce logic switches
- 8 LED readouts TTL buffered
- Clock frequency 1 to 100KHz
- Clock frequency 0.5MHz square wave

### Breadboards

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<thead>
<tr>
<th>Description</th>
<th>Price</th>
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<tbody>
<tr>
<td>2 breadboards, each contain 840 tie points (total 1,680)</td>
<td>$159.95</td>
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90 WILL BEASTY CABLES IMPROVE YOUR AUDIO?
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Fuel cells could potentially be the most efficient and environmentally clean source of power ever developed. Fuel cells are an attractive alternative to conventional power generation because they are highly efficient, and produce drinking water as an added by-product. What more could you ask for in an energy source? The principle of fuel cell operation was discovered by Sir William Grove in 1839. He found that electricity could be generated by supplying hydrogen and oxygen to two separate electrodes immersed in sulfuric acid. For more than a century, however, fuel cells remained a mere curiosity.

The theory of fuel cell operation defied commercial applications for so long because of technical and financial obstacles. It wasn't until the 1960's, during the growth of the space program, that there was a renewed interest in developing fuel cell technology into a viable energy alternative to standard power generation.

There are two important concerns in conventional power generation: efficiency and pollution. Most of the power in the world is generated from heat engines using the heat from combustion of fossil fuels. Mechanical systems involve many energy conversion steps, and their efficiencies are limited by the laws of thermodynamics. That results in considerable power losses.

Fuel cells operate by converting potential chemical energy of fuel into electricity. It operates at a constant-temperature during the electrochemical process, therefore it's efficiency is not limited by thermodynamic laws governing heat engines.

Pollution is a result of combustion, industrial processing, and vehicle exhaust. Those pollutants consist of unburned fuel, partially burned fuel, carbon, carbon monoxide, carbon dioxide, dust, sulfur dioxide, nitrous oxides and so on. Waste heat from power plants warms up the rivers, causing havoc to the natural balance of fish and wildlife. And we all know of the devastating effects of acid rain, which results from man-made emissions of sulfur and nitrogen in the air. The by-product of a fuel-cell reaction, however, is water. Who would object to that?

Fuel-cell chemistry
Fuel cells operate by converting the potential energy of certain chemical reactions directly into electrical current in a flameless, catalyzed reaction. Some types of fuel cells work very well at room temperature.

A basic fuel cell consists of an anode (+) and cathode (−) separated by a conducting electrolyte such as a solution of potassium hydroxide. A fuel, such as hydrogen gas, or hydrazine, is introduced to the negative electrode where it is oxidized, releasing electrons to the load. Oxidation is the process of removing one or more electrons from an ion or molecule. In fuel cells, hydrogen ions are formed at the electrode by electrochemical oxidation of the fuel. If the fuel is hydrogen, hydrogen ions are created by the following ionization reaction:

$$\text{H}_2 \rightarrow 2\text{H}^+ + 2e^-$$

Oxygen, air, or hydrogen peroxide (a source of oxygen) is fed to the cathode, where it is reduced, whereby the O₂ oxygen molecule splits apart. Ionic conduction completes the circuit through the electrolyte. Hydrogen and oxygen react to form water, as this chemical equation shows:

$$2\text{H}_2 + \text{O}_2 \rightarrow 2\text{H}_2\text{O}, \text{or} \quad \text{Hydrogen + Oxygen \rightarrow water}$$

If hydrazine is oxidized, additional nitrogen is formed which is a normal constituent of air, and also safe:
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You may be tempted to say that if hydrogen is such a "clean" fuel, we can just burn hydrogen in air and get pure water as the combustion product plus power. Burning hydrogen would indeed be a considerable improvement over burning coal, oil, or gasoline. However, when air is burned, a large amount of nitrogen is drawn into the combustion chamber and heated to roughly 1000°C. At that temperature, it partially reacts with oxygen and forms oxides of nitrogen. So, even though the reaction product of the main reaction is pure drinking water, the side reaction spoils it all by making the resulting water unsuitable to drink. If hydrogen and oxygen react in a fuel cell at room temperature, that problem is eliminated.

**Space-age power**

The desirable characteristics of fuel cells led to the development of various systems ranging in size from 5-watt portable units, to the kilowatt (kW) power level for military applications, on up to large stationary plants delivering megawatts of power. The lower-power fuel cells were designed primarily for the space program and front-line military use where ease of operation, low maintenance, and low noise are important.

Fuel cells are used solely for power generation of space crafts because of one chief advantage: when power is required for more than a few hours, the battery weight per kilowatt-hour as a function of its operational life is far superior to that of conventional battery cells. A relatively light-weight fuel cell can have a lifespan of five to ten times that of a primary battery.

Fuel cells built between 1960 and 1970 for the Gemini and Apollo space missions and in 1980 for the Space Shuttle Orbiter are among the most successful fuel cells to date. They were needed because of their chief advantages over batteries—weight and lifespan. Those fuel cells used cryogenic reactants of hydrogen and oxygen.

Some space-craft power generation systems use solid polymer electrolyte (SPE) technology in the construction of their fuel cells. That type of fuel-cell assembly consists of an ion-exchange membrane-electrode system with gas distribution, current collection, heat removal, and water management. Many of those assemblies are bolted together between end plates to form an SPE stack assembly.

The Gemini system used three 1-kW SPE fuel-cell stacks. The Apollo system used a larger 1.5-kW fuel-cell stack based on a concentrated 45% potassium-hydroxide electrolyte. The Apollo power plant was designed to operate for over 400 hours. The fuel cell in Apollo 8 lasted for 440 hours, the system produced 292 kWh of power, and 100 liters of water.
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The Space Shuttle system was more advanced in design than either the Gemini or Apollo fuel cells. The Space Shuttle fuel cells are 20 kilograms lighter and deliver six to eight times as much power. Each fuel cell power plant consists of a power section where the chemical reaction occurs, and a compact accessory section connected to the power section, which controls and monitors the power section’s performance. The three fuel-cell power plants are coupled to the hydrogen and oxygen reactant subsystem and the power distribution subsystem. The fuel cells generate heat and water as by-products of electrical power generation. The excess heat is directed to Freon coolant loops, and the water to a potable water storage subsystem.

Some power specifications of each fuel-cell power plant are:
- 2 kilowatts at 32.5 VDC.
- 12 kilowatts at 27.5 VDC.
- 7 kilowatts continuous power.
- 12 kilowatts peak.
- All three fuel cell power plants are capable of supplying a maximum continuous output of 21,000 watts with 15 minute peaks of 36,000 watts.

Some experimental fuel cells have been considered for use with vehicles. The major prohibiting factor in their use is the difficulty in reliably containing hydrogen gas, and the possibility of an explosion. Also, special fuels such as hydrogen, methanol, and hydrazine are more expensive than hydrocarbon fuels.

Many advanced fuel-cell designs have been developed for power utility applications, but because of the typical problems of fuel storage and cost effectiveness, they have not been widely used.

An experimental fuel cell

The author was able to build a successful experimental fuel cell by the technique described below. We must, however, issue this word of caution: This product should NOT be built or experimented with in any way except under the direct supervision of someone who is highly qualified in the fields of chemistry or chemical engineering. Some chemicals and gaseous by-products in a fuel cell could be toxic and/or explosive! All dangerous chemicals are listed in the sidebar. You must be familiar with proper handling and disposal of any chemicals used.

The author’s experimental fuel cell uses two adjoining chambers separated by a membrane, as shown in Fig. 1. An electrode with catalytic properties is placed into each chamber. Both chambers are filled with a liquid electrolyte. One electrode is then purged with hydrogen gas, the other with oxygen or air, and a voltmeter is connected across the electrodes. In order to be able to build a fuel cell you should be familiar with semipermeable membranes and catalysts. Semipermeable means that only some ions can pass through it but other matter is retained. In actual applications, separation of ions is not perfect, and some leakage usually occurs, and is permissible. Total blockage on the other hand would inhibit a reaction. The following materials could be used as semipermeable membranes:
- Unglazed discs of baked clay (an old clay flower pot).
- Fine glass frits (the partly fused mixture of sand and fluxes which glass is made of).
- Cellophane.
- Wet plaster.
- Moist, or hardened cement.
- Zinc oxide or zinc chloride cement.
- Certain types of plastic foam.
- Silicic acid gel, prepared by slowly acidifying sodium silicate solution.
- Gelatin saturated with salt.

Chemicals used in fuel cells

A catalyst is a compound that hastens reactions without actually taking part in the reaction. If you set up a H2/O2 fuel cell with sulfuric acid and carbon electrodes for instance, there will be no electrical energy generated. If platinum- or palladium-coated carbon electrodes are used, the reaction gets going. Union Carbide has used this method and supply such electrodes.

The method the author used to plate carbon was to wrap platinum wire and a platinum net around the carbon rods, which works very well. An easy and low-priced way of producing a large surface of palladium is to coat nickel netting with palladium. That can be done by immersing a
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nickel net in a 2% solution of palladium dichloride over night. The coating looks black. Palladium coated nickel acts like pure palladium. The author had a supply of platinum on hand or he would have used the approach just mentioned.

The amount of palladium dichlorides you need costs about $20.00. Platinum, palladium, silver, nickel (especially Raney nickel) have been used as catalysts in different fuel cells. Platinum-group metals work so well because they have an affinity to hydrogen and will pick up consid-
erable amounts of it for storage in their crystal lattices. A platinum electrode saturated with hydrogen, therefore, is practically an electrode of solidified hydrogen. The pure metal is too expensive, so palladized nickel, platiniized carbon or Raney nickel on a carrier matrix are the first choice.

Impinger-type glass tubes with frits or aquarium-type dispersion tubes are used as gas inlet tubes. The electrodes are wound around the tube in a coil. Copper wire leads are connected. The electrolyte is a 30% potassium hydroxide solution. Oxygen and hydrogen can be bought in small laboratory bottles with reasonably priced lab-reduction valves.

Hydrogen can also be produced from zinc and diluted hydrochloric acid. That leaves you with a solution of zinc chloride which is hazardous to the environment and must be disposed of in a manner prescribed by law.

The entire experiment was conducted in the open air in order to allow the flammable hy-
drogen to disperse. Rotameters were used to check gas flow. They can be replaced by bubble indicators if you prefer. Gas flow was 10-20 liters per hour (l/h) but can be varied. Oxygen flow should be about ½ that of hydrogen flow. The reaction is sluggish at the beginning as hydrogen has to saturate the platinum metal surface.

An indication of about 10 mV may occur for several minutes, which will then rise. There may be steps in this rise, therefore it may be necessary to put a little drain on the system by using a 100-ohm resistor connected across the 2 chambers. It can be removed again after a few minutes. That helps overcome polarization effects. The author measured 998 mV after about 10 minutes. To compensate for the slow start, the cell will generate a voltage for some time after the hydrogen is turned off.

After you finish, the potassium hydroxide solution should be poured into a well-capped plastic bottle. It can be used over again, but it will accumulate carbonate which makes it less effective. Some prefer diluted sulfuric acid for the same purpose because it keeps longer. Air can, in most cases, be substituted for oxygen. The amount must be raised, however, since only ½ of air is oxygen. Hydrogen peroxide can be used in place of oxygen but it dilutes the electrolyte.

Hydrogen can be replaced with hydrogen-containing gases such as “city gas” produced from coal, containing hydrogen, methane, and carbon monoxide. Several variations of fuel cell compo-
ts that react at room temperature are shown in Table 1. The fuel cell can also be used as a one-shot unit for liquid fuel, namely hydrazine, and 30% hydrogen peroxide. Both compounds are rocket fuels but can be controlled very well. They are, however, highly toxic and poisonous. Because hydrazine is known to be a carcinogen, one should not work with it unless you are familiar with handling very poisonous substances. Hydrogen peroxide at 30% concentration will bleach your hands and should also be handled very carefully.

<table>
<thead>
<tr>
<th>Fuel</th>
<th>Oxidant</th>
<th>Electrode Material</th>
<th>Electrolyte</th>
<th>Catalyst</th>
<th>Recorded Voltage (mV)</th>
</tr>
</thead>
<tbody>
<tr>
<td>hydrogen 20 l/h</td>
<td>oxygen 10 l/h</td>
<td>carbon</td>
<td>5% sulfuric acid</td>
<td>none</td>
<td>No reaction</td>
</tr>
<tr>
<td>hydrogen 20 l/h</td>
<td>oxygen 10 l/h</td>
<td>carbon/platinum</td>
<td>5% sulfuric acid</td>
<td>platinum</td>
<td>533</td>
</tr>
<tr>
<td>hydrogen 20 l/h</td>
<td>air 40 l/h</td>
<td>carbon/platinum</td>
<td>5% sulfuric acid</td>
<td>platinum</td>
<td>469</td>
</tr>
<tr>
<td>hydrogen 20 l/h</td>
<td>oxygen 10 l/h</td>
<td>platinum</td>
<td>30% potassium hydroxide</td>
<td>platinum</td>
<td>988</td>
</tr>
<tr>
<td>hydrogen 20 l/h</td>
<td>oxygen 10 l/h</td>
<td>palladium on nickel</td>
<td>30% potassium hydroxide</td>
<td>palladium</td>
<td></td>
</tr>
<tr>
<td>2 ml 24% hydrazine hydrate</td>
<td>10 drops 30% hydrogen peroxide</td>
<td>palladium on nickel</td>
<td>30% potassium hydroxide</td>
<td>palladium</td>
<td></td>
</tr>
</tbody>
</table>

*This reaction was not tried by the author, but works according to literature on the subject.

Fuel cells have been run with “steam reformed” methyl alcohol. At 200°C, methyl alcohol reacts with water to form hydrogen and carbon dioxide as shown in the following equation:

\[ CH_3OH + H_2O \rightarrow 3H_2 + CO_2 \]

or

\[ methyl~alcohol + water \rightarrow hydrogen + carbon dioxide \]

At temperatures higher than room temperature many other reactions are possible. Some of them allow a separation and collection of the water formed. You’re probably wondering why fuel cells are not more widely used. The first big drawback is cost, which is always a primary consideration in power generation. Hydrogen is an expensive fuel compared to other types of fuels, and the storage of hydrogen is still a problem. Perhaps in the future, we’ll use solar energy on a large scale to decompose water into hydrogen and oxygen, which can then be stored. When energy is needed, the two gases can be recombined to water in a fuel cell.
Operates 12-19 VDC for controlled by grip pressure and adjusts saber length.

Visible plasma field is

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Build a fence shocker, solar motor, light, bug zapper, Batt
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If your home or office has more than one telephone extension, you’ve probably had the unpleasant experience of picking up the phone only to find it already in use. You may get an angry response from the person on the other end. If a modem is in use, you’ll be greeted by the obnoxious squall of two computers exchanging bits. Such an interruption usually means a lost connection, or the corruption of a file being transferred.

A solution to that problem is the Phone Sentry—an inexpensive, simple, reliable indicator that warns you when a phone extension is in use. The Phone Sentry is easy to build and install in one evening, and presents no load to the phone line. It’s small, inconspicuous, and costs only $5 a copy.

How it works
To understand how the Phone Sentry works, you need to understand how the telephone system works—or, at least, how the local subscriber loop works, since that’s the part that enters into your house.

The telephone line is held at about 45 volts DC by the local switching office when it’s hung up. When a telephone is taken off its hook, a 1K load brings the line down to 6 volts DC. The line stays at 6 volts DC until you hang up, then it returns to 45 volts DC and is disconnected.

The Phone Sentry operates by monitoring the telephone line voltage and switching on a flashing LED whenever the voltage drops below 20 volts. The Phone Sentry can be placed anywhere on a phone line, not just on an extension in use.

Circuit operation
The Phone Sentry circuit is deceptively simple, yet elegant in design. At the heart of the circuit is IC1, a CMOS CD4093B quad NAND gate Schmitt trigger.

Ordinary CMOS gates switch midway between the voltage of the positive and negative supplies. For a circuit powered from 5 volts, this point (called 0.5 V+) is 2.5 volts. When the input voltage rises past or falls below that point, the output will switch. Normally, that’s a desirable characteristic, and is one of CMOS’s good points. However, when a CMOS input is presented with a slowly changing or noisy input, the symmetrical switching characteristic can cause the circuit to jitter or oscillate as the input nears the 0.5 + V point.

The Schmitt trigger input handles noisy environments by separating the rising and falling voltage-switching points. A Schmitt trigger input will react to a rising input voltage only when it passes a threshold that is higher than 50% of the supply voltage, usually about 70%, or 0.7 + V. A falling input voltage will cause a change only when it falls below a much lower threshold of about 30% of the supply, or 0.3 + V. An input voltage between those two thresholds will have no effect until it rises above 0.7 + V, or falls below 0.3 + V.

The region between the 70% and 30% switching levels is called the hysteresis gap, or dead band. Hysteresis permits a Schmitt trigger input to respond very cleanly to noisy or irregular input signals. It also permits some fancy tricks, such as one-gate oscillators. It is the latter capability for which a Schmitt NAND gate is used in the Phone Sentry.

Figure 1 shows a block diagram of the Phone Sentry. The four gates of the CD4093B are used as three separate elements. One Schmitt-trigger NAND gate acts as an input comparator to monitor the phone line. It in turn gates another NAND gate as an input comparator to monitor a phone line. It in turn gates another NAND gate as a oscillator, which drives a high-current buffer for LED1.

The schematic of the Phone Sentry is shown in Fig. 2, with its circuit waveforms at critical locations shown in Fig. 3. Bridge rectifier D1–D4 eliminates any phone-line polarity problems. It also removes the 80-volt peak-to-
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peak ring signal, which could damage the Phone Sentry or make LED1 flicker.

The output of the bridge rectifier is divided down by R1-R2, with 27% of the input voltage reaching IC1-a. 27% represents the voltage divider of the \([R2/(R1 + R2)]\) ratio, which equals \([1 \text{ megohms}/(1 \text{ megohm} + 2.7 \text{ megohms})] = 0.27\)

The bridge always presents two of the four diodes as a phone-line load, D1-D4 or D2-D3, dropping the line voltage down by 0.7 volts DC each, or 1.4 volts total. Since the input impedances of pins 12 and 13 of IC1-a are almost infinite, they draw no current. What appears across R1 and R2 in series should be about \(45 \text{ V} - 1.4 \text{ V} = 43.6 \text{ V}\).

The voltage at pins 12 and 13 with the phone hung up is therefore \(43.6 \text{ V} \times 0.27 = 11.78 \text{ V}\), which is 2.78 volts above the 9-volt DC supply. The IC, however, is protected from overcurrent burnout by R1 and internal diodes. When an extension is in use, the 6 volts on the line goes down to 
\(6 \text{ V} - 1.4 \text{ V} \times 0.27 = 1.24 \text{ V}\).

Capacitor C1 filters out small spikes that can be generated during the ringing cycle, protecting the IC and eliminating any residual tendency of the LED to flicker.

Because the comparator is a Schmitt NAND gate, its output (pin 11) will be low whenever the input voltage is above about 6.3 volts (70% of 9 volts), and high whenever the input drops below about 2.7 volts (30% of 9 volts). Those switching values fit perfectly with the 11.78 and 1.24 volts generated from the phone line by the rectifier and divider. The output will be low when all phones are on-hook, and high when any phone is picked up, or a modem is connected to the line.

The LED could be driven directly by IC1-a, but B1 would be drained in about 10 hours because LED1 draws 10 milliamps when lit. To extend battery life to at least 100 hours, IC1-b, the low 5% duty-cycle oscillator, is gated by IC1-a, driving LED1 and giving a bright flash with much lower current drain.

The output of the comparator is used to gate an oscillator on and off. That oscillator consists of a second Schmitt NAND gate (IC1-b), R3, R4, C2, and D5. When pin 2 of IC1 is held low by the comparator, the output of the gate is held high. That output is used to charge timing capacitor C2, through timing resistor R3. The junction of components R3 and C2 is connected to pin 1. With the output held high, the charge on C2 will rise to the level of the supply voltage.

When a phone is picked up and the loop voltage drops, the comparator's output goes high and the oscillator is enabled. Since both inputs are now high, the output switches low. The charge of C2 is drained, partly through R3, but more quickly through R4 and D5. When the voltage at pin 1 drops below the Schmitt input's lower threshold, the output of the gate switches high, and the capacitor begins charging again through R3. When the capacitor voltage reaches the Schmitt's upper threshold, the output switch-
<table>
<thead>
<tr>
<th>Model</th>
<th>Description</th>
<th>Price</th>
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<tr>
<td>100 Basic</td>
<td></td>
<td>$19.95</td>
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<tr>
<td>150 Basic+</td>
<td></td>
<td>$29.95</td>
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<tr>
<td>200 Advanced</td>
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<tr>
<td>PRO 400</td>
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charge to the lower threshold the first time. Therefore, the first flash of the LED is longer and brighter than those that follow. That's a nice touch, because all of the Phone Sentries in the house will give an initial bright flash when a phone is first picked up to answer a call.

Construction and installation

The Phone Sentry can be assembled on either a PC board, shown here, or on perforated construction board of similar size. The PC board is about the size of B1, so housing the unit is simple, and its construction is straightforward. Figure 4 shows the parts placement diagram; use a socket for IC1, and install it using proper anti-static handling techniques.

The Phone Sentry is small, with several installation options. Once you decide how to mount it, you can select how to wire both the phone line and LED1. If you put the Phone Sentry inside an extension or a wall-mount jack, then solder a foot of 22-AWG wire to each input terminal.

If you use a small case for plugging into a wall socket, solder the green (tip) and red (ring) wires of a modular plug-to-bare-wire phone cord, and clip the yellow and black wires. You may want to solder LED1 directly to the PC board, or mount it in a visible location with two 6-inch pieces of stiff wire.

You can mount both the PC board and B1 in a standard desk phone. Open the phone and secure both the PC board and battery clip to the baseplate with double-sided foam tape. Drill a small hole in the dialing button escutcheon, and use silicone sealant or an LED clip to mount LED1. Connect the two input wires to the tip and ring wires, insert B1, replace the cover, and plug the phone back in.

If there's no space for the Phone Sentry and B1, use a small plastic box on the side of the phone for the PC board, B1, and LED1, and pass the tip and ring wires through a hole in the box and phone case to the connecting points inside the phone. For a wall phone, mount the same case near the wall jack and run the wiring into the wall jack, so it's independent of the phone.
Here are two simple projects that will allow you to control things using up to eight voice commands.

EXPERIMENTS IN VOICE RECOGNITION

Some of the most fascinating things that electronics experimenters can do are those that seem impossible. Remote control and voice synthesis are two areas of experimentation that were once nearly impossible for hobbyists and amateurs to work with, but integrated circuits have brought both within the reach of even novice tinkerers. Another area that has always been very difficult to work with is voice recognition. And now there is a new IC which brings simple speech-recognition technology within the reach of novice experimenters.

Most voice-recognition projects and experiments have used personal computers as the backbone of the recognition device. A number of voice-recognition expansion cards for both Apple and IBM-compatible computers are available, but they're relatively costly and require the computer in order to be usable. The voice-recognition IC, the VCP200 speaker-independent word recognizer, is a stand-alone device that provides all of the essential elements for speech recognition in a single 20-pin package.

The project
There are a number of applications, both serious and fun, useful and merely entertaining, for the VCP200. Rather than limit this interesting device to a single-purpose project, we are presenting two separate projects: one is suitable for experimentation—and also makes a nifty science-fair project—and the other is less ideal for experimentation but better for actual use in an application of one sort or another. A variety of adjustments and interfacing techniques will be discussed, and some flexible interface and driver circuits will be presented. None of the parts, with the exception of the VCP200 itself, are exotic or costly, and most are probably in your junk box or parts collection.

The experimenter's version is a self-contained device with a microphone and eight indicator LEDs. The addition of a power supply is all that's needed. The project will recognize eight words and short phrases from almost any speaker, and light the corresponding LED in response. Outputs are provided for driving other circuits or devices.

The "working" version of the circuit eliminates the indicator LEDs and their driver ICs, and uses a much smaller PC board. However, it retains the eight outputs and all other circuitry, and is therefore more suitable for building into a motorized model or other project.

Voice recognition
The basic elements of voice or speech recognition have been known for a number of years. Human speech consists of phonemes, which are the smallest individual units of sound that make up words and sentences. The "ah" sound in "father," the "t" sound in "top," and the "rr" sound in "radio" are all examples of phonemes. Any word in a particular language can be created by stringing together the proper sequence of phonemes and spaces of silence. Not all languages use the same phoneme sets; English, for example, lacks a glottal stop and the click found in many African languages.

Electronic voice recognition consists of analyzing the arrangement of phonemes in a spoken sequence and matching them against stored patterns or templates to determine the word or phrase. There are many variations in the actual processes used for each of the three steps: storing the patterns, analysis, and matching. However, the basic techniques used for voice recognition can be loosely grouped into four categories.

In speaker-dependent voice recognition, the intended user of the recognition device "trains" it by carefully pronouncing the list of recognized words, several
times each. The system creates detailed templates, or patterns of that speaker pronouncing those words, and stores them. The system will have a very high success rate in recognizing that speaker pronouncing those words, but it will be less able (if at all) to recognize another speaker saying the same words—and, of course, it will only recognize those specific words that it has been trained to recognize.

A discrete-word speech recognizer can only decode speech when it is a series of separately spoken words. It could not understand "Move the cursor to field one," but the sequence "Goto" (pause) "Field" (pause) "One" would be understood. Speaker-dependent discrete-word recognition systems are the most common types in use.

Speaker-dependent connected-word recognition device must be trained to recognize each different speaker's pronunciation. However, more powerful analysis capabilities allow decoding of words strung together in a long phrase or sentence. This type of recognizer could decode "Move the cursor to field one," but is typically costly and complex. The success rates are also typically lower than for speaker-dependent discrete-word recognition systems.

A much more difficult process is to decode the speech of a variety of speakers. No two people pronounce words in quite the same way. When analyzed electronically and graphically, variations, even with very similar-sounding speakers, are quite marked. That natural variation makes it very difficult for a system to recognize, with a high success rate, the same words spoken by different people.

Speaker-independent voice recognition follows the principle that all speakers have certain similarities in their pronunciation. For example, nearly all speakers pronounce the word "stop" with the following similarities: an initial sibilant ('sss'), a short plosive ('t'), a soft vowel ('ah'), and a final plosive ('p'). By matching selected phonemes and allowing for variation in the matching algorithm, the same words can be identified and decoded from a variety of speakers.

**PARTS LIST**

- **IC3**—VCP200 speaker-independent word recognizer
- **IC4, IC5**—CD4011B quad NAND gate (optional, see text)
- **Other components**
  - **J1**—switch or jumper (see text)
  - **MIC1**— electret microphone
  - **XTAL1**—10 MHz crystal
- **Miscellaneous**: PC board (See text), bus wire, SPST power switch, SPDT mode switch, normally open pushbutton reset switch, 9-volt battery or 8–15 volt DC power supply, 9-volt battery clip, three 14-pin IC sockets, one 20-pin IC socket, mounting screws and standoffs, 4-40 x 3/8-inch screw and nut, hookup wire, solder, aluminum sheet for heatsink.

**Note:** The VCP200 may be available from Radio Shack (it has been discontinued but many stores still stock them) as part number 276-1308, or from VCP, 1 Willings Place, Monterey, CA 93940, for $14.95 postpaid.
The drawbacks to speaker-independent systems are that the number of separately recognizable words is limited, the recognition success rate is generally lower than that of speaker-dependent systems, and the system can be easily fooled by similar words. For example, “swap,” “stat,” “spat,” “spot,” and “spit” all have phoneme patterns that are similar to “stop.” Most speaker-independent word recognition systems will be unable to distinguish between those words.

Most dedicated voice- or word-recognition systems are speaker-independent discrete-word types. Although they have some severe limitations, they excel at simple voice-control tasks involving a few carefully chosen words and phrases. The VCP200 is a speaker-independent discrete-word recognizer.

The dream of designers, control engineers, and science-fiction writers is a system that can recognize normal, connected speech from a wide variety of speakers. Despite much effort, no such system yet exists. The first successful “natural speech” recognizer will almost certainly demand the resources of a dedicated supercomputer to handle the massive analysis and computational steps required. However, keep in mind that speech synthesis, now achieved with single dedicated IC’s, also once required a full-sized computer.

**The VCP200**

The VCP200 speaker-independent word recognizer, from Voice Control Products, Inc. (VCPI), is a mask-programmed Motorola 6804 microprocessor. The 6804 is a 20-pin device that implements most of the standard 6800-series instruction set and capabilities, and contains one kilobyte of onboard ROM. Although an EPROM version is available for user development, production devices such as the VCP200 use a ROM that is mask-programmed at the time of manufacture with the appropriate data and control information. That approach, used for many computationally-based special-purpose devices, is a viable alternative to designing a costly single-purpose chip from scratch.

The VCP200’s ROM contains a phoneme analysis and matching program using a proprietary algorithm. The algorithm analyzes a modified voice input signal and matches it against a selection of stored word-recognition templates to identify twelve different words and short phrases: Yes, No, On, Off, Lights, Left, Turn, Reset, Stop, Slow Reverse, Turn Right, and Go.

The chip is switchable between On/Off and Command modes. In the On/Off mode, it recognizes only the two word pairs On/Off and Yes/No. In the Command mode, it recognizes the other eight words and phrases. A separate output for each word is provided, which is latched low when the word is successfully recognized. If the VCP200 cannot find a close match among its word templates, all eight outputs are left high.

The VCP200 is virtually a stand-alone device, requiring only a 10-MHz crystal and four passive components for operation. The only outside circuitry that is required is a special input amplifier, built from a common op-amp, that delivers a sharply clipped and amplified voice signal. That quasi-digital signal can be easily analyzed by the microprocessor.

The VCP200’s biggest disadvantage is the limited and non-expandable word list. However, considering that the chip is inexpensive and easy to use, that limitation shouldn’t bother anyone who is interested in exploring voice-recognition technology without making a heavy investment of time or money.

Unfortunately for experimenters, VCPI regards the VCP200’s program and word-recognition algorithm as proprietary information. Few details are available, and VCPI’s literature and documentation discusses the technology only in general terms. An interesting exercise for the advanced experimenter would be attempting to work out the essential elements of the algorithm, using standard reference information on voice recognition, digital analysis of analog signals, and pattern matching.

The pinout of the VCP200 is shown in Fig. 1. The chip is powered from a single-ended 5-volt supply, which connects to pins 3, 6, and 1, and must provide about 15 milliamperes. Its oscillator crystal connects to pins 4 and 5, each of which must also be tied to ground via 27-pF capacitors to complete and stabilize the oscillator tank circuit.

Pins 2, 16, 17, and 18 of the VCP200 are not used in a standard application. They are special-purpose control pins that are usually tied to +V or ground, and are connected that way on our PC board. Generally, these pins may be ignored, as they are normally used to set the VCP200 into various test and special-application modes that are not useful to the experimenter.

The reset input, pin 20, is held high for normal operation and brought low for a reset. A simple resistor-capacitor pair connected to this pin will cause a power-on reset. The VCP200 can be manually reset by strobing the pin low at any time, by holding it low, you can safely disable the chip’s operation.

Pin 19 is the operation-mode select input. When this pin is high, the chip is set to the Yes/No mode, and only Yes/Off (pin 9), No/On (pin 8), and Not Sure (pin 10), which indicates a recognition failure, are active. When pin 19 is low, the VCP200 is placed in the Command mode, and all eight outputs are active, with each corresponding to a different recognized word or phrase.

The VCP200’s audio input, pin 7, requires an input signal that is either quiescent, or swings past the digital logic thresholds. That requirement translates into a highly amplified, sharply clipped signal that is “shut off” when it is not of sufficient amplitude. Such a signal is easy to achieve with a standard op-amp, as we’ll see.

Finally, pins 8 through 15 are the VCP200’s outputs. During or after a reset (pin 20 brought or held low), all eight outputs are held high. When the chip successfully recognizes a word or phrase in Command mode, the corresponding output will be latched low until the next recognition attempt occurs. If the VCP200 fails to find a match to an input signal, all eight outputs will remain high. In the Yes/No mode, during or after a reset, pins 8, 9, and 10 (as well as the five unused outputs, pins 11-15)
FIG.3—COMPLETE SCHEMATIC of the voice-recognition circuit. The jumper J1 may be replaced with an SPDT switch to control the operating mode. Either way will work, but the switch is easier to use.
will be high. Some recognition failures in the Yes/No mode can also cause all three active outputs to go high.

The circuit
As said earlier, there are two versions of the circuit. A block diagram of the experimenter's version is shown in Fig. 2. The circuit contains a power supply, an input amplifier and comparator, the VCP200, and output drivers. The power supply is quite conventional, using IC2, an LM7805T 5-volt regulator.

The input amplifier is not a conventional design; the output signal, if it were connected to a speaker, would be quite distorted and unlistenable. The purpose of the two-stage amplifier, with its overall gain of about 800, is to increase the microphone signal to a useful level. The output is then passed to a comparator that keeps the final output signal either quiescent (flat-line) or switching between the supply limits—a quasi-digital signal. The output of the amplifier is passed to the VCP200's audio input, where the signal can then be analyzed.

The VCP200's eight outputs are made available, via current-limiting resistors, so that external interface circuits may be added to control motors, solenoids, and other active elements. Eight LEDs are added to give a quick and easy indication of the circuit's response. The LEDs are driven by CMOS buffers, which isolate the LEDs from the outputs, preventing either the indicators or any outlying circuits from interfering with each other.

Figure 3 shows the complete schematic for the experimenter's version of the voice-recognition project. Note that the schematic of the working version would be exactly the same, except for the omission of LED's 1-8 and IC4 and IC5.

Power for the voltage regulator, IC2, can be from 7.5 to 15 volts. Since the circuit draws only about 22 milliamps peak, a 9-volt battery is a good choice. Capacitors C10 and C11 filter and stabilize the regulator's output.

The signal from the electret microphone, MIC1, is coupled to the LM324A op-amp, IC1, through C1. The amplifier uses IC1-a and IC1-b to form a two-stage device that amplifies the microphone signals with a gain of 500-800. That transforms the weak input signal (under 5 mV) to a signal that swings from one output limit to the other, often with considerable clipping.

The amplifier has a restricted bandwidth, with a more or less flat response from about 500 Hz to 9 kHz. Signals under 300 Hz and over 15 kHz are sharply attenuated. That covers the

FIG. 4-AMPLIFIER (a) AND COMPARATOR (b) output waveforms. The comparator converts the amplifier signal into a clipped, quasi-digital 4-volt p-p signal only when the amplitude of the amplifier's signal exceeds the comparator threshold.
VCP200's input range of 300 to 5500 Hz, with some additional headroom for the easily-lost higher frequencies. The amplifier's characteristics are important, because the quality of the input signal largely determines how well the voice recognizer will work.

To keep the VCP200's input quiet, unless a signal of sufficient strength is present, and to ensure a sharply clipped signal, quiet, unless a signal of sufficient strength is present, and to ensure a sharply clipped signal, the output of the amplifier is passed to a comparator, IC1-c. (The fourth op-amp on the LM324A, IC1-d is not used, and its pins are left unconnected.) The comparator's output remains steady unless the input signal swings past its threshold. Input signals of less than 2.5 volts peak-to-peak will be ignored. However, all signals stronger than that will cause the comparator's output to swing from limit to limit, or about 4 volts peak-to-peak, which is within one-half volt of each supply rail. A comparison of the amplifier and comparator output signals is shown in Fig. 4.

Since the LM324A is operated from a single-ended supply, a "false ground" or offset voltage must be provided. The offset, along with the comparator threshold voltage, is provided by the voltage divider string R5-R8-R9. The amplifier offset is provided by the upper junction, and the comparator threshold by the lower; C6 and C7 stabilize those voltages. That design forces the center voltage of the op-amp's output signal to be separated by a volt or so from the comparator's threshold, and is the key to correct operation. Adjusting the divider string is one of the ways that the circuit's performance can be modified.

The output of the comparator is then routed to pin 7 of the VCP200. The 10-MHz crystal, XTAL1, provides the chip's master clock frequencies, with the oscillator tank circuit completed and stabilized by C9 and C10. To provide a power-on reset, the RC pair R11 and C8 hold the VCP200's reset input low for a few milliseconds after power comes on. As C8 charges through R11, the reset pin is brought high, resetting the VCP200.

The VCP200's mode input, pin 19, is controlled by setting JU1. On the PC board, JU1 is actually three pads which may be connected to an SPDT switch, or simply jumped by. However, jumping is not recommended; a switch will make it easier to experiment with both operation modes. The eight outputs, pins 8 through 15, are left open for the experimenter to use as necessary. Since the outputs are active-low, they can sink about 10 mA and source somewhat less. That is sufficient enough to drive logic devices and transistor drivers. If high-current devices such as relays or motors are to be driven, a buffer/driver must be used. To prevent damage to the VCP200 from an accidental overload of an output, 470-ohm current-limiting resistors (R12-R19) are provided. They limit the output current to about 9.5 mA, even under worst-case conditions.

The eight LED indicators, LED1-LED8, are driven from CMOS drivers IC4 and IC5, which are CD4011B quad NAND gates. However, several other common chips could be substituted here, among them the CD4001B quad NOR gate and the CD4093B quad NAND Schmitt trigger.

**Construction**

Foil patterns are provided for both versions. Although a PC board is recommended, perforated construction board and point-to-point wiring could also be used. If you use point-to-point construction, be sure and keep all wiring, especially in the area of the input amplifier, short. The very high gain of the amp will cause it to pick up and amplify electrical noise if excessively long connecting wires are used. You should use sockets for all the IC's to make them easier to replace if necessary.

If you are going to build the experimenter's version of the project, follow the parts-placement diagram shown in Fig. 5. If you are going to build the smaller "working" version, simply use the smaller foil pattern; parts placement is the same as the larger version, except that the...
LEDs and their drivers, IC4 and IC5, are left out. On both, keep the wire jumpers and resistors close to the board. Insert the disc capacitors so that their bodies are seated against the board, but don’t chip the dielectric material. Be careful to observe the polarity on the two electrolytic capacitors, C1 and C11.

The voltage regulator, IC2, requires special mounting. The middle lead should be bent about 0.1 inch farther from the board than the two side leads, and all three bends should be made so that the regulator’s mounting hole lines up with the hole in the board (see the photo in Fig. 6 for details.) If you are going to be using the project by itself, with no outlying devices powered from the board, no heatsink is needed for the regulator. If you are going to be powering other devices from the regulator that will increase the load to more than 100 milliamps, a heatsink should be added to the regulator. A flat aluminum stock heatsink can be bent into a shallow “U” shape and installed under the regulator. Because there is no space for a large heatsink, the current draw from the regulator should be limited to no more than 250 mA even when using as large a heatsink as possible.

For most experimenters, mounting MIC1 directly to the board will be adequate. In some cases, though, it may be better to mount the microphone remotely. In that case, light-gauge shielded cable should be used to connect the microphone to the board. Electret microphones are polarized, so be sure the positive terminal is connected to the pad that leads to C1 and R1.

If you like, the eight LED indicators can be mounted remotely with a length of ribbon cable. If you mount them on the board, be sure to position them all at an even height. How you finish the remaining steps depends on how you want to use the board. For display and experimentation, you’ll want the input and outputs of the circuit easily accessible with test points. Otherwise you can hardwire driver circuits and the like directly to the board.

In the prototype, the PC board and power switch are mounted to a thick plastic base using spacers and screws, and the battery clip is secured by smaller screws. Although the prototype has no reset switch and is strapped into the Command mode, you can easily add the controls. Just use a slightly larger mounting base and mount the switches in the same manner. If you are using the working version, and will be using it as a part of a complete project or more complex setup, use your judgment as to mounting the board.

Testing

When you have the board (either style) finished, leave the IC’s out of their sockets and connect the power terminals to 9–15 volts DC. Then check for +5 volts DC at pin 3 of the regulator, pin 4 of IC1, pin 2 or 3 of IC3, and (with the experimenter’s unit only) pin 14 of IC4 and IC5.

Disconnect power, insert IC1, and then connect power again. Check for an AC voltage at pin 7 of IC1. It should vary with the level of sound up to about 2 volts peak. Check the voltage at pin 14 of IC1. When the sound level is high enough, the 2-volt signal should be present. If the comparator is functioning correctly, pin 14 should switch between no signal and a 2-volt AC signal, with nothing in between. If you’re using an oscilloscope, look for a 0–4 volt signal at pin 7, and a 0 or 4 volt clipped signal at pin 14.

Once the board has passed these tests, remove the power and insert the rest of the IC’s. When you reconnect power, all LED’s should remain off, or if you’re using a board without the indicators, all of the outputs should be high. Say “Go.” The appropriate LED should light (or the output will go low). Try the other phrases to make other LED’s light. Don’t worry if the circuit doesn’t seem to respond well—it takes a little practice to speak the words and phrases clearly enough for the VCP200 to understand. Table 1 explains how to pronounce the words so that the VCP200 will understand them.

Modifications

The gain of the amplifier may be adjusted by changing the value of R4, R7, or both. Adjusting R7 is preferred. The higher the resistor values, the higher the gain of the amplifier. Lowering the gain will lessen the circuit’s sensitivity to background noise, but will require the operator to speak rather loudly and directly into the microphone. Raising the
FIG. 7—BASIC POWER DRIVER CIRCUIT for interfacing the project to motors, lamps, or other high-current devices. The relay must have a 5-volt coil, but can have any arrangement of contacts suitable for the application.

FIG. 8—TOGGLED, LATCHING interface circuit. The output switches states on successive occurrences of the associated voice command.

FIG. 9—THIS CONTROL CIRCUIT allows the project to be latched into forward or reverse motion while permitting other voice commands to be processed.

gain will allow softer speaking from a greater distance, but at the expense of greater sensitivity to noise.

The frequency response of the amplifier is about 300 to 9000 Hz. Since the VCP200 responds to frequencies from 300 to 5500 Hz, reducing the upper cutoff point of the amplifier to 6000 or 7000 Hz would probably make it less sensitive to noise. If you are familiar with op-amp circuit design, a good way to improve the project would be with a high-precision bandpass amplifier. It should have a nearly flat response from 500 to 6000 Hz, with a sharp rolloff (third-order or better) at each end. The flatter the bandpass response and the sharper the cutoff points, the better the overall performance is likely to be. Higher frequencies are more sharply attenuated by distance and may need extra boost (actually less cut) in order for the VCP200 to successfully decode them.

The comparator threshold is set by the lower output of the resistor divider string R5-R8-R9. Since the artificial ground level or offset voltage of the amplifier is set by the upper output of the same string, some care is needed when adjusting either voltage so as not to disturb the other. The amplifier offset should be kept as close to +0.5 volts (or 2.5 volts) as possible, to ensure proper amplifier operation (i.e., balanced clipping of the signal). Thus, the series value of R8 and R9 should always be the same as that of R5.

If the comparator threshold is very close to the amplifier offset, very low-level sounds will be "digitized" by the comparator and make their way to the VCP200's input. That would permit better operation over distance or with softly-speaking users, but would make the unit prone to interference from noise. If the comparator threshold is set further from the amplifier offset, noise will be rejected but louder speech or shorter-range operation will be required. For extensive experiences, continued on page 57.
HOW OFTEN HAVE YOU WISHED YOU had one of those fancy frequency generators that let you set your frequency accurately without having to fiddle with the uncalibrated tuning knob? Without a high-quality frequency counter and without nearly infinite patience, it is impossible to keep your audio oscillator on frequency without constant tweaking. If you eliminate the expensive extras, while retaining resolution and stability, you'll end up with the synthesizer project presented in this article.

While this synthesizer doesn't have the features of some very expensive products, it does provide 1-Hz resolution at over 500 kHz with crystal-controlled precision, all for less than $70. Once you add a suitable enclosure and power supply, you'll have a digital frequency synthesizer small enough to fit on even the messiest workbench. And it's perfect for providing that odd-ball frequency that your new project needs.

**Direct digital synthesis**

As the name implies, direct digital synthesis (DDS) is a method of frequency generation that uses digital methods rather than the traditional analog oscillator, phase-locked loop, or bank of crystals. The availability of fast digital circuits and D/A converters make this technology available to the average electronics enthusiast.

A review of trigonometry is important to the understanding of DDS before delving into the details of the electronics. Figure 1 shows a circle with a radius whose length is arbitrarily set to one. The radial line labelled R is allowed to rotate about the circle through an angle $P$, which will be referred to as the phase. Drawing a horizontal line from the tip of $R$ until it intersects with the vertical axis defines the length $S$ shown in the figure. As the radius $R$ is allowed to make a complete rotation around the circle, the length of $S$ takes on all values between $+1$ and $-1$, while $P$ varies from 0 to 360 degrees. The length $S$ is precisely the sine function of $P$ sin($P$), shown in Fig. 2-a.

If, rather than allowing $R$ to rotate smoothly around the circle, we make 8 equal steps around the circle, then the values of $S$ form the stepwise approximation shown in Fig. 2-b. As the number of steps are increased, the ap-
FIG. 1—THIS CIRCLE HAS A RADIUS whose length is arbitrarily set to one. As \( R \) rotates around the circle, \( S \) takes on all values between \(-1\) and \(1\).

\[ \begin{align*}
\text{FIG. 2-THE LENGTH } S, \text{ AS } R \text{ rotates, is the sine function of } P \ (a). \text{ If we make 8 equal steps around the circle, then we get the stepwise approximation shown in } b. \text{ If we make 64 steps, the approximation becomes closer to the actual sine function (c).}
\end{align*} \]

proximation becomes closer to the actual sine function, with Fig. 2-c showing the approximation for 64 steps. In practice, analog filtering is used to smooth out the steps, as we'll see in a minute.

From this simplified discussion, a method for generating a varying frequency can be derived. Assume that each step occurs at a precisely determined instant, then by varying the step size the number of steps around the circle can be varied. The fewer the steps, the faster the complete circle is covered, hence the higher the frequency of the sine-wave approximation. Note that fewer steps means a coarser approximation to the actual sine function, with the output eventually reducing to a square wave, which points out one of the limits of this technique. All we need now is a circuit that will synchronize the variable-phase steps to a precision clock.

Figure 3-a shows a block diagram of the system. The block labelled Phase Accumulator repetitively adds the value set by the Step Size Programmer to the sum performing the function of stepping the radius (\( R \)) about the circle in equal phase increments. The phase accumulator behaves like a simple counter, except that rather than incrementing its output by one on each clock pulse, the output advances by the value set by the step size programmer on each clock pulse. The block labelled \( \text{SIN}(P) \) converts the value stored in the phase accumulator to a sine amplitude approximation. The step size programmer is simply a bank of DIP switches, the phase accumulator is a series of cascaded 4-bit adders, and the \( \text{SIN}(P) \) block is a sine look-up table contained in an EPROM.

The digital data present at the output of the \( \text{SIN}(P) \) block must be converted to an analog voltage in order to be useful. A method for doing this is shown in Fig. 3-b, which consists of a D/A converter, filter, and output amplifier. The filter helps to smooth out the jagged steps in the sine approximation, while the output amplifier buffers the output of the D/A converter. In the actual

\[ \begin{align*}
\text{FIG. 3-BLOCK DIAGRAM of the phase accumulator and phase-to-sine converter (a), and the block diagram of the digital-to-analog converter and output stage (b).}
\end{align*} \]

\[ \begin{align*}
\text{FIG. 4-PARTIAL SCHEMATIC of the phase accumulator (a) and the converter and output stage (b).}
\end{align*} \]
FIG. 5—HERE'S THE COMPLETE SCHEMATIC for the phase-accumulator circuitry.
implementation, the buffering and the filtering functions are combined.

The frequency resolution of a DDS system is set by the master clock frequency, $f_c$, and the number of bits, $N$, in the phase accumulator. For the binary accumulator that we have here, the resolution is then $f_c/2^N$. If the step size programmer is set to a binary value, $M$, then the output frequency is $M \times f_c/2^N$. The design presented here keeps $M$ less than $N/4$ to minimize distortion at the output.

Circuitry

There are several manufacturers of complete integrated circuits that can perform the digital
portion of the block diagram, but these parts are expensive and not readily available. Figure 4-a shows a partial schematic of the phase accumulator using components that are inexpensive and easy to get.

The complete phase accumulator consists of six 74LS283 4-bit adders, with their outputs latched by three 74LS374 octal D flip-flops. The outputs of the 74LS374's are fed back to the B inputs of the 74LS283, which forces the sum stored in the latches to be added to the value set by the switches on the A inputs. Since the 74LS374 stores data only at the positive edge of its clock input, the fact that the data presented to its inputs will be changing shortly after the clock causes no errors. The delay through the latch and adder guarantee glitch-free operation. At each clock pulse a new sum is present at the output of the latch. The output of each adder then stabilizes with the new sum allowing the cycle to repeat continuously. This sum represents the value \( P \) in the theoretical discussion, while the value set by the DIP switch represents the size of each phase step.

The sine-wave lookup table is contained within a single 2716 EPROM providing phase-to-amplitude conversion. Although 24 bits are available in the phase accumulator as implemented here, only 21 bits are used to maintain compatibility with readily available crystals. For those who wish to program their own EPROM, both a hex dump of the contents of the EPROM and an S-Record formatted hex dump for use with PROM programmers can be downloaded from the R-E BBS (516-293-2283, 1200, 2400, 8N1) in a file named DIGSYN.HEX. A programmed EPROM is available from the source shown in the parts list.

The data in the EPROM represents the values generated by the mathematical function

\[
127.5(\sin(2\pi P/2048) - \pi/2))\]

truncated to 8-bits, with \( P \) taking on values from 0 to 2047, that is, the addresses of the EPROM. The formula offsets the sine function so that its value ranges from 0 to 255 as \( P \) ranges from 0 to 2047 and avoids negative values which would complicate the next stage.

That matches the function to the 2716 EPROM with its 11-bit address space and with its 8-bit output range. A C-program used to generate the values in the table is shown in Listing 1. Since the EPROM has only 11 address lines, only 11 lines from the accumulator are used in this application. The 8 bits at the output of the EPROM are a digital representation of the amplitude of the sine wave and must be converted to an analog voltage before being filtered and buffered.

Since simplicity and low-cost were design goals, the output of the EPROM is latched by another 74LS374, which allows the full clock period for the EPROM output to settle, permitting the use of inexpensive slow EPROM's. The latch also guarantees a glitch-free input to the D/A converter section.

Figure 4-b shows the D/A converter circuitry. The D/A conversion is accomplished using a DAC08 8-bit D/A converter (an MC1408 can be substituted with some loss in performance). The output of the converter is a current proportional to the digital value present on its 8-bit parallel input. The current is set by R8 to a maximum of 1.06 mA. The digital word presented to the D/A varies from 0 to 255, forcing the current output to vary from 0 to \((255/256) \times 1.06 \text{ mA}\). The current is then fed to op-amp IC4-b which converts it to a voltage that varies from 0 to approximately 1.0 volt. The complete schematic for the phase accumulator circuit is shown in Fig. 5, and the schematic for the analog section is shown in Fig. 6.

First-order filtering is accomplished by C9 in this conversion stage. Op-amp IC4-a provides additional filtering to further smooth out the steps in the sine approximation. The output of this two-pole filter is AC-coupled to the output connection. Figure 7 shows the relative response of the filtering provided in the output stage. The corner frequency of the filter is set by the formula

\[
t_0 = \frac{1}{(2\pi VR7 \times C10 \times R6 \times C11)}
\]

which, for the values shown, is equal to 482 kHz. A high-speed

```
LISTING 1

#define \n
#include <stdio.h>
#include <math.h>

main()
{
  double p=0; /* phase input to sin fcn */
  double s=0; /* output value of true sin fcn */
  int s; /* amplitude truncated to 8 bits */

  double sin(); /* true sin fcn */
  double pi=3.141592654; /* amplitude truncated to 8 bits */
  int addr=0; /* address of EPROM */
  int bytes=2048; /* size of EPROM in bytes */

  printf("%4x
", addr);
  if (addr % 16 == 0)
  { 
    printf("%4x ", addr);
    p = 2.0*pi*(double) addr/((double) bytes);
    s = 127.5(1.0+sin(p - pi/2.0)); /* gives 0 at -90 deg */
    if (s >= 0.5) /* rounds if necessary */
    { 
      printf("%2x",s);
      addr++; /* increment address */
    }

  }
  for (addr = 16; addr < bytes;)
  {
    printf("%4x ", addr);
    p = 2.0*pi*(double) addr/((double) bytes);
    s = 127.5(1.0+sin(p - pi/2.0)); /* gives 0 at -90 deg */
    if (s >= 0.5) /* rounds if necessary */
    { 
      printf("%2x",s);
      addr++; /* increment address */
    }
  }

  while (addr < bytes)
  {
    printf("%4x ", addr);
```

R&E EXPERIMENTERS HANDBOOK

31
A high-speed op-amp is required to effectively filter the waveform. The 4558 op-amp we used is a good compromise between performance and cost.

The clock for all functions is provided by a crystal oscillator running at 4.194304 MHz, which happens to be exactly the 22nd power of two. The clock is divided by two to provide the phase-accumulator clock and EPROM latch clock. Additional inverters are used as delay elements to ensure that the latches are clocked at precisely the right instant to prevent glitches. With the clock and timing as such, EPROM's with access times as slow as 475 ns can be used.

With 21 bits of the phase accumulator used and a clock frequency of 4.194304/2 MHz \( f_c \), the output resolution is precisely 1 Hz. Since 19 bits are presented as the input to the phase accumulator by the DIP switch, the maximum output frequency is:

\[
2^{19} \times \frac{f_c}{2^{21}} = \frac{f_c}{4} = 524.288 \text{ kHz}
\]

While a DDS system can approach \( f_c/2 \), \( f_c/4 \) was chosen as a maximum to limit the total distortion in the output waveform. The top frequency is actually 1 Hz less than that because the maximum setting is \( 2^{19} - 1 \) for a 19-bit binary input. The filter rolloff shown in Fig. 7 attenuates clock-related distortion by over 30 to 1.

**Construction**

A double-sided PC board is available from the source shown in the parts list, and we've also provided the foil patterns in case you want to make your own board. If you're using the PC board, follow the parts-placement diagram shown in Fig. 8. Note that IC5 and IC8 are high-speed CMOS and must therefore be handled carefully to prevent
PARTS LIST

All resistors are 1/4-watt, 5%.
R1—1 megohm
R2, R3—10 ohms
R4, R7—3300 ohms
R5—100 ohms
R6—15,000 ohms
R8, R11—4700 ohms
R9, R10—1000 ohms
R12, R13—4700 ohms × 9, 10-pin SIP resistor

Capacitors
C1—5–30 pF trimmer
C2, C3, C6, C12, C18—C20—0.1 μF, ceramic disc
C4, C5—not used
C7, C14, C15—10 μF, 35 volts, electrolytic
C8—100 μF, 16 volts, electrolytic
C9, C10—100 pF, ceramic disc
C11, C17—22 pF, ceramic disc
C13, C16—470 μF, 16 volts, electrolytic (optional for power supply)

Semiconductors
IC1—DAC08CN 8-bit D/A converter
IC2—2716 2K × 8-bit EPROM
IC3, IC15–IC17—DM74LS74N octal latch
IC4—LF353N dual op-amp
IC5—MM74HC04N hex CMOS inverter
IC6—LM7805 +5-volt regulator (optional for power supply)
IC7—LM7905 —5-volt regulator (optional for power supply)
IC8—MM74HC74AN dual D-type CMOS flip-flop
IC9–IC14—DM74LS283N 4-bit adder
BR1—1-amp bridge rectifier (optional for power supply)

Other components
XTAL1—4.194304 MHz crystal
S1, S2—10-position DIP switch
PL1—AC line cord (optional for power supply)
T1—120VAC/12.6VAC transformer (optional for power supply)

Miscellaneous: PC board, solder, case, mounting hardware, etc.

Note: The following items are available from NOVATECH INSTRUMENTS, INC., 1530 Eastlake Ave. E, Suite 303, Seattle, WA 98102 (206) 328-6902:

- Complete kit of parts (except a case and the optional power-supply parts)—$69.95
- Please add $5.00 shipping and handling. Washington State residents must add 8.2% sales tax.

FIG. 8—PARTS-PLACEMENT DIAGRAM. Follow this diagram if you’re using a PC board. The smaller IC outline beneath IC2 is for experimenting with a faster EPROM such as a 74S472 (see test).

FIG. 9—THE COMPLETED UNIT. This compact PC board can easily be installed in almost any kind of project case.

damaging them. Use a grounded-tip soldering iron (if you’ve got access to one) and ground yourself before picking up the board or an IC. Space for the optional power supply (shown in Fig. 6) is not provided on the PC board, but it can be made on any kind of board. The power-supply circuit is not critical, but be careful due to the line voltages present. Figure 9 shows the completed unit.

Since the majority of the circuit is digital, simple wiring techniques can be used. The author’s original prototype was built using wire-wrap methods for the digital section and point-to-point for the analog section. If you’re wire-wrapping the circuit, some care must be applied to the analog section to prevent digital switching noise from getting into the output. The ground returns for all of the analog section must connect to the power supply separately from the digital section and the analog bypass capacitors must be connected as close as possible to the analog integrated circuits. The oscillator, consisting of IC5-a, R1, R4, C1 and
TABLE 1—SPECIFICATIONS

<table>
<thead>
<tr>
<th>Specification</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>Frequency Range:</td>
<td>1 Hz to 524.287 kHz</td>
</tr>
<tr>
<td>Resolution:</td>
<td>1 Hz over complete range</td>
</tr>
<tr>
<td>Accuracy:</td>
<td>Depends on crystal, typ. 0.005%</td>
</tr>
<tr>
<td>Output:</td>
<td>Low distortion sine wave</td>
</tr>
<tr>
<td>Output amplitude:</td>
<td>Approx. 1 volt peak-to-peak, decreases at top end</td>
</tr>
<tr>
<td>Power requirements:</td>
<td>+5V at approx. 250 mA</td>
</tr>
<tr>
<td></td>
<td>-5V at approx. 50 mA</td>
</tr>
</tbody>
</table>

XTAL1, must be tightly wired. The author's original prototype had the discrete components soldered directly to the wirewrap-socket pins of IC5. A board with an existing ground plane is ideal for wire-wrap construction.

**Operation**

Before turning on the power, carefully inspect the board for shorts, solder bridges, wiring errors, etc. Set the DIP switch to any non-zero value. If you have a frequency counter available, connect it to pin 5 of IC8-a. If you don't have a counter, adjust C1 to mid-range; with the crystal specified, your error should be no more than about 0.02%. Apply power and, using an insulated adjustment tool, adjust C1 to exactly 2.097152 MHz if you have a counter connected. With an oscilloscope connected to the output, vary the DIP switch setting; you will see the frequency change. For higher and higher frequencies the distortion will increase with the maximum distortion at the highest setting.

The frequency output is equal to the binary value set by the DIP switches, with a logical 1 corresponding to an "off" position. For a switch setting of 001,1000,0110,1010,0000 (100 kHz), the author's prototype gave the waveform shown in Fig. 10. A spectrum-analyzer display of the 100-kHz output is show in Fig. 11. Note that the harmonics are at least 40 dB down, corresponding to about 1% distortion. Varying the least-significant DIP switch will change the frequency output by 1 Hz. Since the frequency is set by the DIP switch and the accuracy of the crystal oscillator, the output will be the same even after a power-down, power-up cycle. Table 1 summarizes the specifications for the completed digital synthesizer.

**Experiments**

If the digital parts are changed from 74LS to 74F, the EPROM changed to a bipolar PROM (such as a 74S472 which is accommodated on the circuit board), and the clock oscillator replaced by a faster one, the output frequency can be increased at the expense of resolution. The author has successfully operated the circuit up to a 5.0-MHz output frequency, providing 10 Hz resolution. The circuit is simple and compact enough that several units can be built to provide fixed calibration frequencies needed on your bench. High-speed CMOS logic may be substituted for the low-power TTL devices for lower-power operation. If you decide to change to CMOS, IC3 must be a 74HCT374 as the output of the EPROM is TTL-compatible. Advanced CMOS, 74AC or 74ACT, should not be used because of noise induced by its fast edge rates.
Our solid-state Tesla coil can produce sparks as long as 8 inches with a peak output of about 100,000 volts.

Tesla coils have been around for almost 100 years and, with the exception of vacuum-tube driven coils, not much has changed from the way Nikola Tesla invented them. This article describes a new type of Tesla coil; a true solid-state Tesla coil. One thing that makes our Tesla coil unusual is that the coupling to the secondary coil is by a direct electrical connection rather than by magnetic fields. Direct coupling is not new to Tesla coils but it is seldom seen.

The solid-state Tesla coil is by no means as spectacular as capacitive-discharge Tesla coils but it gives just as good, or better, performance as a vacuum-tube Tesla coil. Sparks as long as 8 inches are possible with a power-line consumption of 2 amps at 120 volts (see Fig. 1), and the output reaches a peak of about 100,000 volts. Although the average power input to the device is around 250 to 300 watts, the peak input power to the Tesla secondary coil is about 800 watts. The Tesla coil is an excellent teaching tool, as many interesting things may be learned with the aid of this device.

Circuit description
The schematic for the solid-state Tesla coil is shown in Fig. 2. The secondary of the Tesla coil, when directly driven by a solid-state driver, appears like a series RLC circuit. That's due to the self-capacitance of the coil with respect to ground. The capacitance is normally very small with the inductance being fairly large. At the resonant frequency, the inductive reactance cancels the capacitive reactance. The effective impedance is limited by such losses as the DC resistance of the coil, AC skin effect of...
the wire, eddy currents induced in nearby objects by the field of the coil, and so on.

Series RLC circuits have relatively low impedances when operated at the resonant frequency. The coil used in this project, when operated at its resonant frequency, looks like a 450-ohm resistive load to the solid-state driver. Series RLC circuits produce high voltages on the inductor and capacitor at the resonant frequency. The high voltage is due to a high current flowing through a high reactance (remember that the inductance is large and the capacitance is small, creating large reactances in each component at a given frequency). That is what produces the corona discharge at the end of the secondary coil.

The heart of the driver is IC1, the SG3524 pulse-width modulator. The duty cycle is fixed at about 45% for best efficiency. The frequency is controlled by the resistance on pin 6 and the capacitance on pin 7. With the values shown, the frequency has a range from 200 to 240 kHz. A flip-flop inside the chip divides that by 2 so that the effective output of the driver has a range from 100 to 120 kHz.

The outputs on pins 12 and 13 are 180 degrees out of phase with each other, and drive the gates of MOSFET's Q1 and Q2, which, in turn, drive the primary of transformer T1. Transformer T1 drives the bases of switching-transistors Q3 and Q4. The components in the base circuitry are used to increase the switching speed of the transistors. Transistors Q3 and Q4 switch the line voltage across the primary of T2, which increases the voltage and drives the end of the secondary coil directly. Note that the line voltage delivered to T2 is half-wave rectified by D1. That is important to the operation of the Tesla coil because a pulsating voltage is needed to produce the best effects.

When the device is plugged into a receptacle it will be in its standby mode. That is, the 21-volt power supply will be operational and the FET's will be driving the primary of T1. The standby mode produces enough power to "tune" the driver to the coil's resonant frequency before full power is applied. (Remember that the resonant frequency can be affected by nearby objects.) The current supplied to the secondary coil is indicated by LED1. Tuning is accomplished by adjusting the frequency via R1 and observing LED1. When resonance is achieved, the secondary coil will have a low impedance which will produce maximum current, lighting the LED. Diodes D3-D6 limit the forward and reverse voltages on LED1 when in the high-power mode.

When the device is switched into the operating mode (or the high-power mode), half-wave line-voltage pulses will be applied to the primary of T2. As the half-wave voltage increases, the current in the secondary coil increases and the energy stored in the inductance and capacitance of the secondary coil will increase. During this time there is no corona from the secondary coil (if the coil is constructed as shown in this article). Sometime before the half-wave line voltage reaches its peak, the corona will appear on the secondary coil, which will dissipate the stored energy very quickly. During the remainder of the half-wave line voltage, the coil will produce corona but the energy level will not be as great as the initial discharge. The coil will produce sixty individual corona discharges every second, although you'll see a continuous discharge.

**FIG. 1—THE SOLID-STATE TESLA COIL**

Can produce sparks as long as 8 inches. The output reaches a peak of about 100,000 volts.

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**Warning!** This article deals with and involves subject matter and the use of materials and substances that may be hazardous to health and life. Do not attempt to implement or use the information contained herein, unless you are experienced and skilled with respect to such subject matter, materials, and substances. Neither the publisher nor the author make any representation as to the completeness or accuracy of the information contained herein, and disclaim any liability for damages or injuries, whether caused by or arising from the lack of completeness, inaccuracies of the information, misrepresentations of the directions, misapplication of the information, or otherwise.

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Use this foil pattern, shown half-size, to etch your own PC board.
FIG. 2—SCHEMATIC FOR THE SOLID-STATE TESLA COIL. The secondary of the Tesla coil appears like a series RLC circuit due to the self-capacitance of the coil with respect to ground.

All resistors are 1/4-watt, 5%, unless otherwise indicated.
R1—1000 ohms, 10-turn potentiometer
R2—3900 ohms
R3, R4—2200 ohms, 1/2 watt
R5, R6—2200 ohms
R7—330 ohms, 1 watt
R8, R9—0.56 ohms, 2 watts, flameproof
R10, R11—3300 ohms

Capacitors
C1—0.001 μF, 50 volts, 5%, polyester
C2—110 μF, 50 volts, polyester
C3, C4—10 μF, 35 volts, tantalum
C5—330 μF, 35 volts, electrolytic
C6, C7—2 μF, 200 volts, nonpolar film-type
C8, C9—0.02 μF, 1000 volts, ceramic disc

Semiconductors
IC1—SG3524 pulse-width modulator
D1—MR751 diode
D2—D6—1N4934 diode
D7, D8—1N4936 diode
D9—not used
D10—D17—1N4004 diode
Q1, Q2—SK9155 power MOSFET
Q3, Q4—2N6678 or SK9140 NPN transistor
LED1—red LED. See text

Other components
F1—3-amp, 250-volt, fast-blow fuse
BR1—VM08 bridge rectifier, Varo
T1—hand-made transformer (the core is TDK # PC30EER25.5-Z and the bobbin is TDK # BEER-25.5-118CP)
T2—hand-made transformer (the core is TDK # PC30EC70-Z and the bobbin is TDK # BHC-70-5116)
T3—hand-made transformer (the core is TDK # PC30EER25.5-Z and the bobbin is TDK # BEER-25.5-118CP)
T4—115VAC/15VAC center-tapped transformer (Triad F-132P)
T5—115VAC/15VAC center-tapped transformer (Triad F-132P)
S1—SPST key switch

Miscellaneous: enclosure, aluminum angle bracket, high-voltage wire (to connect main unit to Tesla secondary), 30-gauge magnet wire for Tesla secondary and L1 and L2, 24-gauge magnet wire for L3 and L4, 18-gauge stranded hook-up wire for L5 and L6, 15-gauge magnet wire for T2 primary, 26-gauge hook-up wire for T2 secondary, 18-gauge magnet wire for both windings of T3, brass rod, discharge ball, hardware, AC linecord, etc.

Note: The following items are available from Corona Coil, PO Box 474, Riverton, UT 84065 (801-254-7653):
- Core, bobbin, and wire for T1, T2, and T3—$30.00
- Set of T1, T2, and T3 (assembled)—$50.00
- Tesla secondary coil—$35.00
- PC board—$15.00
- Aluminum angle bracket (heatsink and PC-board mount)—$5.00
Please add 10% S&H for all items.

A 124-page book by the author, Modern Tesla Coil Theory, is available for $19.95 plus $2.50 shipping from the Tesla Book Co., PO Box 12187, Tulla Vesta, CA 91912 (619-437-8515).
FIG. 3—PARTS-PLACEMENT DIAGRAM. It's best to play it safe and use the PC board for this project; we've provided the foil pattern if you would like to etch your own board.

Construction

Most of the construction is fairly simple if the printed circuit board is used. A parts-placement diagram is shown in Fig. 3, and we've provided the foil pattern if you would like to etch your own board. Figure 4 shows the completed prototype board housed in its aluminum enclosure.

The most difficult item to construct will be the Tesla secondary coil, followed by T1 and T2. The secondary coil may take an hour or so to make if you prepare ahead of time. Preparation includes making some device that will easily rotate the coil form while winding the wire. The author used a small lathe and it took about 15 minutes of actual winding time and 30 minutes to get set up.

Do not deviate at all from the following parameters of the secondary coil! Any deviation will change the characteristics of the coil and it may not operate with the driver unless modifications in the driver are made. Any change in physical dimensions or wire size will alter the resonant frequency and effective impedance of the coil. Any change to the discharge electrode will effect the maximum energy obtainable.

The coil form for the secondary winding is a standard 5-gallon plastic container 10 inches in diameter at the bottom, 12 inches in diameter at the top, and 14 inches long. The bottom of the container becomes the top of the coil. To make winding easier you should drill a hole about an inch in diameter through the center of the bottom of the container. A similar hole should be drilled through a removable lid and then the complete coil form can be rotated easily on a dowel. Start the secondary winding 1 inch from the small-diameter end and close-wind 30-gauge magnet wire for a total length of 10 inches. It does not matter what direction the wire is wound in.

When winding the original coil for this article, shellac was used to lubricate the wire as it was wound and also to act as a sealant afterwards. It was difficult to wind the coil because the coil form was very slick and had a slight taper to it, and, as a result, the wire kept slipping. It may be easier to spray the container with adhesive before winding the wire to make it stay in place. A couple of coats of shellac should be applied to the finished winding. You also must put 3 or 4 beads of silicone sealant around the end of the winding at the top of the coil to keep corona discharges away from the area. If corona discharges appear along the coil at the top it will limit the maximum energy and destroy the coil form.

The discharge ball, or electrode, is a brass-plated metal doorknob, 1-inch in diameter, that can be found in hardware stores (see Fig. 5). The ball is mounted on a 4-inch brass rod; you can drill and tap the ends of the brass rod with a 6-32 tap (or whatever matches the threading on the doorknob) to make mounting easier. The brass rod is connected to the coil form by two pieces of plastic, one on each side of the coil form, over the ½-inch.

FIG. 4—HERE'S THE AUTHOR'S COMPLETED PROTOTYPE housed in its aluminum enclosure. It's important that the case be properly grounded.
hole. A 6-32 screw passes through the pieces of plastic and into the brass rod to hold the assembly together. The wire is soldered to a lug held in place by the 6-32 screw.

A banana jack is used to make connections at the bottom of the coil. Locate the jack about ¾-inch from the edge of the wire on the coil. Silicone should be used to insulate the connections between the magnet wire and the brass rod and banana jack. The finished coil, when built exactly as we’ve shown, will have a resonant frequency of about 110 kHz.

Transformer T1 is made with a ferrite core and bobbin from TDK (see the parts list). Coils L1 and L2 are wound first with 30-gauge magnet wire, 16 turns each, making one layer on the bobbin. The two windings are bifilar wound, as shown in Fig. 6-a: L1 starts on pin 3 and L2 starts on pin 4. Wind both in a counterclockwise direction while looking at the top of the bobbin. Terminate L1 on pin 1 and terminate L2 on pin 2. Put a layer of cellophane tape on top of the winding to insulate it from L3 and L4.

Coils L3 and L4 are made with 5 turns each of 24-gauge magnet wire and are also bifilar wound, on top of L1 and L2, and in the same direction. Coil L3 starts on pin 6 and L4 starts on pin 8. Terminate L3 on pin 5 and terminate L4 on pin 7. This completes the transformer until it is mounted on the PC board.

Put the two core halves through the bobbin and put tape around them to hold them in place. As shown in Fig. 6, L5 and L6 are wound after the transformer is mounted on the board: L5 and L6 are wound with 18-gauge stranded hook-up wire with one turn each. Solder the collector (Q4) end of L6 to the PC board. Go one turn in a counterclockwise direction around the core of T1 and then terminate the other end of L6 at the primary of T2. Solder the collector (Q3) end of L5 to the PC board and go in a clockwise direction around the core of T1 for one turn, terminating the winding at the cathode of D1.

Transformer T2 is also made from a ferrite core and bobbin from TDK (again, see the parts list). The primary is 10 turns of 15-gauge magnet wire, although a smaller gauge, say 18, can probably be used. It does not matter what direction the wire is wound in but the turns should be equally spaced across one layer of the bobbin. Put several layers of cellophane tape on top of the primary to insulate it from the secondary and to provide a smooth surface on which to wind the secondary. The secondary is made with 280 turns (the exact number is not critical) of 26-gauge hook-up wire. The direction is unimportant. You can use magnet wire if you desire but you should put cellophane tape between each layer. The low-voltage end of the secondary is the one that is the closest to the primary winding. When the windings are complete, put the core halves through the bobbin and hold them in place with tape wrapped around them.

Transformer T3 is made with the same core and bobbin as T1. Both windings are bifilar with 18-gauge magnet wire for as many turns as possible. The start of both windings are polarized as indicated by a dot in the schematic diagram (Fig. 2). The pins on the bobbin are not used and should therefore be cut off, and the 18-gauge wires are then soldered directly to the PC board as indicated.

An aluminum angle bracket is used when mounting switching transistors Q3 and Q4. The bracket provides the physical support between the PC board and enclosure and also provides good heat sinking for the transistors. The transistors should
be insulated from the aluminum; insulating hardware is normally included when you purchase the transistors. Use the PC board as a template for drilling holes for the transistors in the aluminum bracket. The angle bracket is mounted to the enclosure by drilling holes and taping them with a 6-32 tap. Thermal conductive compound is used between the transistors and angle bracket and between the angle bracket and the enclosure.

A banana jack is mounted in the back of the enclosure to make connections between the Tesla secondary coil and the high-voltage ferrite transformer. The output voltage from the ferrite transformer may reach 5000 volts peak with no load so it is wise to use extra insulation for the banana jack. Mount a piece of plastic, 1½-inch square, to the back of the enclosure over a 1-inch square hole, and mount the banana jack in the center of the plastic. That will space the banana jack at least ½-inch from the metal enclosure.

The prototype used a 10-turn potentiometer for R1 to make frequency adjustments easier and this allowed the use of a 10-turn dial to mark the frequency settings for different purposes. You can use a regular potentiometer but the 10-turn unit is superior.

An enclosure was fabricated out of ½-inch aluminum with a plexiglass top, but any metal enclosure would be suitable. Just be absolutely sure that you ground the metal enclosure.

**Operation**

**Warning:** The power output from the Tesla coil is dangerous! Make sure no one comes in contact with the output voltage directly from the driver. Make sure nobody tampers with the unit, and keep it out of reach of children. Make sure you use a key.

**FIG. 6—TRANSFORMER Ti IS MADE** by winding coils L1 and L2 first (a). After putting a layer of cellophane tape on top of the first windings, coils L3 and L4 are wound on top of L1 and L2. Coils L5 and L6 are wound after the transformer is mounted on the board (b). See text for detailed instructions.

**FIG. 7—SEEN HERE IS THE DISCHARGE** from the ball electrode into the air.

**FIG. 8—THE SPARKS WILL JUMP** even farther if a grounded electrode is placed near the discharge ball.
Connect the Tesla secondary coil to the driver with a 3-foot insulated wire (it is a good idea to keep at least 3 feet from the secondary coil). You should always unplug the driver when you are making connections between the driver and secondary coil to be absolutely safe. The wire connecting the coil and driver carries a dangerous amount of power so be certain the wire is well insulated. In a dimly lit room you should be able to adjust the tune control to set the driver at the coil's resonant frequency. Observe the LED and watch for one place in the tuning control's adjustment where the LED glows brighter than anywhere else. You should be able to adjust the LED and watch for one place in the tuning control's adjustment where the LED glows brighter than anywhere else. Never apply full power to the driver unless you can obtain resonance first. Damage to the driver will most likely occur if resonance is not maintained.

Once you obtain resonance you can switch to the full-power mode: the LED will glow very brightly. With no objects around the coil you should observe a snappy brush discharge 5 to 6 inches in length emanating from the discharge electrode (see Fig. 7). It might be somewhat louder than you would expect. Very slight adjustments in the tune control may improve the discharge. You should be able to get 7-inch streamers with a grounded electrode above the coil (see Fig. 8). Be aware that any change of the physical surroundings around the coil will change its resonant frequency and the tune control will need to be adjusted to maintain resonance. When operating the Tesla coil, be aware of the temperature of the enclosure where the aluminum angle bracket is mounted. Shut off the power if the area gets too warm. The prototype was operated for 2 full minutes, and you could just start to feel some warmth on the enclosure. However, you should operate the Tesla coil only for short periods of time.

Once you have a working unit you can start to experiment with different things. Try removing the discharge ball and use a point instead. Try changing the distance of the ball electrode from the coil. Try holding an incandescent lamp a short distance from the coil—but be very careful. Different lamps will produce different discharges.
STORAGE AND RETRIEVAL OF DIGITAL information has sparked an ex-
citing revolution in computers and consumer electronics. You 
find semiconductor memories in 
nearly all “intelligent” electronic 
systems, including car radios, 
television, VCR’s, portable disc 
players, and computers. Without 
the on-going advances in mem-
ory technology, the high-tech revo-
lution would rapidly grind to a 
halt.

In this article, we will examine 
several important concepts be-
hind semiconductor memory de-
vices, including basic tech-
nologies, memory organization 
and configuration, design con-
siderations, and applications.

Memory types

Semiconductor memory de-
vices can be classified in one of 
two ways: permanent or tempo-
rary. Although basic operating 
principles of both are similar, 
each plays a different role, and 
each has unique advantages and 
advantages. We will discuss 
both types in detail.

As the name suggests, informa-
tion in permanent memory is 
retained at all times, even after 
removal of system power. Perma-
nent memory is also called non-
volatile and read-only memory. 

Permanent memory is most often 
used to store fixed program in-
structions or numerical con-
stants that do not change during 
the life of a product. For example, 
personal computers use perma-
nent memory to hold the basic 
input/output system (BIOS) that 
initializes the computer and pro-
vides it with a core of low-level 
functions. There are four basic 
types of permanent memory: 
ROM, PROM, EPROM, and 
EEPROM. Let’s discuss each 
type.

ROM

The read only memory (ROM) is 
the oldest and most straightfor-
w ard type of permanent semi-
conductor memory. The information 
that’s programmed into a ROM is 
specified by the buyer, but the 
ROM itself must be built by the 
manufacturer.

A ROM is relatively inflexible. 
After it’s been programmed, it can 
never be altered. If the informa-
tion in a ROM must change, a 
whole new device must be manu-
factured and substituted for the 
old ROM, and that is an expen-
sive, time-consuming process. 

Hence the ROM is economically 
feasible only when used in great 
volumes for thoroughly debug-
ged applications. 

One advantage of the ROM is 
its ruggedness. Since the pro-
gram is an actual physical part of 
the device itself, it can withstand 
relatively large amounts of elec-
trical and physical abuse, yet still 
maintain its contents. The auto-
mobile industry uses ROMs ex-
tensively in on-board computers.

PROM

The programmable read only 
memory (PROM) offers a tremen-
dous advantage over the ROM in 
that it can be programmed by the 
end user, who is then less depen-
dent on manufacturers’ lead 
times. A PROM can be “burned,” 
or programmed, only once be-
cause it cannot be erased.

The term “burn” comes from the 
method used to program a 
PROM. A factory-fresh PROM 
consists of a matrix of fuse-
ble links. An intact link produces a 
binary 0 at the selected location; 
a burned (open-circuit) link pro-
duces a binary 1, as shown in 
Fig. 1. (We’ll discuss how to get at 
a particular location in a PROM 
later in this article.)

To burn a PROM, a special 
piece of equipment called a 
PROM burner generates high-en-
ergy pulses which destroy the de-
sired links to match the contents 
of a user data file.

PROM’s are slightly more ex-
pensive than ROM’s on a per-unit 
basis, but their flexibility often 
justifies higher cost. Many 
PROM’s are available through re-
tail electronics outlets.
FIG. 1—A PROM BEFORE PROGRAMMING consists of a matrix of fused links joining each row-column intersection. Programming blows desired links.

**EPROM**

The erasable programmable read-only memory (EPROM) overcomes one of the main disadvantages of the PROM: its inability to be reused. After a link has been burned, it can never be restored. By contrast, typical EPROM's can be reliably burned and erased thousands of times.

The PROM is built around traditional bipolar transistor technology, which uses both a great deal of power and occupies a lot of space. The EPROM, on the other hand, uses newer metal-oxide semiconductor (MOS) technology, which requires little current and occupies little space. In an EPROM, information is stored as small packets of charge buried deep within the substrate of the IC, as shown in Fig. 2.

An EPROM is programmed much like a PROM. A special EPROM programmer selects an address in the device, places the desired binary information on the data lines, and then pulses the EPROM's PROGRAM pin. That pulse is what locks the bit pattern into the substrate of the chip.

To erase an EPROM, it's necessary to remove the charges in the IC's substrate. That's accomplished by exposing the circuit (the die itself) to short-wave-length ultraviolet (UV) light for a prescribed period of time. The excitation created by the UV light allows stored charge to dissipate, so the IC gradually returns to its pre-programmed state. The UV light is introduced into the EPROM through a transparent quartz window in the top of the IC package.

Use caution when working with EPROM's. Even though it takes about 20 minutes of exposure to a concentrated UV light source to erase an EPROM, some common sources of light, such as sunlight, fluorescent light and "black-light", may contain enough UV to trigger random charge dissipation and introduce errors in the device. So be sure to cover the quartz window with a piece of opaque material.

EPROM's cost more than PROM's, but cost-per-bit is actually lower because MOS technology allows the designer to squeeze several times more information in the same amount of space. One disadvantage of the EPROM is that it must be physically removed from the system to be erased and re-programmed.

**EEPROM's**

The electrically erasable programmable read only memory (EEPROM) is similar to the EPROM, but overcomes its main disadvantage: the inability to program it in-circuit. That feature offers exciting possibilities in applications where software must adapt to changes in the operating environment.

The EEPROM is no panacea, however. It's slower than other types of memory, and it requires a relatively long time to update the altered data. As a result, EEPROM's are best suited for holding information that changes infrequently. Information that changes often is best left to the work of temporary memory, the other broad class of semiconductor memory.

**Temporary memory**

Information held in a temporary semiconductor memory device can be altered and updated frequently, but will be maintained only as long as power is supplied to the device. If power fails, memory contents will be lost. That type of memory is usually referred to as volatile memory. It is also known as random access memory (RAM). The name refers to the fact that any location may be accessed as quickly as any other. By contrast, in a sequential device like a tape drive, access speed depends on the location of the desired information. However, random locations in ROM's, PROM's, EPROM's, and EEPROM's can be accessed with equal speed. Nonetheless, when people speak of RAM, they almost invariably are referring to temporary memory.
Static RAM

Static RAM (SRAM) is the oldest and most straightforward form of temporary semiconductor memory. A typical SRAM consists of several flip-flops, or cells, as shown in Fig. 3. Each cell stores one bit of information; multiple cells are arranged in a two-dimensional array. To access a particular cell, row and column addresses must be set up, and then several control signals must be pulsed.

Since data is always available from the flip-flop matrix, the SRAM tends to be a fast device. Its primary disadvantage is limited capacity. Each flip-flop occupies a relatively large area on the IC, so the maximum number of cells is limited.

Dynamic RAM

Dynamic RAM (DRAM) uses an entirely different technology to accomplish data storage. The key difference lies in the design of the cell itself. As shown in Fig. 4, each cell in a DRAM stores information as a packet of charge across a MOS transistor, similar in principle to the EPROM works, but unlike the SRAM, which uses a flip-flop to hold one bit of data.

To allow frequent updates, each cell must be capable of changing state almost instantly. To allow rapid change, the storage capacitance must be extremely low; so low in fact that it cannot sustain its charge for more than a few milliseconds. Therefore each DRAM location must be refreshed about every two milliseconds. If a cell is not refreshed, it will simply lose its data. However, refresh cannot happen by itself; external circuitry is required, as well as additional circuitry within the DRAM itself. Fig. 5 shows a block diagram of the internal structure of a DRAM. The added complexity and cost of refresh circuitry is the main disadvantage of DRAM.

On the other hand, DRAM offers several distinct advantages over SRAM. Storage capacity is much greater. Common DRAM's provide one megabit ($2^{20}$) of storage, and four-megabit IC's are just over the horizon. In addition, 16-megabit memories are being developed, and 64-megabit DRAM's are on the drawing board.

Power is another consideration. DRAM's require less current to operate; there are far fewer components per cell to dissipate power. The power savings can be substantial in applications that need a great deal of memory. DRAM's also have a standby mode that essentially disables all functions except refresh. In standby mode, a DRAM requires just a few milliwatts of power to maintain its information. In some cases, the low power requirement makes battery backup practical. SRAM's also have a standby mode, but they typically need more than 100 milliwatts of power. Now let's examine some of the advantages and disadvantages of each type.
the technologies used to fabricate semiconductor memory devices.

Fabrication technologies

Every semiconductor memory chip houses sophisticated, sensitive microcircuitry. Each minute component must be integrated deep into the substrate of the chip (or die), which itself rests within a hermetically sealed case of plastic or ceramic. The process of circuit integration involves a complex combination of optical and chemical processes to form a working IC. Memory devices manufactured today are typically made using either bipolar or MOS fabrication technologies. In addition, a new hybrid of the two technologies, called Bi-MOS, has begun to appear. Although the actual manufacturing processes of these kinds of devices are too involved to cover here, we can review the characteristics and uses of those technologies.

Bipolar technology

The bipolar transistor (with emitter, base, and collector) was the first component successfully integrated into a semiconductor wafer in the form of the TTL IC. Many simple logic functions could thus be synthesized easily and efficiently. The resulting low cost and high availability made TTL a mainstay of digital logic design through the 60's and early 70's. Even to this day, TTL remains a cornerstone of basic logic design. When memories were the obvious choice.

Although there are several SRAM chips in the TTL family (notably the 74S200 and 74S201), TTL suffers from several major drawbacks that severely restrict the capacity of bipolar SRAM. First, bipolar logic requires a relatively large area on the chip for each logic gate. Many gates are needed to build a SRAM, so space is depleted rapidly. In addition, bipolar logic requires significant operating current per gate. Since current ultimately translates into heat, the number of cells is limited even further. Size and power restraints usually limit the number of bipolar memory cells to fewer than 1000 bits.

MOS technology

The development of MOS technology is largely responsible for the incredible advances in high-tech electronics since the late 1970's. The materials and chemicals used in MOS fabrication are different from those used for bipolar fabrication, but the process is fundamentally the same. The most familiar MOS family is complementary MOS (CMOS), but there are many variations, including PMOS, NMOS, VMOS, DMOS, and HMOS.

CMOS, NMOS, and HMOS devices are the most widespread variations of MOS technology in use today. CMOS has been used extensively in memories, and to produce a family of devices that is functionally similar to the TTL family. CMOS dissipates far less power than TTL and can run on a much wider range of supply voltages (3–15 volts DC). N-channel MOS (NMOS) technology is used to produce memories that are fast, dissipate little power, and can fit many components on a chip. Although early devices required several supply voltages, modern NMOS ICs operate from a single 5-volt supply. High-performance MOS (HMOS) is an NMOS variation that's used in modern high-speed low-power microprocessors.

In spite of their obvious advantages, all MOS devices suffer from one key weakness: they're extremely sensitive to static electricity. There are important precautions that should be taken. Be sure to follow manufacturers' guidelines for handling MOS devices.

Memory operations

To the external world, the organization of a semiconductor memory device appears as a sequence of locations. Each location may have 1, 4, 8, or some other number of bits, but regardless of the number of bits per location, each location has a unique address. The number of unique addresses depends on the number of address lines. If there are 8 address lines, then there are 28 or 256 addresses. Although externally a semiconductor device appears to have a sequential organization, internally the cells are arranged in a square.

The relationship between the number of physical cells (bits) and the number of logical locations (addresses) depends on the number of bits per address. For example, a memory IC could have 1 megabit of cells arranged as 1 x 1 megabit, as 4 x 256K, or even as 8 x 128K. Internal decoding circuitry varies depending on how the organization is to appear externally.

For example, Fig. 6 represents a simple ROM. The format of the ROM is 256 addresses with four bits per address. The memory array is a 32 x 32 square, giving 256 addresses. And for 256 addresses the chip requires eight address lines (28 = 256) to identify each location uniquely. The lower five address lines (A0–A4) select one of 32 possible rows (25 = 32). The upper three (A5–A7) select one of eight columns (23 = 8). There are four 1-of-8 decoders, so four columns (one from each group of eight) will be active for each selection.

After a valid address is presented to the address lines, the data bits at the intersections of the selected row and columns will be sent through the respective 1-of-8 decoders to several three-state buffers. If the READ ENABLE signal is brought low, the data present at the buffers will be delivered to the ROM's output. But when READ ENABLE is high, the high impedance of the three-state buffer will simply disconnect the ROM's outputs from the circuit.

SRAMs, along with PROMs, EPROMs, and EEPROMs, are more sophisticated. Figure 7 shows a simple SRAM organized...
as $4096 \times 1$. Addressing is similar to the ROM in the previous example but, in this case, there are 12 address lines that provide $2^{12}$ or 4096 (4K) addresses. One bit of data is available at each address location.

A read/write control signal determines whether data will be read from or written to the IC. If R/W is logic 1, data will be read from the cell. If R/W is logic 0, data will be written to the cell.

To read a bit of data, a valid address must be supplied, R/W must be high, and the CHIP SELECT input must be low. To write a bit of data, the same conditions apply except that R/W must be low. The timing relationships between the signals at various pins can be critical, depending on the circuit.

### Timing considerations

Today's generation of RAM IC's has been designed to operate at high speeds, so timing characteristics for address, data, and control lines are important. There are several important parameters that we will discuss.

**Access time** specifies how long it takes after addressing a specific location before valid data appears at the IC's output. A slow memory device may have an access time of as much as 450 nanoseconds, while a fast device might access data in as little as 25 nanoseconds. Common memory devices today have access times of about 100-150 ns. As a rule of thumb, the faster a memory device is, the more expensive it will be.

**Settle time** specifies the amount of time that must pass after setting up the address, data, and CHIP SELECT signals, before the R/W may be pulsed low to write data into the IC.

In addition, the write pulse must be held low for a minimum amount of time to ensure that the data is accepted into memory. That is the duration of the write pulse. The address, data, and enable signals must be held steady for a minimum time after the write pulse; that period is called the hold time.

Those timing parameters apply to SRAM's; DRAM's have even more intricate timing requirements. Although the basic principles of reading and writing are similar to those for the SRAM, there are some extra features and parameters that must also be considered.

The first involves memory addressing. As discussed earlier, DRAM's can provide millions of bits on one device. For example, addressing 1 megabit ($2^{20}$) would require 20 address lines. It's possible to build an IC with 20 or more pins, but to save space and reduce pin count, several address lines are multiplexed on a single pin.

Figure 8 shows the block diagram of a 1-megabit x 1-bit DRAM. Note that only ten address lines enter the IC, so you might think that you could access only $2^{10}$ (1024) locations. In fact the 20-bit address is broken up into two parts, each of which is supplied separately. The lower ten bits select the desired row in the memory array, and the upper ten...
bits select the desired column. The row-address lines are strobed into the IC by pulsing the row address strobe (RAS) input, and the column-address lines by pulsing the column address strobe (CAS) input. External circuitry must ensure that the proper set of address lines is applied to the IC before pulsing a strobe input.

After the IC receives the full address, CHIP SELECT and R/W may be set up, as with an SRAM, to read or write data. The access, setup, and hold times apply to DRAM's as well.

Refresh
As mentioned earlier, DRAM's require periodic refreshing, otherwise their stored charge will dissipate. There are several ways of refreshing a DRAM system, all of which use the RAS and CAS inputs. The simplest method is called RAS-only refresh. It involves holding CAS high, which in turn holds the output in a high-impedance, or disconnected, state. The refresh circuitry then selects each row in turn, pulsing RAS low for each row as it is addressed. It does not matter whether all rows are refreshed in one sustained burst, or one row between, for example, read or write operations. As long as a cell is refreshed in time, its data will remain intact.

Hidden refresh is a variation on RAS-only refresh in which CAS is held at logic 0 (for example, valid data is maintained on the output) while rows are selected and refreshed. Depending on system timing, CAS may be held low for several microseconds, during which several rows may be refreshed.

There are other variations, but all refresh circuits add a fair amount of complexity to a circuit. Fortunately, however, there are refresh-controller IC's for many different DRAM sizes and configurations. Those IC's reduce cost, increase reliability, and decrease required PCB board space.

EPROM emulator
You can easily assemble your own hand-made “EPROM” using two common TTL IC's and several Germanium diodes. Figure 9 shows the schematic for a 16 x 4 memory circuit. It’s loosely called an EPROM because it can be reprogrammed at any time by rearranging the diodes in the matrix. Although the circuit is unsuitable for high-performance or microprocessor-based applications, it can be used to supply pre-programmed bit patterns to discrete logic circuits. It also provides an excellent demonstration of basic memory operation.

There are eight rows and eight columns, yielding 64 bits of memory. Two demultiplexers allow access to a particular memory cell. One demultiplexer decodes the row and one decodes the columns. A 74138 1-of-8 decoder selects the row, and a 74157 quad two-input multiplexer selects the columns. Address lines A1-A3 drive the 74138 to select which one of eight rows will be pulled to ground. The columns are arranged in pairs; address line A0 determines which member of a pair is connected to the output.

The IN270 diodes determine the bit pattern in the circuit. Germanium diodes are used because of their low forward voltage drop (0.3 volts); silicon diodes have a higher voltage drop and will not work with TTL IC's.

Every column is pulled high via a pull-up resistor. If a diode is absent when a particular row is selected, the column will provide a 5-volt output. However, if a diode is in place, it will be forward biased via the pull-up resistor, through the 74138, and then to ground. The corresponding output thus becomes a logical 0.

For example, if address 0000 is selected, 74138 output yo (row 0) is connected to ground, and all 74157 inputs are connected to the B position. Because there are diodes connected to each the B inputs in row 0, the output would be 0000. If the address was 0001, row 0 remains selected, but the 74157 inputs are switched to the A position. The A cells have no diodes, so all outputs would be high (1111).

Parallel memory
Semiconductor memories (both temporary and permanent) can be placed in parallel to increase the number of data bits available per address, as shown in Fig. 10. The circuit is built from several 2147 SRAM's (4096 x 1). By connecting the address and control lines in parallel, the same address in all IC's will be selected simultaneously. The data bits, of course, are kept separate. You could just as easily place 8, 16, or 32 IC's in parallel to create 4K x 8, 4K x 16, or 4K x 32 memory blocks.

Conclusion
Memory is an integral part of the high-tech revolution. Even the most basic processing circuit would be useless without some sort of memory to store variable data.

As you can see from our comparison of the many different permanent memory devices, there are distinct advantages and limitations to each type. What you choose depends on your individual needs—the ROM is inflexible but rugged, while the PROM can be programmed by the user, but only once because it can't be erased. The EPROM can be programmed and erased over and over again but uses a lot of power and space, while the EEPROM can be programmed while in circuit, but is slow.
**ELECTRONIC FUSE**

**How do you troubleshoot power-related problems without blowing fuse after fuse? Just use our electronic fuse!**

T.L. PETRUEZELLIS

The electronic fuse is a sensitive fast-acting adjustable circuit breaker that will quickly become one of your most useful bench-top accessories. If you have been stumped by a faulty electronic circuit and consumed a number of costly or hard-to-locate fuses, you will appreciate this inexpensive circuit breaker. All you have to do is connect the electronic fuse to the device under repair, and then adjust the current threshold control to the value you need anywhere from 1/20 to 10 amperes.

Additional applications for the electronic fuse include charging circuits for marine/mobile/aircraft systems, as well as new circuit designs. The electronic circuit breaker could be used after the design of a new circuit to help choose the correct value fuse. The electronic circuit breaker is connected in place of the original fuse of the device under repair or test. If the breaker “trips,” a red LED will light and power is cut off. When you're ready to continue, simply press the reset button.

**Circuit description**

As shown in Fig. 1, two test leads are connected in series with the normally closed relay contacts of RY1, a 12-amp fuse (F1), and the two-turn primary of T1, a torroid transformer. The secondary of T1 is wound underneath the primary on the half-inch torroid. The secondary coil is 100 turns of 30-gauge magnet wire with a total resistance of 8 to 10 ohms. The secondary is connected to a high-low range switch (S1). The switch connects to a resistor network to provide stability and ease of operation. The low range permits values from 1/20 to 6 amperes, and the high range includes values from 1 to 10 amps, with overlapping between ranges. Capacitors C1 and C2 form a high-frequency filter to help reduce spikes and line noise.

Op-amp IC1-a amplifies and rectifies the AC input and applies it to IC2-a, an LM339 comparator, which is used to adjust the threshold, or current, via potentiometer R4. A clamp is formed by D3 which holds the input of IC2-b to a constant level. A filtered DC output is amplified by IC2-b and fed to Q1, a 2N3904 transistor. The transistor changes the output of IC2-b to the proper level and polarity in order to trigger SCR1. When the input current exceeds the threshold set by R4, the SCR will turn on. The relay will now open and LED1 will indicate that the circuit has been “tripped.” The LED will remain on and the power to the device under test will remain off until the reset button (S3) is pressed.

Current consumption for the electronic fuse is about 10–15 mA at idle and about 100 mA when
PARTS LIST

All resistors are 1/4-watt, 5%, unless otherwise noted.
R1-107,200 ohms, 1%
R2-442,000 ohms, 1%
R3-387,000 ohms, 1%
R4-165,000 ohms, 1%
R5, R6-300,000 ohms
R7-50,000 ohms, audio-taper potentiometer
R8-1500 ohms
R9-12,000 ohms
R10-18,000 ohms
R11-13,000 ohms
R12-4700 ohms
R13-2000 ohms
R14, R15-1000 ohms

Capacitors
C1-200 pF, 50 volts, ceramic
C2-100 pF, 50 volts, ceramic
C3, C4-1 µF, 50 volts, electrolytic
C5-100 µF, 50 volts, electrolytic

Semiconductors
IC1-LM358 low-power dual op-amp
IC2-LM339 quad comparator
D1-D3-1N914 diode
D4-1N4004 diode
LED1-red light-emitting diode
SCR1-NTE 5404 silicon-controlled rectifier
Q1-2N3904 NPN transistor

Other components
S1-DPDT toggle switch
S2-SPST toggle switch
S3-normally closed pushbutton switch
F1-12-amp fast-blow fuse
RY1-DPDT relay, 12-volt coil, 12-amp contacts (or use two sets of contacts in parallel, see text)

Miscellaneous: PC board, project case, fuse holder, alligator clips, 30-gauge magnet wire, 24-gauge stranded wire, 16-gauge stranded wire, PC-board scrap for wire spool, hardware, solder, etc.

Note: The following items are available from T.L. Petruzellis, 340 Torrance Avenue, Vestal, NY 13850:
- PC board only—$8.25
- Kit of parts including the torroid core and wire (you have to wind it yourself), IC's, and project case (does not include a power supply)—$44.95

Specify wires with alligator clips or 3-prong female power outlet (see text). Add $3.00 S&H. NY residents must add 7% sales tax. Please allow 4-6 weeks for delivery.

FIG. 1—THE ELECTRONIC FUSE is almost like an adjustable circuit breaker, where you can adjust the trip point anywhere from 0.1 to 12 amps.

FIG. 2—PARTS PLACEMENT DIAGRAM. Because various controls are mounted directly on the PC board, you may have to drill tiny pilot holes on the circuit board in the center of each control location, place the unpopulated circuit board directly on top of the case, and then transfer the holes before installing the components on the board.

FIG. 3—THIS WIRE SPOOL allows easy winding of the torroid transformer (see text).
constructed from a 0.5-inch powdered-iron torroid. A wire spool was made from a scrap of PC-board material, about 1 1/4-inches long by 1/4-inch wide with V-shaped notches cut at both ends (see Fig. 3), and 30-gauge magnet wire was wound on the spool between the two notches. The spool was then pushed in and around the core of the torroid (like a sewing needle) forming a 100-turn coil (T1's secondary) all the way around the entire torroid core (you unspool the wire as you make the turns). The ends of the 30-gauge magnet wire were stripped and carefully soldered to 24-gauge wires. Five-minute epoxy was then brushed over the secondary coil. After the glue dried, the two splices were glued to the edge of the torroid with another spot of epoxy to reduce the stress on the 30-gauge wires.

The primary coil was wound over the secondary using two turns of 16-gauge wire with insulation heavy enough for about 12 amps. Heavy linecord can be used for the primary if you like. The torroid was placed over the square notch on the end of the PC board (as shown in Fig. 2), and attached to the board with a plastic strip placed over the torroid and fastened with two screws. One of the 16-gauge wires was connected in series with the 12-amp fuse; the other end of the fuse was connected to an alligator clip. The other 16-gauge wire was connected to one end of RY1's normally closed contact. The remaining relay contact was connected to another alligator clip. Note that the relay used in the prototype is a double-pole unit with the contacts wired in parallel to handle higher current. Figure 4 shows the prototype.

A later version of the electronic fuse replaced the alligator clips with a chassis-mounted female power receptacle. The device under test is plugged into the outlet on the electronic fuse and a 1200-watt heating element coil, but an electric fry pan or toaster could be used instead. The thermostat in a fry pan must be turned up to maximum or disabled. The heater is connected to the output of a variac and the input of the variac is connected in series with an ammeter and the electronic fuse (see Fig. 5). The variac output is slowly stepped up in small increments. A calibration sheet is placed under R4's adjust knob.

Calibration must be done for both the high and low ranges. Begin by selecting the low range, and turn R4 clockwise to about midway. Next turn on the variac and rotate R4 to the trip point. Place a pencil mark on the calibration sheet, back down the variac, and reset S3. Bring up the variac to the point you just marked for one amp, and watch the meter to ensure that you are drawing one amp as the breaker "trips." Now proceed with the next value, adjust R4 past midway, set the variac for two amps, and rotate R4 down to the trip point. Repeat the procedure for each fuse value in the low and high ranges.

Operation

Operation of the electronic fuse is quite simple. The alligator clips connect to the fuse holder of the device under test, essentially substituting the electronic fuse for the fuse that was in the original circuit. First choose the high- or low-sensitivity position of S1: the low range covers 1/4 to 6 amps and the high range covers 1 to 10 amps with overlap between the two ranges. Next adjust R7 for the current setting that best represents the desired fuse value. Turn on the power switch S2 and reset the electronic fuse by pressing S3. Now turn on the device being tested; if LED1 lights, the "fuse is blown" and you must reset the circuit by pressing S3. Continue to troubleshoot until the repair is completed.

Calibration of the Electronic Fuse was performed by using a 1200-watt heating element coil, but an electric fry pan or toaster could be used instead. The thermostat in a fry pan must be turned up to maximum or disabled. The heater is connected to the output of a variac and the input of the variac is connected in series with an ammeter and the electronic fuse (see Fig. 5). The variac output is slowly stepped up in small increments. A calibration sheet is placed under R4's adjust knob.
Build this inexpensive color-bar test
generator and brush up on your video skills.

THOMAS GOULD WB6P

IF YOU'RE INVOLVED IN TV SERVICING and repair, or just enjoy tinkering around with video or amateur television, you'll be interested in this color-bar test generator. This convenient device produces an NTSC color-bar pattern that can be used for video performance testing and monitor adjustments. For added flexibility, just the encoder section can be used to generate composite video from your computer's RGB and sync outputs. With a dedicated color-bar generator, you can eliminate the need for a test tape or your camera—all for under $70!

Before we delve into the theory behind the color-bar generator, let's briefly discuss the various components that make up the composite NTSC video signal: synchronization, luminance, and chrominance information.

The NTSC signal
A typical NTSC composite color video signal is shown in Fig. 1-a. (NTSC is the National Television Systems Committee, who has set the standards for color encoding and decoding systems in the U.S. since 1953.) The picture on a color TV is formed by three electron beams of varying amplitudes and phases: red, blue, and green. Each of those beams are scanned horizontally and vertically over the screen. As the beams scan, their currents and amplitudes change to create the light and dark areas on the picture-tube face and form the image that you see displayed on the screen.

The composite video signal is made up of three basic components: the scan control information called the synchronizing pulses (Fig. 1-b), the luminance signal, which is the brightness information and is often referred to as the Y signal (Fig. 1-c), and the color information called the chrominance signal (Fig. 1-d). Let's briefly discuss each type of video information.
Synchronizing components

In order for a picture to be reproduced properly, the TV receiver must scan its screen exactly in step with the camera in the studio. To make sure the camera and the receiver are synchronized, a series of pulses are sent to the camera telling it when to start at the top of the screen and when to begin a new line at the left of the screen. Those same pulses are sent to the receiver along with the video information. The signals that tell the camera and receiver when to start at the top are called vertical sync pulses, while those that start scanning each line at the left are known as horizontal sync pulses.

In the NTSC system, each frame of complete video image contains 525 lines. That is accomplished by horizontally scanning at approximately 15,750 lines per second, and vertically scanning at 30 frames per second. (The vertical scan rate is actually 60 Hz, but it takes two trips, or fields, down the screen to complete one frame.) The process of returning to start a new scan is called retrace or flyback.

Luminance

Black and white information is contained in the Y or luminance signal, which determines the instantaneous brightness of the electron beams as they scan over the screen. In fact, it is all that is used for the single electron beam in a black-and-white TV set. A negative-going video detector detects a luminance signal in which the negative signal extremes correspond to bright areas of the picture. The waveform shown in Fig. 1-c would, therefore, produce vertical bars of decreasing brightness from left to right. Note that the output is black during retrace so the electron beams will not be seen.

In the NTSC color system, the Y signal is made from the red, green, and blue cameras by an additive technique: 30% of the red signal, 59% of the green signal, and 11% of the blue signal are added together to form the Y signal. The luminance signal can also be expressed as

$$E_Y = 0.30E_R + 0.59E_G + 0.11E_B$$

where $E_R$, $E_G$, and $E_B$ are the voltages of the red, blue, and green signals, respectively.

The combination of different amplitudes of color signals is what determines the various shades of gray in a monochrome receiver—white having a luminance of one, black a luminance of zero. The ability of a receiver to determine a corresponding level of gray from color levels is an important feature in the compatibility of color and monochrome TV's because the black-and-white signals can be obtained from the three primary color signals.
video signals and composite sync to generate the composite video signal.

Composite -sync timing signals, color encoder IC6 takes the separate red, green, and blue

FIG. 3—SCHEMATIC OF THE COLOR-BAR GENERATOR. Sync generator IC5 provides the composite-sync timing signals, color encoder IC6 takes the separate red, green, and blue video signals and composite sync to generate the composite video signal.

Chrominance

The color information, or chrominance, (which is ignored in a black-and-white TV) is made up of red, blue, and green signals required to drive the picture tube, minus the luminance signal. Those "color-difference" signals are designated as R - Y (red minus Y), and B - Y (blue minus Y). Color-difference signals are used solely for color reproduction. A special matrix circuit in the receiver can extract a G - Y (green minus Y) signal from the B - Y and R - Y signals. The advantage of changing the color signals into color-difference signals is the reduction of three color signals into two.

The R - Y, B - Y, and G - Y signals are decoded at the receiver by adding the Y signal back to each of the difference signals. A 3.58-MHz subcarrier is sent by the transmitter and used in the receiver to restore the original color information.

The frequency and phase angle of the 3.58-MHz subcarrier in the receiver must be the same as that

Motorola Other components:
L1—400-nS delay line (TK1001)
L2—30 µH, 2.52-MHz transformer (TK1603)
XTAL1—3.58-MHz colorburst crystal
XTAL2—503-kHz ceramic resonator
Miscellaneous: enclosure, stand-offs for mounting circuit board, 4-pin friction-lock connector for J2, straight-header connector for J1, four shorting jumpers, wire, solder, etc.

Note: The following items are available from Geko Labs, 13019 250th Place SE., Issaquah, WA 98027-6730, (206) 392-0638: etched, drilled and plated-through PC board $30.00; a complete kit including all parts, PC board, and assembly instructions $80.00; a complete assembled and tested unit $125. Add $5.00 S & H with any order. Washington residents add 8.1% sales tax.
in the transmitter for proper color reproduction. Synchronization is performed by transmitting a small sample of the 3.58-MHz subcarrier during the horizontal sync pulse. That color sync interval is also known as the colorburst. The colorburst signal is used as a reference to synchronize the phase and amplitude of the color subcarrier. The colorburst also determines the tint and saturation of the color that is displayed.

Theory of operation

Figure 2 shows a block diagram, and Fig. 3 a schematic of our video generator. Sync generator IC5 is used to provide the composite-sync timing signals. The outputs are composite sync, composite blanking, and a buffered output of the sync oscillator. The sync generator uses a 503-kHz ceramic resonator (XTAL2) as a base oscillator. The 503-kHz frequency is divided by 32 for the horizontal sync, and further divided to derive the vertical sync-timing signals. Those signals are all combined into the composite-sync signal which is sent to the MC131377 color encoder (IC6). The color encoder takes the separate red, green, and blue video signals and composite sync to generate the composite-video signal.

The 3.58-MHz colorburst crystal, XTAL1, is the reference oscillator for the chroma information. Capacitor C12 allows fine tuning of the reference oscillator to be exactly 3.579545 MHz. The combination of R2 and C5 set the timing for the insertion of the colorburst signal on the back porch of the composite video signal. The values used for R2 and C5 set the burst timing to approximately 0.4 μs after sync, and a burst width of 0.6 μs. The network of L2, C9, R6, C8, C11, and R3 provide bandpass filtering for the chroma component. A delay for the luminance channel (–Y) is provided by R4, L1, and R5 to compensate for the internal delay of the chroma signal.

RGB generators IC1, IC2, and IC3 make up the red, green, and blue video signals that drive the video encoder section to make the color bars. One half of IC1 is used as a divide-by-2 counter, which generates the 252-kHz clock for the four-bit counter IC1. The non-inverted blue, red, and green signals are the divide-by-4, -8, and -16 outputs of IC1, respectively. The blue, red, and green signals are inverted, and blanking is added by IC2. The TTL level is reduced to 1 volt p-p by R8, R9, and R10, as shown in Fig. 4.

Construction

The video generator uses a double-sided PC board that is available from the source mentioned in the parts list. We recommend that you use a PC board for this project because the frequencies involved require a large ground plane. Both the component and the solder side of the PC board are shown in the article if you wish to make it yourself.

Construction is fairly straightforward. Install all components according to the parts placement diagram shown in Fig. 5. Make sure C15–C17 are inserted into the board with correct polarities. Install all IC’s last, observing correct orientation. Since the IC’s are static sensitive, make sure you follow the manufacturer’s recommendations for proper handling.

Checkout

The measurements listed in this section will help to make sure the video generator is working properly. The power on tests should be made with the +12- and +5-V power sources on. Set R8, R9, and R10 to mid range. If you are the impatient type you can go right to the video output test point TP12 and see what you get. If you’re lucky you’ll have a video signal that probably needs some adjustments. If that’s the case you can proceed directly to the video adjustments section. If not, proceed slowly through the following steps to isolate the problem and verify each of the listed voltages, frequencies and waveforms. Keep in mind that you will need an oscilloscope for the video level adjustments.

- Pin 2 of J1 and +5 V—> 2000 ohms (power off)
- Pin 4 of J1 and +12 V—> 1 megohm (power off)

Power-on Tests

- +12-V supply—57 mA
- +5-V supply—29 mA
- IC1 pin 16— +5 V
- IC2 pin 14—+5 V
- IC3 pin 14—+5 V
FOIL PATTERN OF THE COMPONENT side of the double-sided PC board.

81-0101-1

FOIL PATTERN OF THE SOLDER SIDE of the PC board.

- IC5 pin 19—+5 V
- IC5 pin 6 (TP2)—2.504-kHz, 5-V p-p (TTL level) square wave
- IC5 pin 5 (TP1)—TTL-level composite-sync signal
- IC5 pin 13 (TP4)—TTL-level blanking signal
- IC3 pin 5 (TP5)—252-kHz signal
- IC1 pins 11–13 (TP6, 7, 8)—Divided down signals

**Sync Generator**
- IC5 pin 19—+5 V
- IC5 pin 5 (TP1)—TTL-level composite-sync signal
- IC5 pin 6 (TP2)—504-kHz, 5-V p-p TTL-level square wave
- IC5 pin 13 (TP4)—TTL-level blanking signal

**RGB Generator**
- IC1 pin 16—+5 V
- IC2 pin 14—+5 V
- IC3 pin 14—+5 V
- IC3 pin 5 (TP5)—252-kHz signal
- IC6 pins 3–5—1 V p-p signal
- IC6 pins 17 and 18—3.58-MHz oscillator signal
- IC6 pin 16—8.2 VDC
- IC6 pin 1—Ramp signal
- IC6 pins 10 and 13—Chroma signal (Fig. 1-d)
- IC6 pins 6 and 8—Luminance signal (Fig. 1-c)
- IC6 pin 9—2 V p-p (Fig. 1-a)

**Color Encoder**
- IC6 pin 14—+12 V
- IC6 pin 2—TTL-level composite-sync signal (Fig. 1-b)

**Video Level Adjustments**
- IC6 pin 4 (J2C)—Adjust R8 to 1 V p-p
- IC6 pin 3 (J2B)—Adjust R10 to 1 V p-p
- IC6 pin 5 (J2D)—Adjust R9 to 1 V p-p
- J3 pin 1—Terminate into a 75-ohm connector
- Video Output (TP12)—Composite 1-V p-p signal. If you can't get this signal, adjust R8 for the proper peak level and null out the continued on page 129
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mentation, try using a 10K potentiometer in place of R8, and a 1K resistor in place of R9. 10K the tap of the potentiometer as the output to the comparator. That way, a steady amplifier offset is maintained while allowing considerable adjustment of the comparator threshold.

Interfacing
The outputs of the VCP200 can only source and sink small amounts of current and must be protected from reverse EMF and noise. Fortunately, a variety of interface circuits can be devised. One simple power driver is shown in Fig. 7. The desired output from VCP200 is connected to the SELECT input, where it drives the base of the PNP transistor. The transistor supplies power to the relay, which can have any type and arrangement of contacts necessary.

In some cases, it may be handy to be able to toggle an output device on and off. The circuit in Fig. 8 permits just that. Upon power-up, the output of the flip flop will be low. When an active low from the VCP200 is applied to the SELECT input, the output will latch high. The next active low will cause the flip flop output to drop low again.

A more sophisticated output circuit is shown in Fig. 9. On power-up, both outputs will be high. When the Q output of the VCP200 is selected, that output will be latched low. If the Q output of the VCP200 is selected, the output will be toggled high. If a stop signal from the VCP200 is received, both flip flop outputs will be toggled high.

The circuit in Fig. 10 allows a complete reset of the voice recognition circuit and any outlying circuitry with a voice command. When the RESET output of the VCP200 is selected, the monostable multivibrator, composed of the first two gates and the RC junction, produces a pulse that is routed back to the VCP200's reset pin, pin 20. That forces a reset of the voice-recognition circuit. The pulse can be tapped by another CMOS gate (either inverting, as shown, or noninverting, or both), and used to reset outlying circuitry. Given the imperfect nature of voice control, this circuit is recommended.

You now have some basic building blocks on which you can base your voice-control experiments. Keep in mind that, even though the command words understood by the VCP-200 are best suited for controlling a robot, they can be used to control virtually anything.

THE "WORKING" VERSION of the project is made on this board, which omits the LEDs and their drivers.
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IF YOU'RE CONCERNED WITH ENERGY cost and conservation, as most people are, you'll be interested in our energy consumption monitor (ECM). Without the ECM, it's difficult for the average person to determine how much an individual home appliance costs to run. That's especially true for appliances with variable duty cycles such as a refrigerator which will have its compressor and fan motors, lights and other loads on at different times.

Our energy consumption monitor can display the accumulated cost in cents for the connected home appliance load. What did you spend to operate your toaster yesterday? What about your TV or air conditioner? Is the cost of energizing that spare freezer unit worth the few pennies saved when you bought your meat on sale? The ECM will help you to answer those questions quickly.

The ECM can also be used as a power meter by connecting a DMM to the voltage output of the monitor. Using the DC scale of your meter, each volt represents 100 watts. For example, a reading of 0.56 volts would translate to 56 watts.

To give you an idea of what the average residential Long Island, NY consumer pays monthly for operating various appliances, refer to Table 1. The monthly cost was based on a rate of 13¢ per kilowatt-hour (kWh). The average Long Island resident uses about 600 kWh's per month, which translates into a monthly electric bill of $91.81.

The current electric rates for the Long Island, NY area are among the highest in the U.S. and vary depending on the season and the total amount of kWh's used. The summer rates are 12.87¢ for 0–250 kWh's used and 14.1¢ for 250–350 kWh's. The winter rates are 12.87¢ for 0–250 kWh's and 12.33¢ for 250–350 kWh's. Of course electric rates will vary, depending on the size of your family, the region of the country in which you live, and the utility company who services you. The information provided is only a rough basis to compare your own power consumption to that of the average Long Islander. Let's see how this useful device works.

### About the circuit
The ECM circuit consists of four sections, as shown in the block diagram of Fig. 1. A power converter generates a voltage that is proportional to the true or real power consumed by the load. That voltage feeds both a bargraph and a voltage-to-pulse converter. The bargraph gives an approximate indication of the amount of power used, and the voltage-to-pulse converter produces a pulse whose frequency is proportional to the power. The pulse triggers the counter module which displays the cost of powering the monitored load.

### The power converter
In order to determine the actual power consumed by an appliance, we must find the phase angle between the voltage and current in the overall circuit. We know that

\[ P = V \times I \cos \theta \]

where \( \cos \theta \) is known as the power factor. Since the voltage of the monitored load is fairly constant at about 117 volts, we can say that the power is proportional to \( I \times \cos \theta \). To obtain the phase angle, both the voltage and current must be monitored. Transformer \( T_1 \) supplies the voltage, while the current-proportional voltage is obtained by stepping up (by a factor of 20) the voltage drop across shunt resistors \( R_1-a-d \) via \( T_2 \).

---

**ENERGY CONSUMPTION MONITOR**

**Build this energy consumption monitor and find out how much it costs to run your household appliances.**

---

The ECM is capable of accurately monitoring the effective power of inductive loads. If a capacitive load is connected to the ECM, only the apparent power, not the effective power, will be monitored, causing some degree of inaccuracy. That shouldn't pose much of a problem because just about all reactive household loads are inductive. However, some appliances such as refrigerators, freezers, and air conditioners use capacitor-start inductive motors, which are characterized by a high starting torque. Those types of motors will present a capacitive loading effect on the power line, but only during start-up, which is a very short time interval compared to...
# TABLE 1—AVERAGE WATTAGE, USAGE AND COST OF HOUSEHOLD APPLIANCES

<table>
<thead>
<tr>
<th>Appliance</th>
<th>Wattage</th>
<th>Estimated Monthly Usage (Hours)</th>
<th>Monthly Consumption (kWh)</th>
<th>Monthly Cost*1</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Food Preparation</strong></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Broiler</td>
<td>1,140</td>
<td>6.3</td>
<td>7.2</td>
<td>$0.94</td>
</tr>
<tr>
<td>Coffee maker (drip)</td>
<td>1,200</td>
<td>9.8</td>
<td>11.8</td>
<td>$1.53</td>
</tr>
<tr>
<td>Microwave oven</td>
<td>1,450</td>
<td>10.9</td>
<td>15.8</td>
<td>$2.05</td>
</tr>
<tr>
<td>Oven range</td>
<td>12,200</td>
<td>4.8</td>
<td>58.6</td>
<td>$7.62</td>
</tr>
<tr>
<td>Toaster</td>
<td>1,146</td>
<td>2.8</td>
<td>3.2</td>
<td>$0.42</td>
</tr>
<tr>
<td><strong>Home Entertainment</strong></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Color TV (tube)</td>
<td>240</td>
<td>180.0</td>
<td>43.2</td>
<td>$5.62</td>
</tr>
<tr>
<td>Color TV (solid state)</td>
<td>145</td>
<td>180.0</td>
<td>26.1</td>
<td>$3.34</td>
</tr>
<tr>
<td>VCR</td>
<td>20</td>
<td>120.0</td>
<td>2.4</td>
<td>$0.32</td>
</tr>
<tr>
<td>Radio</td>
<td>71</td>
<td>100.9</td>
<td>7.2</td>
<td>$0.94</td>
</tr>
<tr>
<td>Stereo</td>
<td>109</td>
<td>83.3</td>
<td>9.1</td>
<td>$1.18</td>
</tr>
<tr>
<td><strong>Refrigerator</strong></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Frost free, 10-15 years old</td>
<td>—</td>
<td>continuous</td>
<td>141.2</td>
<td>$18.36</td>
</tr>
<tr>
<td>Ref./freezer, frost-free, 10-15 years old</td>
<td>—</td>
<td>continuous</td>
<td>153.0</td>
<td>$23.80</td>
</tr>
<tr>
<td>18-cubic foot ref./freezer, new</td>
<td>—</td>
<td>continuous</td>
<td>100.65</td>
<td>$13.08</td>
</tr>
<tr>
<td>16-cubic foot ref./freezer, new</td>
<td>—</td>
<td>continuous</td>
<td>77.66</td>
<td>$10.10</td>
</tr>
<tr>
<td><strong>Air Conditioning</strong></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Room AC, 6,500 BTUs*4 (before 1980)</td>
<td>EER*2 7.2–930</td>
<td>116.0</td>
<td>108.0</td>
<td>$14.04</td>
</tr>
<tr>
<td>Room AC, 6,500 BTUs (after 1980)</td>
<td>EER 8.5–770</td>
<td>116.0</td>
<td>89.0</td>
<td>$11.57</td>
</tr>
<tr>
<td>Room AC, 6,500 BTUs (after 1980)</td>
<td>EER 9.5–680</td>
<td>116.0</td>
<td>79.0</td>
<td>$10.27</td>
</tr>
<tr>
<td>Central, 3-ton AC (before 1980)</td>
<td>SEER*3 8–4,500</td>
<td>180.0</td>
<td>810.0</td>
<td>$105.00</td>
</tr>
<tr>
<td>@ 12,000W/ton (after 1980)</td>
<td>SEER 9.5–3,790</td>
<td>180.0</td>
<td>682.0</td>
<td>$88.60</td>
</tr>
<tr>
<td>40,000 BTUs (after 1980)</td>
<td>SEER 11.0–3,270</td>
<td>180.0</td>
<td>589.0</td>
<td>$76.57</td>
</tr>
<tr>
<td>Fan (window)</td>
<td>200</td>
<td>150.0</td>
<td>30.0</td>
<td>$3.90</td>
</tr>
<tr>
<td>Water heater</td>
<td>—</td>
<td>—</td>
<td>350.0</td>
<td>$45.50</td>
</tr>
<tr>
<td>Washer (1 load/day)</td>
<td>512</td>
<td>16.8</td>
<td>8.6</td>
<td>$1.12</td>
</tr>
<tr>
<td>Clothes dryer (1 load/day)</td>
<td>4,856</td>
<td>17.0</td>
<td>82.6</td>
<td>$10.74</td>
</tr>
<tr>
<td>Dishwasher (1 load/day)</td>
<td>1,201</td>
<td>25.2</td>
<td>30.3</td>
<td>$3.94</td>
</tr>
<tr>
<td>Iron</td>
<td>1,100</td>
<td>—</td>
<td>5.0</td>
<td>$0.65</td>
</tr>
<tr>
<td>Vacuum</td>
<td>630</td>
<td>6.1</td>
<td>3.8</td>
<td>$0.50</td>
</tr>
<tr>
<td>Clock</td>
<td>2</td>
<td>708.3</td>
<td>1.4</td>
<td>$0.18</td>
</tr>
<tr>
<td>Blow dryer</td>
<td>1,235</td>
<td>6.8</td>
<td>8.4</td>
<td>$1.10</td>
</tr>
</tbody>
</table>

**Notes**
1. The monthly cost is based on an average rate of 13¢/kWh.
2. Energy efficiency ratio.
3. Seasonal energy efficiency ratio.
4. BTUs/EER = watts.

All figures noted in this table were obtained from Long Island Lighting Company's (LILCO) Energy Conservation Department.
the continuous operation of such appliances.

Figure 2 shows a block diagram of the power converter circuit. The voltage from the potential transformer T1 (Fig. 3-a) is delayed by 90° (Fig. 3-b), controlling transistor switch Q1 (Fig. 3-c), which turns on during the negative cycle of the wave. Now, let's consider how three types of loads are monitored: purely resistive, equally resistive and inductive, and purely inductive.

In a purely resistive AC circuit, the current is in phase with the voltage, therefore the half-wave rectified signal from the current transformer will look like that of Fig. 3-d. Since the switch (Q1) is on until the first 90° of the wave, the peak of that wave (Fig. 3-g) will be passed on to the peak detector consisting of IC1. We can now say that $V_{OUT} = V_{PK}$ of the current transformer.

In a circuit consisting of equal resistance and inductive reactance, the current will lag the voltage by 45°. That signal, when half-wave rectified (Fig. 3-e) and gated by Q1 will look like that of Fig. 3-h. As you can see, the switch allows only the first 45° of the wave to be sampled by the peak detector, therefore

$$V_{OUT} = V_{PK} \sin(90 - \theta).$$

since

$$\sin(90 - \theta) = \cos\theta,$$

then

$$V_{OUT} = V_{PK} \cos\theta.$$

An ideal inductor does not dissipate any power, and its AC current will lag the applied voltage by 90°. As a result, once the half-wave rectified current waveform (Fig. 3-f) of such a load is switched by Q1, the resulting output is zero, therefore

$$V_{OUT} = V_{PK} \cos\theta.$$

(Fig. 3-f.)

The schematic of the energy consumption monitor is shown in Fig. 4. Components R6-R8 and C5-C7 form a 90° phase shift network which switches Q1 on via R9 and R10 during the negative-going part of the wave. The voltage present at the secondary of the current transformer (T2) is half-wave rectified by D5. Diodes D3 and D4 limit the secondary voltage to approximately 40 volts peak to protect D6 and Q1 from excessive voltage should a high-current surge occur. When Q1 is turned on, it will couple any of the half-wave rectified signals to R11, and to the peak detector consisting of IC1-a, D7, C9, and R12. The wiper of R11 is set to calibrate the peak detector output so that it produces 1 volt for every 100 watts consumed by the load. Finally, that voltage is buffered via the voltage follower IC1-b to feed an external voltmeter, the bargraph meter, and the voltage-to-pulse converter.

The voltage-to-pulse converter is basically a voltage controlled oscillator (VCO). The power voltage (from IC1-b) charges C10 via R13 and R14 until the capacitor voltage attains the trigger voltage of the Schmitt trigger, consisting of IC2, R16, and R17. Once triggered, the negative voltage swing from the output of IC2 quickly reverses the capacitor charge via R15 and D8, and is ready to repeat the cycle again. The higher the voltage feeding the RC timing network, the higher the pulse repetition, or frequency, will be. That pulse is used to increment the counter module through voltage-divider resistors R18 and R19. Diode D9 assures that the counter sees a pulse of the proper polarity. A nice feature of the display counter is that it is powered by a single AA battery mounted on the back. That makes sure the count is retained if the ECM is...
unplugged, or in the event of a power failure.

At the heart of the bargraph is IC3, a quad comparator. The power voltage drives all of the comparator's inverting inputs while each of the non-inverting inputs are tied to different voltage references derived by the voltage divider network of R20 to R24. As the voltage signal increases above the reference voltage level, the open collector output of that particular comparator goes low, switching its corresponding LED on. Diodes D14, D15, and D16 ensure that the previously lit LED is turned off as the power increases, thus allowing no more than one LED to remain on at a time.

The power-supply section is fairly straightforward. The transformer's (T1) voltage is half-wave rectified, and is then filtered by C1. The voltage divider R2 and R3 is used to boost the output voltage of regulator IC4 to approximately 18 volts. IC4 could easily be replaced with a 7818 voltage regulator, thereby eliminating the need for R2 and R3 (a shorting jumper would have to replace R3). Except for the voltage-divid-
er resistors, the negative supply is basically a mirror image of its positive counterpart.

Construction

Figure 5 shows the authors completed prototype. Transformers T1 and T2, S01, F1, LED1–4, counter display, S1, J1 and J2 are mounted on the enclosure, while the remaining secondary circuitry is installed on a single-sided PC board. The foil pattern is provided if you would like to make your own, or you can obtain an etched and drilled board from the source mentioned in the parts list. Mount and solder all components according to the parts placement diagram shown in Fig. 6, observing correct polarity. The 7815 regulator should be fitted with a heat sink. You can do that by drilling a 3/4" x 1-1/2" x 1/16" piece of aluminum and mounting it to the TO-220 case.

The ECM should be enclosed in a suitable metal case as hazardous line voltage is present. It is important to use no. 14 AWG or heavier gauge wire for all primary wiring. Make sure the neutral side of the plug corresponds to that of the socket. The photograph in Fig. 5 shows where the hot (power supply black lead) and neutral (white lead) conductors are connected.

Grounding should be made by terminating the green grounding conductor of the power-supply cord and socket ground lead to a closed-loop connector. Mount the connector through the transformer mounting screw and secure to the chassis ground through a washer to bite through the painted or plated metal case. You can also sand the paint away to make a good contact. Resistors R1–a–R1–d should be adequately ventilated by using a louvered enclosure top. Those resistors could get quite hot if constant heavy loads over 1000 watts are monitored.

Now it's time to mark a decimal point on the counter display. Using a fine-tip black felt pen, mark the decimal point on the display between the third and fourth digit so that, when the monitor is properly calibrated, each count represents 1/1000 of a cent.

When wiring T2, remember to wire the 6-volt winding across the shunt resistors R1–a–R1–d so that you're using it in a step-up mode.

Locating a 3.6K resistor for R20 may be rather difficult since that is a non-standard value. The author happened to have a few of them in his parts collection, but you may consider wiring a 3.9K and a 47K resistor in parallel to obtain that value.

Calibration and testing

Before applying power to your circuit, double check your wiring. If you're using IC sockets, leave IC1, IC2, and IC3 out of the circuit. Apply power and check for +18 volts and –12 volts at the outputs of IC4 and IC5, respectively. Those voltages may be slightly lower by a fraction of a volt. If you have removed the IC's and the voltages are okay, then unplug the unit, install the ICs, re-apply power, and re-check the supply voltages.

The next step is to check the transformer phasing. In order to do that, temporarily install a jumper from ground to the cathode of D5. Now connect a 100-watt light to the load socket. Using a voltmeter on the AC scale, make sure the voltage between TP1 and TP2 is lower than that measured between TP1 and ground. If it isn't, reverse the two PC-board connected T2 leads. Re-check and remove the jumper.

With NO LOAD connected to
FIG. 6—PARTS PLACEMENT DIAGRAM AND WIRING connections. Use 14 AWG wire for all primary leads, and make sure you wire the hot and neutral leads of the power supply cord to the proper terminations on the AC socket.

-OD-

8-CONDUCTOR RIBBON CABLE

-CONDUCTOR RIBBON CABLE

R28 R22

..--R23---,

Cl et ro e3
tro +1

D12-114-

l 53

i

IC5 D11--1101-

-1+

D2

r,.1--526----

ri, R2

--44111--

LI

1°10-44

D1

--01--

1D4

1

525

-521- -R20-

I-1

D

2

IC3

T1

D7

95

fa

01

T2

O

T1

k)1 GROUND

TERMINATION

CO1

524

R18

1

COUNTER TERM. #4

SO1 HOT

BACK PANEL

GND

R11

NEUTRAL

WHITE LEAD HOT

BLACK LEAD HOT

GROUND TERMINATION

THIS IS THE FOIL PATTERN of the solder side of the PC board.

the ECM, connect a jumper between the +18-volt supply and TP2. Connect a DC meter to the power-voltage output and check to see that the voltage varies from 0 to approximately 16.5 volts as R11 is varied from one end to the other. As you do that, the LED's should increment at about 0.27, 1.8, 5, and 10 volts. Now, using the formula

\[ V_{\text{CAL}} = 36 / \text{rate} \]

where \( V_{\text{CAL}} \) is the calibration voltage and rate is your cost in cents per kilowatt-hour (check your billing statement or power company for that rate). Adjust R11 to read that value on the voltmeter. That will enable you to calibrate R14 so that you obtain one pulse per second (1 Hz) at TP3. A doubling or halving of \( V_{\text{CAL}} \) should approximately double or half the pulse rate. Remember, each pulse represents \( \frac{v}{1000} \) of a cent.

Disconnect the jumper used in the previous procedure and connect a 100-watt light as a load. Using an oscilloscope, monitor the waveform at TP2 and set R7 so that the sampling ends at the very peak of the incoming waveform, which should look like the waveform of Fig. 3-g.

Finally, power calibration is the last to be performed. With the 100-watt light connected adjust R11 so that a DMM, connected to the external voltmeter jacks, displays 1.00 volt DC. You may want to verify that wattage by measuring the voltages across shunt resistor R1, and the line. With those two voltage readings, the power may be calculated using the formula

\[ P(\text{watts}) = \frac{V_{\text{SHUNT}} \times V_{\text{LINE}}}{R_{\text{SHUNT}}} \]

where \( R_{\text{SHUNT}} \) is the shunt resistance (four 0.39-ohm resistors in parallel = 0.0975 ohms). \( V_{\text{SHUNT}} \) is the voltage drop across R1 and \( V_{\text{LINE}} \) is the AC line voltage.

That completes the assembly and calibration of the ECM. There is one point that should be mentioned here. The voltage to pulse converter will not start until there is a load of approximately 30 watts, meaning that the counter will not increment unless the load is heavier than that value.

For those of you wondering if investing in an energy consumption monitor is worthwhile, consider this: You'll be able to determine how much it costs to run a particular appliance for a certain length of time. So it's easy enough to figure out if it's actually cheaper to run the microwave oven for five minutes or the conventional oven for ten minutes, and so on. Using the energy consumption monitor, you'll also be able to determine if buying extra meat at really good sale prices actually saves you money in the long run. The greatest advantage of the energy consumption is keeping one step ahead of your power company.
How American Cablevision's "bullet" zapped signal pirates.

KEN FOLEY

ON WEDNESDAY, MARCH 13, 1991, American Cablevision of Queens fired their first infamous electronic "bullet." According to American Cablevision, they fired a direct hit. Within minutes their switchboard was overloaded with calls from subscribers whose television sets had gone black. American Cablevision was elated—the victims had unsuspectedly taken the bait.

The next morning, American Cablevision sent armies of technicians to service the homes of the complaining customers. They replaced the cable converter boxes, and took the dead boxes back to the electronic coroner's laboratory, performing hundreds of autopsies. According to official American Cablevision records of the mass epidemic, the "Certificates of Death" were identical—illegal chip "zaps".

On Wednesday April 24, 1991, American Cablevision filed a civil suit in New York City federal court against three hundred and seventeen alleged cable pirates. That was the first time such a large number of cable crooks had been arraigned together. American Cablevision offered the defendants a deal: Pay five hundred dollars within twenty days, or face prosecution and fines from one thousand, to one hundred and ten thousand dollars.

"I think this is something that everybody's going to have to start doing," said American Cable President Barry Rosenblum. American Cablevision has approximately three hundred and thirty thousand paid subscribers in Queens and Brooklyn, and estimates it forfeits hundreds of thousands of dollars each year to video marauders, and plans to fire more bullets. The electronic
bullet is the brainchild of Jerrold Communications of Hatboro, Pennsylvania. It was first fired in 1990, by Greater Media Cable of Philadelphia.

In three separate assaults, Greater Media Cable blasted away, netting three hundred and sixty eight illegal converters, which garnered a bounty close to twenty thousand dollars.

We spoke to Jim Bathold, spokesman for Jerrold Communications, to confirm American Cablevision's story that the electronic bullet is a signal fired from a cable company's headquarters directly into a customer's cable converter. If the box is legitimate, the customer never knows he was just zapped. But if blackmarket chips were installed in a basic converter to circumvent paying the monthly service charge, the bullet uses the chips' own programs to neutralize the decoder and halt the cable service immediately.

Mr. Bathold then elaborated "Yes, that is basically how the bullet works," he confirmed. "But it would not be in our best interest to elaborate, or explain the operational procedure in detail. Otherwise it tells subscribers, 'Here we come.' We have not put one word out there in writing of how it works—no press packages or news releases. We especially wouldn't go into detail with electronic hobbyists," he choked out laughing.

Hoping to fare better in Jerrold's engineering division, we were fortunate to reach an engineer that was also a reader of Radio-Electronics. His boss, Stan Dori, said: One of the approaches pirates have been taking for years to defeat scrambling is to physically use a decoder box to unscramble the scrambling method. That is, to reverse engineer the legitimate descrambler's software.

The bullet came into being because one of Jerrold's customers (a cable company) told them of rumors that pirates were defeating Jerrold's scrambling technology. And the cable company wanted to aggressively pursue them. So Jerrold acquired a number of the pirate devices through various methodologies, and reverse engineered them so that a counter measure could be developed. That counter measure was the bullet, an offensive signal that Jerrold can send down the data stream to neutralize what the pirates reverse engineered. That's the bullet—double-reverse engineering.

Dori continued, "So by understanding what the pirates are doing and not doing to defeat current technology, we're able to launch a counter-offensive signal, the bullet, to defeat them."

In the hopes of discouraging customers from buying illegitimate descramblers, information regarding the bullet is being leaked from the cable industry, which claims they are losing up to three billion annually from piracy.

According to Jodi Hooper of the National Cable Television Association, "People think cheating on cable services is like a school prank. They don't really think they are committing a crime and stealing. They just don't take it seriously." Hooper also indicated that some cable companies are offering complete amnesty to people who come forward before their systems are audited and the bullet is released. She says if the culprits wait until they are discovered, they will chance the possibility of criminal prosecution and heavy fines.

Richard Aurello, president of Time Warner's New York City Cable Group, compares cable piracy to shoplifting. "Now that we have the technology, we're going to use it to rope them in." But it's a migraine for the cable industry. Most of the cable companies began scrambling their satellites in 1986, and are now concentrating on detecting people with decoders and illegal hookups.

The National Cable Television Association says about eight million homes nationwide are linked illegally to basic cable signals. And an additional three million homes illegally tap into pay services such as Cinemax and HBO.

But from 1975 through last year, the number of basic service subscribers nationwide grew from nine million to fifty-five million. The U.S. Telephone Association reports that the average basic cable rate nationwide jumped sixty-eight percent between 1986 and 1989.

So even though the cable companies are reporting that losses from theft have tripled during the same period, cable industry revenue has jumped about seventy percent from over ten billion in 1986 to almost eighteen billion last year.

Such large revenues have caused some consumer groups to become skeptical of the cable companies' claims of being financially wounded by theft. "There is no justification for using speculative high-theft figures to justify outrageous rate increases," says Ken McEldowney, head of San Francisco-based Consumer Action.

Another method the cable companies are using to detect pirates, is the "closed circuit radar gun," or time-domain reflectometer. The major drawback with the reflectometer is that it has to be physically attached to the cable entering each home to detect unauthorized connections or decoders. Other than that, sleuthing is still done primarily by inspectors who spend their days eyeballing exterior cables for tampering.

So naturally if the cable industry succeeds in scaring thousands into confessing, it will score a two-headed victory. First by recovering millions in lost revenue having people sign up—as was the case for Utah's TCI Cablevision in 1989 where they ran a blitz advertising campaign showing guilt-ridden signal pirates imprisoned—and second by having the option of keeping the bullet in reserve as a secret weapon and not necessarily having to pay the hefty zapper fee to Jerrold Communications.

Now Time Warner, the second largest cable company with over six million subscribers in thirty-six states, is threatening to start firing bullets nationwide. Are they bluffing?

If they are not bluffing, they will undoubtedly catch more cable thieves who are foolish enough to run to their cable company to complain that their pirated cable box is not working properly.
How good is your amplifier? Our inexpensive THD analyzer will let you know.

JOHN F. KEIDEL

HAVE YOU EVER WONDERED EXACTLY how good an amplifier is, or whether it actually measures up to the manufacturer’s specification of its Total Harmonic Distortion, or THD? Or are you curious if the amplifier you’ve designed is better or worse than a store-bought one? If the answer is yes to any of those questions, then you should build our inexpensive THD analyzer. You can use it to test “home-brew” amplifier breadboard circuits or commercial equipment such as stereo receivers, preamps, and power amps. The analyzer uses an ultra-pure 1-kHz test signal to measure THD at a user-selected voltage level for voltage amplifiers, or a desired power level for checking power amps up to 600 watts. It will detect THD levels down to 0.005 percent! It features a built-in one-percent THD calibrator, a full array of input and output processing controls, and uses your digital multimeter (DMM) as a readout device.

Circuit description

As shown in Fig. 1, an NE5534N low-noise, low-distortion op-amp, IC1, is configured as a Wien Bridge sine-wave oscillator. Carefully matched RC values (R2-C1 and R3-C2) in the frequency-selective positive feedback network contribute to its low distortion level. Resistor R1 and bulb LMP1 form the stabilized negative feedback network that provides a constant-amplitude output signal. DC offset control R4 keeps DC current out of LMP1, which minimizes second harmonic distortion content. Filter network R6-C6 further reduces any residual distortion. After passing through fixed and variable attenuators, plus a buffer amp (IC2), the signal emerges at output jack J1.

The output signal from J1 drives the input of the device under test (DUT), usually an amplifier. The DUT’s output, which includes some degree of distortion, is applied to the input of the analyzer at jack J2. The fundamental frequency (1 kHz) is then removed from the output signal of the DUT, leaving only harmonic distortion components. Combination notch/high-pass filter circuits IC3 and IC4 (both TL074’s) perform the removal function. One feature of IC3’s three-stage RC active filter is that it maintains a constant 45-dB notch depth over its full tuning range. The filter is connected in series with an identical second filter (IC4), to provide a 90-dB notch of the fundamental signal. The resultant frequency response of the combined filters is 27 dB down at 20 Hz, which helps suppress 60-Hz hum and
FIG. 1—SCHEMATIC DIAGRAM of the simple THD analyzer. An NE5534N op-amp (IC1) is configured as a Wien Bridge sine wave oscillator. Carefully matched RC values in the frequency-selective positive feedback network contribute to the low distortion level. Resistor R1 and incandescent bulb LMP1 form the stabilized negative feedback network that provides a constant-amplitude output signal.
other low-frequency noises. Above the 90-dB notch frequency, the response is flat (± 0.5 dB) from 2 kHz to 100 kHz.

When S4, the THD/REF switch, is in the THD position, a signal containing only THD components is channeled through from the output of the \( \times 1/\times 10 \) THD amp (IC4) to the input of the AC-to-DC converter, IC5. Although both polarities of the applied signal are rectified by this circuit, only the positive averaged signal is fed to the low-pass filter R41-C20. The output buffer, IC6, is a CA3193 precision op-amp, stable enough to provide accurate volt, millivolt, and microvolt DC level output signals to an external DVM.

Dual op-amp IC9, a CA3260, serves as a calibrator. The first stage affords precision half-wave rectification of the master oscillator's signal. That same applied signal is AC coupled by C30 to the second stage, biased at \( \frac{1}{2} V_{CC} \) for linear transfer to its output. Voltage divider R49-R50 mixes a very small portion of half-wave output at pin 7 of IC9 with a much larger full sine wave seen at pin 1. Since the signal swings slightly more positive from its quiescent level than it does in the negative-going direction, it is considered to have a specific amount of second harmonic distortion. That amount, by design, is one percent.

Wall transformer T1 feeds half-wave power diodes D4 and D5 through connector J4, fuse F1, and power switch S5. Capacitors C22 and C25 are the principal filters for the positive and negative supplies, respectively. Smaller filters, C23 and C27, along with high-frequency transient suppressors C24 and C28, are included at the output side of regulators IC7 (a 7815 +15-volt regulator) and IC8 (a 7915 −15-volt regulator).

Construction
Breadboard assembly of the analyzer is not recommended, although a seasoned builder may wish to attempt it. It's best to either make your own PC board from the foil pattern we've provided, or order one from source mentioned in the parts list.

Mount all components as shown in Fig. 2. Check orientation of all polarized parts as you install them, and the use of IC sockets is suggested. All power-line wiring, including the LED1 indicator, uses two wires plus a shield. Connect the shield wires
FIG. 2—PARTS PLACEMENT DIAGRAM. Note that JU1 is actually an 8-pin DIP socket in which a jumper is placed in either the far-left or right side (see text). Also note the six components soldered to the front-panel-mounted controls.

Together and then to ground, to prevent hum pick-up. Also, ground the frame of potentiometer R10. Bare ground wires for each BNC connector may be wrapped around the connector body prior to installing the retaining nut. The ground binding post (J8) on the rear panel provides an optional, external earth-ground connection when measuring microvolt-level THD signals. Just connect J8 to circuit ground. The grommet used to hold bulb LMP1 should have a ¼-inch inside diameter. It is glued to the top side over the hole for maximum resiliency.

In addition to the components that are soldered to the circuit board, also note that five resistors and one capacitor are soldered across the terminals of panel-mounted controls Si, R10, S2, and J2 (see Fig. 2).

Capacitors C1 and C2 must be matched to better tolerances than their marked 1% values. If you don’t own or can’t borrow a capacitance meter, you can build and use the simple circuit shown in Fig. 3. Adjust the calibration potentiometer with any one of the four 0.01 µF capacitors (C1, C2, C11, or C13) inserted as CX to read 1.000 volt on your DMM’s 2-volt DC range. (You can consider the reading to be 0.01000 µF.) Now measure the remaining three capacitors and select the two that are closest in value. Absolute value is not important; we simply want them to be the same value. However, if one capacitor measures 80 pF lower than another, you can solder an 82-pF mica capacitor on the underside of the PC board in parallel with the selected capacitor. The leftover 1% 0.01 µF capacitors can be used for C11 and C13 without having to be closely matched.

Likewise, resistors R2 and R3 must be close in value. Using your DMM on its 20K resistance range, select two 15.8K resistors that are the closest in value. If the match is less than perfect, solder a small-value resistor in series with the lower value to raise it to the exact value of the higher one. You can mount two resistors in place of one by putting one through each hole in the board and soldering the raised ends together. Again, the leftover 1% 15.8K resistors can be used for R18 and R27 without having to be closely matched.

You may wish to build your own enclosure for the THD meter. A silk-screened front panel measuring 9¼ inches wide by 3 inches high can be purchased from the source mentioned in the parts list. If you decide to purchase the front panel, and build your own enclosure, build it to fit the front panel and drill several ¼-inch holes in the top panel near the voltage regulators to allow heat to escape. Otherwise you can purchase the same enclosure used for the prototype: the exact
model number is listed in the parts list, and the front panel is designed to fit it perfectly. Figure 4 shows the completed prototype.

**Adjustments**

Using a DMM on its millivolt-DC range, connect it between pin 6 of IC1 and ground. Adjust R4 for a reading of 000.0 mV on the meter. Next, set the INPUT switch (S2) to "<20V," the THD AMP switch (S4) to "x10," and the THD REF switch (S5) to "THD." With the DMM still set to read millivolts DC, connect its leads across the + and − DVM binding posts (J6 and J7) and adjust R43 for a reading of 000.0 mV on the meter.

Filter-null adjustments may be made with an oscilloscope or DMM on its lowest AC voltage range (typically 2 volts). Insert a jumper in the JU1 jumper block (NULL-OPER) in the left-most position when facing the front panel (next to R22, or the "null" position). Connect a short coaxial cable between OUTPUT jack J1 and INPUT jack J2. With power on, and S2 in the "<20V" position, S4 in the x1 position, S5 in the "THD" position, and the rotary ATTEN DB switch (S1) on VAR potentiometer (R10) in the minimum attenuation position, or fully counter-clockwise.

Connect a scope or voltmeter to the IC4-pin-7 side of R31, and connect the ground lead to any ground in the area. If you're using a scope, set VOLTS/DIV switch (on the scope) to any position between 5 and 50 mV/div. Carefully adjust potentiometer R30 for the best null on a scope screen, or lowest reading on a meter.

Transfer the test probe to the
IC3-pin-1 side of R20. Adjust R22 for a minimum reading on the scope CRT or voltmeter. Insert the jumper previously placed in the NULL position in the C12 side of the JU1 jumper block. You may be able to squeeze 1 or 2 dB more null from the system by shifting the DMM to the DVM binding posts (J6 and J7), setting the meter to its 200 mV DC range, setting S4 to $\times 10$, and trimming the adjustments of R22 and R30.

**Checkout and use**

Connect a scope or DMM set to read AC volts to output jack J1. Rotate S1 and R10; the output signal voltage should vary accordingly. The controls are attenuators, not gain controls, so maximum signal occurs in the counterclockwise position.

Check the INPUT P-P switch S2 by applying the signal from J1 to J2, setting S5 to “REF,” and measuring the AC output at J3. The output signal should be maximum with S2 in the $<$20V position and minimum in the $>$20V position. It’s important to note that if the input signal to J2 goes much higher than 20V peak to peak, and S2 is in the $<$20V position, clipping will begin to occur.

You can use the analyzer’s calibrator to check all remaining functions. Connect a coaxial cable from J5 to J2. Set all toggle switches to the upright positions, and connect your DMM to J6 and J7 to read DC volts or millivolts, as required. Now, if the THD signal reads 24.0 mV, which is 0.024V, and you switch the S5 to “REF,” the DMM should then read 2.40V, or a number very close to that. When the resulting fraction (0.24/2.4) is multiplied by 100 it should produce a THD percent figure of one percent.

Figure 5 shows the test setup for THD measurements. Say that we are measuring percent THD of a 50-watt amplifier that’s connected to an 8-ohm power resistor load. By ohms law, it will take 20 volts across 8 ohms to produce 50 watts. Now let’s say continued on page 128

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**THD Measurement**

When a pure or undistorted sine wave is applied to a circuit containing vacuum tubes, transistors, or op-amps, which is used to provide linear transfer of the signal, some degree of distortion always appears at the output.

As the near-perfect sine wave travels through the device under test (DUT), its shape is altered due to inherent non-linearities within the circuit. Waveform alteration is the equivalent of adding harmonics, or multiples of the fundamental frequency, in varying phases and amplitudes to the fundamental signal. It can be shown mathematically and by measurement that these spurious harmonic components are vested within the output signal’s waveshape.

Spectrum analyzers are used to sweep over the frequency range of interest, separating the fundamental and its harmonics into individual signals. It provides a CRT display of these various signals, including odd and/or even harmonics in their proper amplitudes and frequency locations relative to the fundamental. Similar instruments use digital signal processing techniques rather than a sweep method to provide more in-depth data including phase angles.

Wave analyzers are essentially tuned filters that include a measurement window that is manually frequency shifted through the range of interest. It offers meter-readout amplitude measurements of the various harmonics relative to the fundamental frequency.

THD analyzers are used to notch out the fundamental from the DUT’s output signal leaving only the harmonic components intact. It then measures the sum total of the residual signals in terms of their RMS or average values, and compares them to the overall amplitude of the output signal which is taken as a 100 percent reference. The ratio of the THD measurement to the reference, multiplied by 100, equals the percent THD.
SIMPLE FM TRANSMITTER

This handy FM transmitter makes a great one-evening project, even for first-time builders!

JAMES A. MELTON

THERE IS NO THRILL LIKE THE THRILL you get from operating equipment you have built yourself. If you have never built a project from a magazine before, let this FM transmitter be your first—you'll see how much satisfaction and fun you can have!

The FM transmitter is designed to run from a 9-volt battery and is made from readily available parts. The author's primary use is as a baby monitor, but the uses of a transmitter like this one are almost limitless. It is very sensitive, and easily capable of picking up a conversation in any part of a room. The dimensions and values given here will allow static-free reception within the perimeter of most homes.

No license is required for this transmitter according to FCC regulations regarding wireless microphones. (The emissions must stay within a band of 200 kHz, its output between 88 and 108 MHz, and the field strength of the radiated emissions must not exceed 50 µV/m at a distance of 15 meters from the device.) If powered from a 9-volt battery and used with an antenna no longer than 12 inches, the transmitter's radiated power will be within the FCC limits. The FCC takes a dim view of persons operating outside the legal power limits, so please do not substitute any components in this circuit which would alter the output power.

Circuitry

Take a look at the schematic in Fig. 1. Audio is picked up from the room by an electret microphone and amplified by Q1. Resistors R2–R5 set up the DC operating bias of Q1. Capacitor C3 serves to improve the AC response to the audio voltage, and C2 blocks the DC bias and couples the AC to the next stage, where the RF action takes place. The amplified AC voltage from Q1 is routed to the base of Q2. Transistor Q2 and associated circuitry (C5 and the inductor) form an oscillator that operates in the 80–130 MHz range. The oscillator is voltage-controlled, so it is modulated by the audio voltage that is applied to the base of Q2.

Resistor R6 limits the input to the RF section, and its value can be adjusted as necessary to limit the volume of the input. That will help control the amount of distortion you have on very loud inputs. Resistors R7–R9 set the DC operating bias of Q2, another 2N2222 that's used as the oscillator and modulator of the transmitter. Capacitor C5 is a 6–50 pF trimmer capacitor that's used to tune the oscillator tank circuit, and C4 routes the RF from the oscillator to ground to prevent unstable operation.

Construction

The FM transmitter is built on a piece of perforated construction board with 0.1-inch hole spacing. Component spacing is not critical, but placement is. You should place the components on the board in a layout that is similar to the prototype shown in Fig. 2. Generally, you will also want to make the transmitter as small as possible.

Let's start from the left side of the schematic and work to the right. You'll want to cut out a piece of perfboard that is 12 holes wide and 30 holes long. That will give you plenty of room to work with, but still produce a small unit. First lay out two power lines on the board with bare wire: the positive supply from the battery will be on top, and the negative (ground) will be on the bottom. A 1K resistor (R1) supplies the bias voltage for the microphone. Remember to install the resistor vertically, next to the positive supply line, and bend the other end of the lead to the board. Go through the board and down toward the ground bus. Now insert the microphone leads into the
When powered from a 9-volt battery and used with an antenna no longer than 12 inches, the radiated power will be within the FCC limits.

**Operation**

To use the transmitter, set up a radio in the area at least 10 feet from the project. Find a blank spot on the dial and turn the radio up so you can hear the static.

Connect a 9-volt battery to the transmitter and listen to the radio. Slowly adjust the tank capacitor (C5) until you “quiet” the receiver; this is the tuned spot. Note that when you remove your hands from the transmitter, you will detune the circuit somewhat. It is usually best to leave it detuned, and tune the radio in to get the best reception. If you cannot get the tuning range you desire, you can squeeze the coils in the tank circuit closer together to raise the frequency, or pull them apart just a little bit to lower it.

The circuit works best when powered by a battery, but if a wall-derived supply is needed, make certain that the ripple voltage is as low as possible, or you will get hum in the receiver.
IT'S BEEN ONLY A DECADE SINCE COM-
pact disc digital audio was intro-
duced. In that short time, the
compact disc or CD has brought
high-quality audio reproduction
to the masses, and taught us to
appreciate good sound. We're not
exaggerating when we say that
the CD has changed the way we
listen to music.

It's rare for a new technology
and format to catch on so quick-
ly—especially one that threatens
to make its predecessors ob-
solete. CD was a success not only
because of consumer acceptance,
but because it also offered some-
thing to manufacturers, record-
ing companies, and retailers.

It wasn't the CD's "gee whiz"
appeal—nor was it the promise of
perfect audio reproduction—that
caused sales to catch fire. It was
convenience. When compared to
the LP that it replaced, CD's were
a dramatic breakthrough. They
can store more audio in a pack-
age a fraction of the size. They
can be lent to even your most
careless friends without getting
scratched. They even play back
more conveniently, because you
can skip tracks that you don't
want to listen to, or re-arrange
the order in which the songs play
back.

It's convenience, also, that
makes the venerable compact
cassette our music medium of
choice. (Cassettes outsold CD's
every year until last year.) They fit
in your shirt pocket, and they
stand up reasonably well to
abuse. They're ideal for use in a
car or in a personal stereo be-
cause they're relatively immune
to shocks. So what if they can't
come close to the audio quality of
a CD or even an LP?

How about DAT?
In the belief that consumers
had fallen so much in love with
the idea of digital audio because
of their exposure to CD, Japa-
nese manufacturers reasoned
that Digital Audio Tape (DAT)
would be to the CD what the com-
 pact cassette was to the LP. Unfortu-
nately, it didn't work out that
way for a number of reasons.
First, the record industry,
spearheaded by the RIAA (Re-
cording Industry Association of
America), threatened lawsuits
against any Japanese manufac-
turer who exported the DAT ma-
chines to the U.S. The RIAA was
concerned about DAT's potential
to make virtually perfect copies of
CD's. (They seemingly missed the
fact that, for most people, cas-
ettes do the same thing. And de-
spite that, pre-recorded cassettes
have outsold both LP's and CD's
combined since 1982! They've
outsold blank tapes as well.) The
threats of lawsuits were enough
to stop DAT dead in its tracks,
despite considerable accolades
for the format in the audio and
general press.

Although some DAT machines
were available on the "gray mar-
ket" of unofficially imported
goods, DAT officially arrived in
the U.S. market in 1990—with
generally disappointing results.
Whether it was the years of delay,
the taint of the lawsuits, the ex-
 pense of the machines, or the
lack of pre-recorded software that
have killed DAT in the consumer
market, we'll never know for
sure. Perhaps DAT failed because

BRIAN C. FENTON

Two new digital audio formats—Sony's
Mini Disc and Philips' Digital Compact
Cassette—promise to battle each other as
they create consumer confusion.
Enter DCC

In January of 1991, Philips announced that "a new era of audio reproduction has started." DCC, a digital extension of the compact cassette, would offer "the best opportunity available for consumers and industry to enter into the field of digital recording." Tandy Corporation announced that they would be the first U.S. licensee of Philips' technology, and would introduce a home recording deck in late 1992.

The most important feature of DCC is that it doesn't make the familiar cassette obsolete. All DCC players will play back existing analog cassettes, so even when you make the jump to DCC, you can still listen to your existing library of tapes. (You won't, however, be able to record analog cassettes on your DCC machine, or play DCC tapes on your standard cassette deck.) That "backward compatibility" could convince some consumers to upgrade to DCC even though they like what they already have. After all, an upgrade won't just give them better sound, but as we'll see, more convenience as well.

A DCC deck is essentially a standard cassette recorder that includes some extra digital electronics and a new head design. The dimensions of a DCC cassette are essentially the same as that of a standard cassette, but the digital cassette's sides are flat—the case doesn't get fatter where the head enters the shell. Also, since the DCC standard demands that all DCC players feature auto-reverse, there's never a need to flip the tape over, so you don't need to have holes for the reels on both sides of the cassette. That means that one full side of the cassette can be used for information and graphics—something the recording companies love.

The spool holes and the tape surface are protected against dust and fingers by a sliding metal cover, which also locks the tape hubs. There's no need for an carrying case, so the digital cassette is easier to use and store, especially in a car.

The key to maintaining compatibility with standard cassettes is a new thin-film semiconductor head, manufactured using a process similar to that used for integrated circuits. The first layer of the head contains one set of 9 magneto-resistive heads for digital playback, and a pair of similar heads for analog playback. On the second head layer is one set of 9 integrated recording heads for digital recording. We'll see shortly why 9 digital heads are required.

PASC makes it work

The key to the DCC system is the a new digital coding technique called PASC, or precision adaptive sub-band coding. The goal of PASC is to produce a signal equivalent to that of a CD. The results? A dynamic range better than 105 dB, and a total harmonic distortion, including noise, of less than 0.0025%.

PASC is based on two important psychoacoustic principles. The first is that we can hear sounds only if they're above a certain level, called the hearing threshold. The second is that loud signals mask soft ones by raising the hearing threshold.
The hearing threshold, as you might expect, varies from person to person. Even a very sensitive ear, however, won’t be able to hear a sound if it is masked by a louder sound. (You couldn’t, for example, hear an unamplified violin at a rock ‘n’ roll concert!) The theory behind PASC’s efficiency can be expressed by the question, “If you can’t hear it, why record it?”

During encoding, the PASC processor analyzes the audio signal by splitting it into 32 subband signals. By continuously taking into account the dynamic variations of the hearing threshold, the PASC processor encodes only the sounds that will be audible to the human ear. Each subband is allocated the number of bits that are required to accurately encode the sound within it. If a subband doesn’t require any bits—because it contains sounds that are masked, for example—its bits are re-allocated to other subbands so that the sounds within them can be encoded more accurately. On average, the PASC system needs to encode only one quarter the number of bits that a CD or DAT encoder would to reproduce a given audio signal.

The encoded data is multiplexed into an 8-channel data stream, and error-detection and correction codes are added. The eight channels are recorded on 8 parallel tracks on the DCC tape. The ninth track can be used to carry auxiliary data, such as song titles, recording times, and the like). The auxiliary track could be used to generate hundreds of characters of text per second, so decks could include readouts for song lyrics or other information about the selection.

DCC, an elegant extension of the most popular music carrier we have, seemed to be a sure-fire hit. It had something for everyone, including hardware manufacturers, record companies, retailers, and consumers. It now appears, however, to have run up against a formidable competitor: Sony’s Mini Disc.

**Sony’s Mini Disc**

In May of 1991, in what seemed to be a deliberate attempt to derail DCC before it got moving, Sony announced a brand new recordable audio format, the Mini Disc or MD. Sony, however, denied that their MD was meant to compete with DCC. In response to the question of what MD replaces, the President of Sony Corporation of America answered “We are replacing nothing. We are creating new markets.”

The Mini Disc format is specifically designed for portable applications (personal stereos, boom boxes, etc.) and was scheduled for introduction, conveniently, in late 1992—the same time that DCC decks were due. The disc, about 2 1/2 inches in diameter, looks—and acts—like a cross between a compact disc and a micro floppy computer disk. Like a compact disc, the Mini Disc is an optical medium—it is read by a laser and can store up to 74 minutes of digital audio. Like a floppy disk, the mini disc can be magnetically recorded again and again.

How did they manage to get the same capacity as a CD on a disc that has about 1/4 the surface area? Interestingly, by treating audio in much the same way as DCC does. Sony’s encoding scheme, which is called ATRAC, or adaptive transform acoustic coding, is also based on the psychoacoustic principles regarding the threshold of hearing and the masking effect.

Because the ATRAC encoder ignores sounds that fall below the threshold of hearing (which varies dynamically because of signal masking) it can encode data five times more efficiently than CD or DAT systems. That’s even better than DCC’s 4:1 advantage!

Can a recording that “leaves out 80% of the bits” sound as good as a CD? In theory, if all you’re leaving out is things you can’t hear, then yes. In practice, we don’t know yet. At Sony’s announcement, they demonstrated a prototype by playing some pop/rock for a half minute or so. It sounded OK, we guess, considering that the listening environment was a crowded hotel meeting room. No A/B comparisons were provided between CD and MD. Sony claims that “only 2% of the population will be able to hear the difference.”

The Mini Disc is constructed of four layers, including a newly developed magnetic layer of terbium ferrite cobalt. Since magneto-optical discs can’t come in contact with the recording heads, it’s important that the magnetic material be able to...
MAGNETO-OPTICAL OVERWRITE TECHNOLOGY. When the magnetic layer is heated by the laser, it becomes possible for the magnetic head to change its polarity. The polarity is then detected by the laser during playback by noting the direction of reflection.

change polarity when subject to a very small magnetic field. The new material fills the bill.

The Mini Disc requires both a laser and a magnetic head for recording. When the magnetic layer is heated by the laser (to a temperature of about 400°F), it loses its coercive force—that is, it becomes very easy to magnetize. The head then supplies a magnetic field to set the material's magnetic polarity. When the heated spot cools, the new polarity is "locked in" and, thus, the digital data are recorded.

Sony's Mini Disc has a couple of advantages over other optical recording methods. The structure of the head is much simpler because the laser can be on continuously during recording and playback. And the low-coercivity of the magnetic material greatly reduces the power required, making portable operation feasible.

One feature of Mini Disc touted by Sony is that the portable Walkman players will have "shock-proof memory." One of the problems with current portable CD players is that they don't work too well unless they're standing still. Any sharp jarring causes the laser to mis-track. Mini Disc players shouldn't suffer from that problem because data is read off the disc at a rate far faster than required by the ATRAC decoder, creating a data buffer of three seconds. If the laser mis-tracks, the listener won't hear it. The buffer will feed data to the decoder while the laser finds its way back to the right spot. Sony's announcement included a demonstration where a prototype player was shaken vigorously without any audible result. The prototype continued to play even after the disc was removed until the 1-megabit buffer was empty! Of course, there's no technological reason why portable CD players couldn't offer their own shock-proof memory buffer. But since the buffer would have to be 5 times the size, it would add greatly to the cost.

Who wins?

Ever since we forecast that DAT would be a sure-fire success, we've been reluctant to make predictions. But let's look at some of the issues involved, and how DCC and MD stack up.

For consumers—assuming that both formats offer high-quality audio—DCC has the decided advantage in that existing libraries of cassettes won't be obsolete. Both formats have the potential to supply such convenience features as song title and lyric readouts, but MD offers much faster random access of tracks. Although it's too early to say for sure, prices for home DCC decks should be around $600 when introduced, while a portable MD player is expected to cost more than $400. For consumers, we give DCC a slight edge.

The recording companies will have a hard time taking sides. Both technologies will use the serial copy management system or SCMS, an anti-piracy system. Manufacturers will be able to duplicate DCC at 64 times normal speed on equipment similar to what is now used for standard cassettes. Mini Disc players will be able to play back not only magneto-optical discs, but pre-recorded optical discs as well—discs manufactured using the same process as is used for CDs. Various recording companies have expressed support for each format. Which way will the record companies go? For us, it's too close to call.

Hardware manufacturers should prefer DCC because standard tape transports can be used. Retailers, always reluctant to have to stock the same titles in various formats, are dreading the thought of re-vamping their stores to accommodate either DCC or MD.

What about you? In the long run—since both formats seem destined to compete with each other for your money—it's you who will decide whether DCC or MD is the personal recording format of the 90's and beyond.
This simple circuit will protect your stereo speakers in the event of amplifier failure.

IF YOU'VE HAD FIRST-HAND EXPERIENCE with damaged speakers due to a faulty amplifier, or if you value your speakers enough to want to prevent such damage in the first place, then the circuit described in this article is for you. The circuit will protect speakers against an amplifier that may have a shorted output stage and thus deliver excessive DC voltages that will easily ruin a speaker coil. If your amplifier has a sound-processing delay after the power has been turned on, a functionally similar circuit is already built in. This article will give you a basic idea of the delay's function and how it works to protect the speakers. The circuit is designed for solid-state amplifiers and is not necessary for tube-type amplifiers that have output transformers. (An output transformer blocks any DC from the speaker terminals.)

MARK A. VAUGHT

The circuit is very versatile and can be customized for many different applications. Plans for home stereo, automotive, and commercial PA/guitar amplifier applications are included.

The most common cause of speaker failure is catastrophic amplifier failure. That's in contrast to the conception that the speaker has been overdriven by an amp that's operating normally. Most often the speaker power level has been chosen to match the driving amplifier. Semiconductors designed to handle high current, such as bipolar power transistors and MOSFET's, usually short when they blow out. Often these devices are connected directly to an amplifier's DC power-supply rails or through a small amount of resistance that can't effectively limit the current when the short occurs.

The DC level of an amplifier's power supply is designed to accommodate the peak power levels that occur when the amp is driving the speaker at full power. A 100-watt amplifier has powersupply rails of at least 40 volts. Under normal operating conditions, that level would never be applied to the speaker coil for more than a few seconds. However, if an output device in the amp shorts, the DC is applied to the speaker continuously. In the case of a 100-watt amplifier, that causes a power dissipation of:

\[ P_D = \frac{(40V)}{\text{speaker resistance}} \times 40V \]

Speaker resistance is usually one to two ohms less than the AC impedance. If a blown amplifier is connected to a 100-watt speaker with a 7-ohm DC resistance, the power being dissipated is:

\[ P_D = \frac{(40V)(7\Omega)}{140V} = 228 \text{ watts} \]
The speaker will be able to dissipate that power for only a couple of seconds before the coil is damaged due to excessive heat.

When the protector circuit senses a DC voltage on the speaker line, it activates a relay whose contacts are in series with the speaker; after two seconds the relay disconnects the speaker until the DC is removed. A fuse is inadequate for this application because the value needed to protect the speaker against DC will blow out at peak power levels during normal operation. Conversely, a fuse value chosen to allow peak power operating levels will not protect the speaker against a DC voltage. The protector circuit allows peaks to occur in the power level and also protects the speaker against DC. It should be used in conjunction with a fuse value calculated from peak power levels. The fuse should be placed as close to the amplifier as possible, if not in the same chassis, and is therefore not shown in the protector circuit’s schematic.

**Circuitry**

The protector circuit’s schematic is shown in Fig. 1 and the power supply is shown in Fig. 2. Up to four individual protector circuits can be powered from one supply, although most applications will require only one or two circuits per chassis. The optional 9-volt DC output jack can be used to power a footpedal or fuzzbox, eliminating the need for a DC wall transformer when the circuit is used to protect guitar-amplifier speakers. The power supply can be modified for different applications, and we’ll talk about them later.

Referring to Fig. 1, the voltage-divider resistors R3–R6 are used to bias the positive and negative inputs of the window comparator formed by IC1-a and IC1-b. The inputs are biased at plus and minus 3 volts. The voltage divider also provides a 9-volt reference for the negative input of comparator IC1-c.

Resistors R1 and R2 form an input voltage divider fed from the speaker terminals of an audio amplifier. The input divider is referenced to analog ground, and the output of the divider is connected to the negative and positive inputs of the window comparator (IC1-a and IC1-b). The outputs of IC1-a and IC1-b are open-collector stages, wired together, and pulled high through R7. That forms a wired function and completes the window comparator. When the output of the R1-R2 voltage divider exceeds the reference levels set by R4 and R5, the output of the window comparator goes low and removes the bias from Q1. The input voltage at which that happens is determined by the value of R1. The formulas for calculating R1 are presented later in this article. Transistor Q1 is turned off while the output of the window comparator is low, thus allowing timing-capacitor C1 to begin charging through R8.

Under normal input conditions (an AC audio signal), the output of the window comparator will return to a high level when the input returns to the plus and minus 3-volt range. That biases Q1 into conduction and immediately discharges C1. If a DC signal large enough to trigger the window comparator is present on the input, then Q1 will remain in its off state and C1 will charge until it reaches 9 volts with reference to the power-supply ground. When C1 reaches 9 volts it triggers comparator IC1-c causing its output to go high and bias Q2 into conduction via R9. When turned on, Q2 grounds one end of relay RY1 thereby activating it and disconnecting the audio passing through its contacts from the speaker. The relay contacts will remain open until the DC input is removed from the protector circuit. Diode D1 protects Q2 against reverse-bias spikes generated by the relay coil.

The circuit has two separate grounds: a speaker ground and a power-supply ground. Under no circumstances should these two grounds be connected together. If two circuits are used together, then three entirely separate grounds will exist: a power-supply ground and a speaker ground for each circuit (see Fig. 3-a).

Some stereo amplifiers, especially those used in car-radio amplifiers, have differential or floating-ground outputs for each channel and cannot be connected together. Figures 3-a, -b, and -c are AC model diagrams showing the equivalent connection paths between two circuits.
operating from a single power supply. Figure 3-a clearly shows that the current flowing in either Loop 1 or Loop 2 is not a function of the other. The speaker grounds return to the common power supply rails through the voltage divider resistors R3 – R6. Figure 3-b shows the power-supply capacitor from 3-a, which is seen as a short to AC, replaced by a wire. Finally, in Fig. 3-c, the equivalent resistance between the two speaker grounds is shown as 30K.

If the amplifier speaker grounds are connected inside the amplifier, they should NOT be connected at the speaker protection circuit's chassis. High current is assumed to be flowing in each speaker ground lead and connection of separate ground leads anywhere except inside the amplifier will degrade performance.

Calculations

To calculate the value for R1, which determines the time it takes C1 to reach nine volts, the following example analysis is presented. The first step is to calculate the RMS (average) voltage applied to the speaker terminals to obtain the rated amount of power. We'll arbitrarily use 100 watts and a speaker load of 8 ohms. From the equation:

\[ P = \frac{V^2}{R} \]

we can substitute values

\[ 100W = \left(\frac{28.28V}{8\ \text{ohms}}\right)^2 \]

divide by 8

\[ 12.5 = I^2 \]

take the square root of both sides

\[ I = 3.53 \text{ amps} \]

From the equation:

\[ V = IR \]

we can substitute values

\[ V = 8(3.53A) = 28.28 \text{ volts RMS} \]

As a final check use the formula

\[ P = IV \]

we can substitute values

\[ P = 3.53A(28.28V) = 99.82 \text{ watts} \]

To calculate the value for R1 we use the equation:

\[ (\frac{V_{\text{in}}(R2)}{V_{\text{out}}}) - R2 = R1 \]

and substitute values

\[ = \frac{(28.28V(4.7K))}{3V} = 4.7K = 39.60K \]

To calculate the fuse value for amplifier short-circuit protection, use the equation:

\[ V_f = \frac{V_{\text{RMS}}}{0.707} \]

and substitute values

\[ 28.28V/0.707 = 40 \text{ volts} \]

From the equation:

\[ I = \frac{V_f}{R} \]

we can substitute values

\[ 40V/8 \text{ ohms} = 5 \text{ amps} \]

If you would rather avoid making all of the calculations, Table 1 shows the correct resistance values to be used for R1 for 10- to 300-watt applications. Appropriate fuse values are also provided in Table 1.

The next step is to calculate the maximum time that C1 will charge, and the voltage level it will reach before it is discharged, under normal operating conditions. This is a necessary analysis in order to prove that the circuit will not trigger falsely when peak audio power levels are reached. The lowest frequency normally associated with audio is 20 Hz. It has the longest time period (50 milliseconds) in the audio spectrum so we'll use it for analysis of the speaker protector circuit. (An actual audio signal is quite complex, but the complexity of the waveforms only decreases the time that C1 will charge, so we'll therefore use 20 Hz.)
Capacitor C1 will charge whenever the input voltage exceeds the RMS voltage level necessary to produce 100 watts if R1 is equal to 39.6K. Figure 4 shows the analysis waveforms for the circuit; shown is the input test signal (a), the Q1 base voltage (b), and the Q1 collector voltage (c). Referring to Fig. 4-a, to calculate the time that the input waveform is between 28 and 40 volts, we'll first assume that e1 is the instantaneous voltage level (28.28 VRMS), E_M is the maximum or peak voltage level (40.00Vp), the frequency is f (20 Hz), pi (π) is equal to 3.14, t is the time for sine wave to reach 28.28 V_RMS, and that 2π radians equals 360 degrees (and we'll stick to degrees from this point on). That out of the way, from the equation:

\[ e_1 = E_M \sin(360f) \]
we divide by E_M

\[ \frac{e_1}{E_M} = \sin(360f) \]
we now take the inverse sign

\[ \sin^{-1}(\frac{e_1}{E_M}) = 360f \]
and divide by 360f

\[ (\sin^{-1}(\frac{e_1}{E_M}))/360f = t \]
now we substitute values

\[ (\sin^{-1}(28.28/40))/360(20Hz) = 45/7200 \]
= 6.25 ms

From those equations we can conclude that, for the sine wave of Fig. 4-a to travel from 0 to 28 volts (0.707 x peak value), it takes 6.25 ms, or one eighth of the total period (50 ms) of the waveform. 6.25 ms is also the time it takes the sine wave to return to zero volts. Therefore:

6.25 ms(2) = 12.5 ms
12.5 ms is the total time C1 will charge (4-c). The last step is to calculate the voltage level of C1 at t = 12.5 ms, and we'll assume that e_C is the capacitor voltage at t. E is the power-supply voltage (12V), e equals 2.718, t equals 12.5 ms, C equals 22 µF, and R equals 68K. Now we take the equation:

\[ e_C = E(1 - e^{-t/(CR)}) \]
and substitute values

\[ = 12(1 - 2.718^{-12.5\text{ms}/(22\mu\text{F}68K)}) \]
= 12(0.0084)
= 100 mV

The speaker protector circuit will disconnect the amplifier from the speaker after a 2-second interval using the values shown for R8 and C1. That amount of time will protect the speaker under most circumstances. Charging time for C1 to reach 9 volts can be calculated by rearranging that equation and assuming that e equals 2.718, ln is the natural log (the inverse of e^x), E is the power-supply voltage (12V), e_C is the capacitor voltage (9V), t is the time for C to charge to 9V, C equals 22 µF, and R equals 68K. The rearranged equation is:

\[ t = CR(ln(E/E - e_C)) \]
now we substitute values

\[ t = (22 \mu\text{F}(68K))(ln(12/12 - 9)) \]
= 1.49(1.39)
= 2.0 seconds

To change the time delay for the speaker protector circuit to disconnect the speaker from a DC voltage use the equation R8 = t/1.39C to recalculate the value of R8.

Construction

Construction of the protector...
FIG. 5—PARTS-PLACEMENT DIAGRAM for the protector circuit. The 5-amp relay will mount right on the board, while the 15-amp relay must be mounted on the edge of the board using double-sided tape.

FIG. 6—PARTS-PLACEMENT DIAGRAM for the power-supply board.

circuit depends on the intended use. Once you have a clear idea of the application, then you can customize the circuit to meet your needs. To use the circuit to protect car-stereo speakers, replace R4 and R5 with 3-volt Zener diodes. That will ensure that the window-comparator reference voltages, with respect to analog ground, will be independent of the DC supply voltage. For power levels below 100 watts, you can use the 5-amp relay shown in the parts list; above 100 watts, you must use a 15-amp relay.

The parts-placement diagram continued on page 129
IF YOU THOUGHT A MUSIC ON-HOLD feature for your telephone was only for high-budget professionals, think again. We'll show you how you can add FM music on-hold to any analog telephone line with a Touch Tone telephone. It's ideal for home offices or for people who want to project a high-tech appearance.

Some of the features of this design include: LED status indicator; audio volume control, built-in antenna, only one operating adjustment, and a mute function to eliminate "hiss" in between stations. You can build this impressive device in under three hours, for only $70.

Construction, test, and alignment is made easy due to the use of specialized IC's, namely a single FM receiver chip, IC4, and a DTMF decoder, IC1. There are no special coils to wind, and no tricky circuit adjustments are required. All you need is a DMM to test and align the circuit. Let's now take a look at how the unit works.

On-hold circuit

A block diagram of the unit is shown in Fig. 1, and the schematic in Fig. 2. The FM on-hold device connects to an analog telephone line via an RJ11 modular jack. It's powered by an external +15-volt DC, 150-mA power pack that plugs into a standard 120-volt AC outlet. The 15-volt DC supply passes through polarity-protection diode D11 to the input of IC5, a 7812 +12.0-volt DC voltage regulator. Capacitors C24 and C25 provide decoupling and anti-oscillation protection for the regulator. The regulated output of IC5 is fed to the input of IC6, a LM324N balance amplifier. The purpose of this amplifier is two-fold: it acts as a balanced to unbalanced matching network, and its gain is set to 0.1 to act as a line-voltage attenuator. Capacitors C1 and C2 block the phone line's 48 volts DC from entering the amplifier. The ringing-voltage is limited by R1 and R2. The ratio of R3 to R1 sets the gain of IC3-d to 0.1. Resistor R4 biases IC3-d between its supply voltage and ground allowing it to operate from the single +6.0 volts DC power-supply line. The output of the balance amplifier passes through coupling capacitor C3 and is then decoded by IC1, a Motorola MC145436 dual-tone multi-frequency (DTMF) decoder IC.

The output of IC1 is a 4-bit word, whose codes are listed in Table 1. It is connected to IC2-b, a 4082 dual quad-input AND gate, so that the output of that IC (pin 13) is normally low, and goes high only when the "*" key is pressed. Therefore, when the "*" key is decoded by IC1, pins 1, 2, and 13 are high while pin 14 is low. To switch the output of IC2-b high, four logic-high inputs must be present. The high inputs are provided by IC1 pins 1, 2, and 13 and IC2-a pin 1.

In order for IC2-a's output to go high, it must also have four logic-high inputs. Two of those are provided by R7, D10, and C27. Those components ensure that the internal power supply is operating. That will prevent the unit from seizing the phone line if power is lost or removed while it is connected to the phone line. The remaining two inputs are provided by a logic high from IC1 pin 12, which is the DV, or DATA VALID, output pin. DV assures proper operation of IC1 by providing internal checks. When those checks are valid, DV will output a logic high. That prevents false triggering due to voice or other tones, such as music, that occur during normal telephone usage.

When the "*" key is depressed, IC2-b pin 13 goes high, which in turn charges C4 and turns on switching transistor Q1. That activates relay RY1. Diode D1 prevents DC voltage from bleeding back into IC2-b pin 13. The time-base oscillator for IC1 is formed from a 3.58-MHz crystal XTAL1 and R5.

The normally open contacts of RY1 close and D7, R9, RY2, R10, C5, LED1, transformer T1 (Sec), and the four diodes from the polarity bridge (D3-D6) are connected across the telephone line and effectively "seize" it. That combination of components is referred to as the seizure network. The unit is now in a "standby" mode and LED1 lights dimly. If jumper J1 is in the IN position and a station is tuned in on the FM tuner, that station will be heard on the telephone line. If J1 is in the OUT position, the station will not be heard until the phone is hung up.

RY1 will stay activated for approximately four seconds. That
Impress your callers by adding an FM music on-hold feature to your telephone.

delay is determined by the RC network of R6–C4. Diode D2 prevents relay-coil induction-induced "spikes" from appearing on the +12-volt DC power-supply line.

If the telephone is hung up within the four-second time-out period, additional loop current will flow through the seizure network and activate RY2. That causes normally open contacts of RY2 to close. The project is now in the "on-hold" mode. LED1 will be lit, and the selected radio station will be heard in the telephone line regardless of the position of jumper J1. If the telephone is not hung up within the four-second time-out period, RY1 will deactivate. The loop current flowing through RY2 keeps the seizure network across the telephone line and the unit remains "on-hold."

"normal" mode. LED1 will not be lit, and the caller will be disconnected if the telephone is hung up.

Latching push-button switch S1 is used to tune in the desired station. When it is in the IN position, the seizure network is placed across the telephone line and the output of the tuner is also connected (regardless of the status of J1). That allows you to hear the output of the FM tuner and adjust the station tuning and volume. (A feature of the receiver is the elimination of interstation "hiss," therefore no audio will be present until a station is tuned in.)

FM receiver circuit

At the heart of the receiver circuit is IC4, a TDA7000 Signetics FM receiver. This IC has a frequency-locked loop system with an intermediate frequency (IF) of 70 kHz. The IF can be chosen by active RC filters. The only function that needs tuning is the oscillator's resonant circuit, which selects the reception frequency.

The antenna is made up from the telephone line and the RJ11 cable. The RF signal travels through that path and is coupled via DC blocking capacitor C6 to the RF input bandpass filter. This broadband low-Q filter consists of C10, C11, and L1. Its primary purpose is to pass RF energy in the 88.0- to 108.0-MHz range while attenuating RF energy from above and below that frequency range. The bandpass filter serves to suppress potential interfering energy from outside the commercial FM broadcast band.

The bandpass filter also acts as a split-capacitor (also known as a tapped capacitor) input impedance-matching network to IC4. It matches a 75-ohm RF input impedance to IC4's 1.5K input impedance. The reverse RF input is decoupled by C12.

After the RF signal passes through the input bandpass filter, it goes to the input of the internal Gilbert cell mixer where it is mixed with the local oscillator (LO) signal. As mentioned earlier, the frequency of the LO is designed to produce an IF of 70 kHz. The tunable LO, connected between pins 5 and 6 of IC4, consists of tank components L2 and D9.

Varactor diode, D9, is DC-voltage tuned by the voltage-divider circuit consisting of R13, R18, and R12. The low end of the tuning range is set by R13 while the high end is set by R12. A high impedance path to the oscillator is provided by R11, keeping it from appearing on the DC tuning control voltage. C21 acts as an RF "short" to ground which prevents the oscillator's RF from entering D9. The IF output of the mixer is routed to a three-stage broadband low-Q IF filter network.

The first section (C20 and C19) determines the cut-off frequency for the second-order low-pass IF filter. The second section (C8 and C7) determines the upper and lower passband. The third section (C9) determines the passband of the third section of the low-pass filter network.
After the signal is passed through the IF filter section, it is demodulated. The quadrature detector is tuned by C14. The frequency-locked loop (FLL) filter, which suppresses IF harmonics and prevents them from appearing at the output of the demodulator, is controlled by C18.

The demodulated audio signal from pin 2 passes through a deemphasis network consisting of C22 and R14. A load for the audio output current source is also provided by R14.

The audio signal passes through C23 and R15 to the inverting input of audio amplifier IC3-c. Feedback resistor R19 controls the gain of the amplifier from 0 to 10. Transformer T1 matches the amplifier's output impedance to the telephone line impedance.

**Construction**

The author's prototype is shown in Fig. 3. The entire FM on-hold circuit is mounted on one double-sided PC board. The use of a single-sided board will work as long as the jumper wires are added to the top where necessary. We recommend that a PC board be used because of the VHF range involved in this project. We have provided foil patterns of the

**PARTS LIST**

- **All resistors are 1/4-watt, 5%.**
  - R1, R2, R11—100,000 ohms
  - R3, R4, R7, R13, R15—10,000 ohms
  - R5—1 Megohm
  - R6—39,000 ohms
  - R8—2000 ohms
  - R9—2700 ohms
  - R10—1200 ohms
  - R12—130,000 ohms
  - R14—20,000 ohms
  - R16, R17—470 ohms
  - R18, R19—100,000 ohms horizontal PC-mounted potentiometer

- **Capacitors. All are 50 volts DC, 10% tolerance, mono or ceramic disc unless otherwise indicated.**
  - C1, C2, C6—0.022 µF, 250 WVDC, 20% tolerance
  - C3, C13, C17, C23—0.1 µF
  - C4, C27—1 µF, 10 volts, 20% tantalum
  - C5—47 µF, 63 volts, 20% electrolytic
  - C7, C20, C21—3300 pF, 50 volts
  - C8, C14—330 pF
  - C9—150 pF
  - C10, C11—39 pF ceramic disc
  - C12, C22—2200 pF
  - C15—220 pF
  - C16, C18, C29—0.01 µF, 20%
  - C19—180 pF

- **Semiconductors**
  - D1—D7, D10, D11—1N4003, 1 amp 200 PIV rectifier diode
  - D8—not used
  - D9—MV209 varactor diode (Motorola) or ECG-604 LED1—Red LED
  - IC1—MC145436 DTMF decoder (Motorola)
  - IC2—4082 dual 4-Input AND gate
  - IC3—LM324N quad op-amp
  - IC4—TDA7000 FM Receiver (Signetics-Philips)
  - IC5—7812 +12-VDC, 1-amp regulator
  - IC6—78L05 +5-VDC, 0.1-amp regulator
  - Q1—MPSA13 NPN Darlington transistor

- **Other components**
  - L1—0.138 µH fixed inductor (Coilcraft no. 132-09 or equivalent)
  - L2—0.060 µH shielded variable inductor (Coilcraft no. 150-02J08S or TOKO no. MC122)
  - RY1, RY2—DPDT relay 12 VDC (Aroma
to no. DS2YE-S-DC12)

- **Miscellaneous:** Male power jack, female PC board-mounted lug receptacles, 117-VAC power pack (15 VDC at 150 mA), PC board, 6-foot modular line cord, male RJ11 to lugs, project case (Builder's Choice), and 3 14-pin IC sockets
FIG. 2—SCHEMATIC OF THE FM ON-HOLD unit. The output of IC1, a DTMF decoder, is a 4-bit word that controls the on-hold logic. The FM receiver, IC4, uses a frequency-locked loop system with a 70-kHz intermediate frequency, which is tuned by a tank circuit consisting of L2 and D9. Spurious reception is eliminated by a mute circuit in the IC.

Component side and solder side of the PC board if you wish to make it yourself. If you choose not to use a PC board, the use of a prototype style board is recommended. You should note that the use of wire wrapping will not work for the receiver portion of this project due to ground return path impedance problems. You can use IC sockets for all ICs except IC4, the TDA7000 FM receiver. The use of an IC socket at VHF frequencies should be avoided.

Figure 4 shows the parts-placement diagram of the unit. Before you begin construction, there are a few things to keep in mind:

- Use proper soldering techniques—The importance of proper soldering cannot be emphasized enough for VHF circuits. We recommend that the flux residue be removed from the completed PC board using a mild non-CFC cleaner that’s not harmful to plastics. Always read the manufacturer’s label.
- Static sensitive devices—Observe electrostatic discharge precautions when handling individual semiconductors as well as the completed circuit board.
- Component leads—Pre-form component leads before installing them in the board.
- Non-polarized capacitors—When installing these components, orient them so their values can easily be read. This will help if troubleshooting is needed later on.
- Resistors—Mount resistors so they can be read from left to right and top to bottom. This also aids in troubleshooting.
- T1—Bend the tabs flush against the PC board. The audio transformer has a “P” indicating the primary side. The primary mounts towards the outside of...
the board. If in doubt, the primary should measure about 500 ohms.

- C6—Mount vertically with the body in the hole closest to D4 and D6.
- L1, L2—It's important that the shield have a good electrical connection with the PC board mounting pads. Don't leave the soldering iron on too long as this plastic part might melt.
- IC4 (TDA7000)—When soldering this chip, be careful not to keep a hot soldering iron on the pins too long.
- LED1—For proper mounting height of the LED, cut two ½-inch pieces of insulating tubing. Insert the tubing over both leads. Install the LED with the flat side (short lead) toward T1.
- D9—Mount flush against the board. That will minimize any stray capacitance effects.
- IC sockets—Mount three 14-pin IC sockets (IC1-IC3) flush against the board. Orient the notch towards pin 1, which is indicated on the component side of the board.
- XTAL1—The leads of this crystal can be connected either way to the PC board. Mount it in the vertical position. Do not bend the leads where they exit the body.
- RY1, RY2—These relays are the same type, so they're interchangeable.

The following pre-test steps should be done after all components have been installed. Check that all components are mounted in their proper location. Verify polarized components are properly oriented and that all pads and connections have been properly soldered and de-fluxed. Once those steps have been completed, you can begin bench testing.

**Testing and alignment**

The only instrument needed to test the unit is a DMM. Connect the power pack (or a +15- to +28-volt DC power source) to the DC input. Connect AC power to the power pack. Don't connect the unit to the phone line at this time. Next, verify proper operation by making the check out measurements indicated in Table 2. After you have made those measurements, you can proceed with the alignment.

You'll need a plastic alignment tool, a signal source in the FM broadcast band, and a method to hear the audio output. The simplest way of aligning the unit is to connect it to the phone line. The unit was designed to not be sensitive to the tip and ring polarities. Therefore, it doesn't matter which phone lead connects to which terminal on the PC board.

Once the phone line is connected, dial your own number to eliminate the signal tone and offhook warning tone. Turn the receiver on by depressing push button switch S1. Set the tuning potentiometer to the extreme counter clockwise position (low end of the band). Note that due to the mute function, there is silence until a station is received.

Turn the volume control potentiometer ⅓ and ⅔ clockwise. Adjust the slug in L2 until the station operating at the lowest dial setting in your area is received with the loudest audio output. Use care when adjusting the slug as it is quite delicate and can easily be broken.

Next, set the tuning potentiometer to the extreme clockwise position (top end of the band). Tune back down towards the bottom end of the band (counter clockwise) until the station operating at the highest frequency is received. Tune through the entire range...
to verify all stations available to your area are being received. The receiver section was designed with a mute function built-in to allow only the strongest stations to be received. That makes tuning easier and suppresses images ("ghost" stations that appear in the wrong part of the tuning dial). Release the push-button and hang up the phone.

You can check for proper operation by having a friend call and be placed on hold by depressing the star "*" key (LED lights dimly) and then hanging up the phone.

**Installation and use**

A special feature of this project allows you to select when the music is present in the handset. Some telephone services (call waiting, call forwarding, voice mail) require the use of the "*" key. With J1 in the OUT position (circuit open), music will not be heard in the handset when the "*" key is depressed. It will, however, be heard by the caller when the phone is hung up. With J1 in the IN position (circuit closed), music will be heard every time the "*" key is depressed. Install the jumper according to your available service requirements.

If you would like to connect an external antenna or RF source, such as cable, to the tuner, you can connect it to the junction of C6, C10, and C11. It may be advantageous to disconnect the phone-line antenna by breaking the connection at C6.

It's easy to use the FM on-hold unit. To place a caller on hold press the star "*" key on any Touch Tone telephone. That places the unit in a standby mode and the LED lights dimly. The telephone must be hung up within four seconds for the caller to be placed on hold. When that's done, the LED lights brightly. If it's not hung up within 4 seconds, the unit resets itself and the LED goes out. The caller will be disconnected if the phone is hung up.

After a caller has been placed on hold, all you have to do is pick up the telephone to return to the conversation (any telephone connected to the line, Touch Tone or rotary). When the handset is picked up, the brightly lit LED will extinguish, the music will go off, and you will be connected to the caller.
The purist will tell you that if something's worth doing at all, it's worth doing well—and that's the case with the author of this story. The author always loved music, and was probably doomed to permanent audiophilia from day one. Even in 1960 at age 12, when he and his sister pooled resources to buy their first 45-RPM record (Pat Boone's "Love Letters in the Sand"), he recalls that, even on a monophonic, crystal-cartridge record player, there was an audible difference between the quality of a decent LP and the 45. Even though it should have sounded better than an LP, the 45 was bassier, noisier, and somewhat distorted, being pressed on what a broadcast engineer later poetically described as a blend of straw and chicken manure. Predictably, he soon became dissatisfied with the quality of the record player, and his junior-high and high-school years were marked by repeated attempts to upgrade the equipment without spending any money.

Along came the late '70s, and a new product was aimed at the purists: beastie cables (the name has been changed to protect the author!), which are expensive heavy-gauge speaker cables. The old purism surged forward, remembering the effects of cable resistance on damping factor and the effect of cable capacitance on high-frequency response—but the ads also spoke of skin effect, which has to do with the fact that at high frequencies, alternating currents tend to travel mainly at the outside surface, or skin, of a conductor. Since the skin effect essentially removes current from the center of a conductor, effectively reducing its cross-sectional area, it causes an increase in the impedance of a conductor at high frequencies.

Actually, when frequency is high enough, a tube or pipe will have the same effective resistance as a wire of the same diameter. That fact can be used simplistically to account for the use of waveguides rather than wiring at microwave frequencies. But concerning high frequencies, how high is high?

Supposedly, the skin effect becomes important above about 30 MHz, but a recent ad for speaker cables claimed perceptible benefits from reducing skin effect at 20 kHz. Soon after, articles ap-
peared in professional trade journals mentioning the likes of Lucasfilm using the enormous cables, so some real research was in order.

The goal was to find out whether beastie cables did in fact:
- Reduce the amount of power lost in the cable enough to provide a significant improvement in efficiency,
- Increase the damping factor of the speaker/amplifier system enough to provide audible improvement,
- Provide any significant benefit in the frequency response of the system.

No other benefits are claimed for these cables, so it is not necessary to look for undiscovered or presently unmeasurable effects.

The problem was attacked both analytically and experimentally. The equivalent circuit of a real loudspeaker driven by a real amplifier through real cables is shown in Fig. 1. Any effects produced by the cables must show up in the cable resistance, capacitance, or inductance. The efficiency and damping-factor questions depend almost exclusively upon the cable resistance, whereas the frequency-response question is mainly a function of the capacitance. Wire inductance is so small compared with the semi-inductive nature of speaker impedance at high frequencies that it can be ignored, as we will see.

The cable resistance is made up of three components: the contact resistance, the ohmic resistance of the wire, and any contribution from skin effect. The ohmic resistance can easily be found from wire tables in most electronics reference books. Table 1 shows the resistance of a representative sampling of copper cables, listed according to gauge. For years, selection of cable gauge has been made according to the criterion of 10% loss. In other words, for a given cable length, what resistance will give no more than 10% (0.46 dB) power loss at the speaker? Figure 2 shows the calculations involved in determining that value. For short cable runs, the resulting gauge is surprisingly small.

About fifteen years ago when the author was an audio consultant, he would specify 18-gauge cable for amplifiers up to 100 watts feeding impedances of 8 ohms or more with runs of 25 feet or less. For each halving of impedance or doubling of amplifier power or distance, the wire size would increase by two gauges; 16 gauge for 100 watts into 8 ohms at 50 feet or 4 ohms at 25 feet, etc. That rule of thumb includes a safety factor so that the loss will always be less than 10%.

The National Electrical Code specifies cable gauges based upon safety considerations; if a wire carries too much current over a long enough period of time, it can become dangerously hot and start a fire. Going back to the rule of thumb, a speaker with an average impedance of 8 ohms fed by a 100-watt amplifier will draw about 3.5 amperes at full power. However, even running at full tilt, it's unlikely that the average power will be greater than one-third of your amplifier's maximum, so the rule of thumb provides a large safety margin from a fire-prevention standpoint.

The damping factor can be defined as the ratio of a speaker's impedance to the total resistance in series with the speaker. Since the simple loss calculation in Fig. 2 depends upon the combined resistance of the speaker and the cable, the resistive power loss will be related to the damping factor. Thus we can find a relationship between damping factor and low-frequency loss. In the Audio Cyclopedia Howard Tremaine established that there is no value in trying for a damping factor greater than 20. That is based on the fact that the speaker's voice-coil resistance appears in the circuit, and its value—typically 6 to 7.5 ohms for an 8-ohm speaker—sets a practical limit on the benefits of reducing other resistances. The effective damping factor is equal to:

$$Z_{SPEAKER} = \frac{R_{VOICE COIL} + R_{AMP} + R_{CABLE}}{R_{SPEAKER}}$$

A stated amplifier damping factor of 20 would represent a total resistance of 8 ohms divided by 20, or 0.4 ohms in series with the amplifier. That would give an effective damping factor of:

$$\frac{8\Omega}{6\Omega + 0.4\Omega} = 1.25$$

assuming a 6-ohm voice-coil resistance. With most amplifiers having output impedances on the order of 0.1 ohm or less, this would mean that the cable resistance could be 0.3 ohms. The loss in dB corresponding to an 8-ohm speaker fed through a 0.3-ohm cable is:

$$= 20\log\left(\frac{8\Omega}{8\Omega + 0.3\Omega}\right) = -0.32dB$$

That means that for an optimum effective damping factor, the resistive cable loss should be...
less than 0.32 dB. Just for comparison purposes, a 1-dB cable loss, which would result from a 0.9-ohm cable resistance, would result in an effective damping factor of 1.14, which is not much lower than 1.25.

As mentioned earlier, skin effect increases the effective impedance of a wire, and can be best explained by looking at Fig. 3. The skin depth of a conductor is the distance into that conductor, measured from the outside surface, at which current density is 1/e times that at the surface. (The symbol e stands for the base of natural logarithms, and equals approximately 2.72.) For a direct current, the current density (amperes per unit cross-sectional area) is the same throughout the wire. For AC, the current density is less at the center of the wire and greater at the surface.

At low frequencies, the skin depth (which depends on characteristics of the bulk conductor material) is usually greater than the radius of the conductor, which means that for all practical purposes the current density is the same throughout the conductor. Larger-diameter conductors can exhibit measurable skin effect at relatively low frequencies, including audio frequencies.

The simplest indicator of skin effect is the ratio \( \frac{R_{AC}}{R_{DC}} \), where \( R_{AC} \) is the resistance per unit length of a wire to alternating current of a certain frequency and \( R_{DC} \) is the ohmic resistance per unit length. As long as \( \frac{R_{AC}}{R_{DC}} \) equals 1, skin effect is negligible. When \( \frac{R_{AC}}{R_{DC}} \) rises significantly above 1, skin effect may begin to matter. We say may, because it only matters if the total resulting increase in cable resistance causes a perceptible effect in the reproduction. For a frequency of 15 kHz, \( \frac{R_{AC}}{R_{DC}} \) equals 1.1 when a 15-gauge solid wire is used. Larger wires will exhibit a greater proportional increase in resistance as frequency increases. Of course, since the resistance of large wires is lower to begin with, the actual change in measured resistance may or may not matter.

Stranded wire is extremely difficult to analyze. Naturally, each strand has a certain surface area, so that all the strands connected in parallel would have a very large surface area. In actuality, much of the surfaces of the individual wires are in contact with each other, making the actual effective surface area virtually impossible to determine—unless the individual strands are insulated from each other, as in litz wire. At any rate, we can use solid wire as a worst case to analyze, knowing that we'll really be using stranded wire that has less skin effect.

The actual resistance, capacitance, and inductance of a cable are distributed evenly along its length. Telephone engineers found out long ago that, for analysis purposes, a cable's R, C, and L can be lumped into a single component if certain conditions are met. The conditions depend upon the attenuation constant and length of the cable. The attenuation constant (\( \alpha \)) is given by:

\[
\alpha = \sqrt{\frac{R^2 + \omega^2 L^2}{C^2}} + \frac{R G}{C^2}
\]

where R, L, C, and G are the cable's resistance, capacitance, inductance, and leakage conductance per unit length, and \( \omega \) is the angular frequency, or \( 2\pi f \).

The author does not like to lie awake nights solving equations like that, and tables of attenuation constant versus frequency are not generally available for the kinds of cables used for speaker leads. However, tables for 19-gauge pulp-insulated telephone cable indicate that a 3-kilometer cable section can be analyzed using the lumped-constant

<table>
<thead>
<tr>
<th>Cable Type</th>
<th>C (pF/ft)</th>
<th>L (µH/ft)</th>
<th>R* (ohms/ft)</th>
</tr>
</thead>
<tbody>
<tr>
<td>22-ga. cheap</td>
<td>10.7</td>
<td>0.29</td>
<td>0.0178</td>
</tr>
<tr>
<td>18-ga. zip</td>
<td>14.0</td>
<td>0.28</td>
<td>0.007</td>
</tr>
<tr>
<td>20-ga. twist</td>
<td>18.0</td>
<td>0.36</td>
<td>0.0107</td>
</tr>
<tr>
<td>4-ga. cable</td>
<td>50.8</td>
<td>0.29</td>
<td>0.0007</td>
</tr>
<tr>
<td>16-ga. &quot;drop cord&quot;</td>
<td>22.4</td>
<td>0.36</td>
<td>0.006</td>
</tr>
<tr>
<td>shielded &quot;guitar cord&quot;</td>
<td>105.8</td>
<td>0.30</td>
<td>0.048</td>
</tr>
<tr>
<td>16-ga. zip</td>
<td>12.5</td>
<td>0.23</td>
<td>0.0127</td>
</tr>
</tbody>
</table>

*One-way resistance, not loop resistance; that includes the contact resistance of the terminations.

**FIG. 3**—THE SKIN EFFECT increases the effective resistance of a wire. The skin depth is the distance into the conductor at which the current density is approximately 1/e of that at the surface.

**FIG. 4**—A COMPUTER SOLUTION, or prediction, of the model in Fig. 1 yielded these results. The worst-case loss is well under 1 dB at 20 kHz.
method at 1 kHz with a total attenuation under 1 dB and a phase accuracy within 5 degrees. Although it may not be immediately obvious to the casual observer, attenuation constant is proportional to the square root of frequency, so that would mean that the same accuracy could be expected at 20 kHz if the length were reduced by \( \sqrt{20 \text{ kHz}/1 \text{ kHz}} \) that works out to about 2100 feet. Since we rarely extend speaker cables anywhere near that far, we can safely use the lumped-constant method with no qualms. That's what was assumed in Fig. 1.

Table 2 shows the types of cables chosen for the analysis, along with their measured resistance, capacitance, and inductance. The values were measured using a Hewlett-Packard 4261A LCR meter and a test frequency of 1 kHz. Instead of a speaker, a resistance of 7.9 ohms and an inductance of 6.3 microhenries were used in the calculations. Instead of "real" beastie cables, we used ones that were on hand, including a very large (4-gauge) stranded cable. If those cables showed no measurable detrimental effects on efficiency, damping-factor, or frequency response, then the alleged beastie benefits would turn out to be solutions to a nonexistent problem!

A computer solution of the circuit of Fig. 1 yielded the results plotted in Fig. 4. A 10-foot length was assumed for each cable, and it included the effects of cable capacitance and inductance, but not the skin effect. Notice that the worst-case loss was well under 1 dB at 20 kHz.

Computer solutions without experimental verification are not always trustworthy, so the actual response of the cables was measured on the setup shown in Fig. 5. A 10-foot length was assumed for each cable, and it included the effects of cable capacitance and inductance, but not the skin effect. Notice that the worst-case loss was well under 1 dB at 20 kHz.
significant skin effect, and large ones will, but even a large percentage change in a large wire’s small resistance is of little consequence. Trying the $R_{ac}/R_{dc}$ values for various cable diameters in conjunction with the computer analysis, 12-gauge solid wire is found to have about the worst skin effect of any cable.

Therefore, if any type of speaker cable could cause frequency-response problems, the high capacitance and the skin effect of 12-gauge Romex should make it the ideal bad example. Another run was made using a 40-foot length of Romex, both into a dummy load and into the test speaker. In the graph of Fig. 8, we see a hefty \( \frac{1}{2}\)-dB droop at 20 kHz, compared to the response at 20 Hz. The overall signal loss and damping-factor degradation are less than those of the smaller cables that are shown in Figure 8, due to the lower resistance of 12-gauge cable.

Since that still wasn’t significant, a computer simulation of 100 feet of 12-gauge Romex was performed, with response run clear out to 50 kHz. The results are shown in Fig. 9. Here, at last, is something the beastie people can sink their teeth into! Anyone who can hear 50 kHz will find a full 4.5-dB drop resulting from the use of 100 feet of 12-gauge Romex—providing, of course, they’re using an amplifier and speaker that can reproduce it. Of course, the skin effect is still only about half a dB, and effective damping is not degraded, so maybe they’d better drop those points from their ads.

The results of that rather involved bit of research clearly indicate that ordinary speaker cables, including the ones that any knowledgeable audiophile would sneer at, do not significantly degrade frequency response. They vindicate the rule-of-thumb advice (18-gauge for 100 watts, 25 feet, into 8 ohms) except for a slight degradation in damping factor; 1.21 with a 25-foot cable run. For optimum damping factor, that rule should be changed to 18-gauge for 100 watts, 20 feet, into 8 ohms. Also, only 20-gauge, 22-gauge, and guitar-cord cables are a serious detriment to damping factor.
THE TECHNOLOGY TO PRODUCE low-cost digital clocks has existed for years. Unfortunately, the style in which these clocks have displayed the time has been mostly limited to four digits (representing hours and minutes). Few digital clocks, if any, have fully taken advantage of the great capabilities of today's microprocessors to provide a more novel display... that is until the HyperClock.

The HyperClock has a custom-programmed microcontroller that generates all the signals necessary to display time in eight eye-catching (yet easily readable) modes. Among its features are the ability to simulate a sweep second hand with a ring of 60 LEDs; it can graphically display the level of ocean tides; it has a "fading" display mode that causes the LEDs to gradually change when updated; it has a hourly chime/alarm output; supports 50- or 60-Hz powerline operation; it has a battery backup; and an intelligent date display that knows the last day of each month. Let's take a look at those features in greater depth.

The digital display

In four of the HyperClock's eight modes it can display time with a clever twist that those of you that grew up before the "digital-clock revolution" will appreciate. You may recall that it was fairly common for folks to speak of how many minutes it was till the next hour. For example, people would say "it's ten before five" rather than "it's four-fifty." Well, the HyperClock can actually "display" the time in that manner.

To help explain how that is done, take a look at Fig. 1-a; it shows the six seven-segment LED displays that are used to form the clock's digital display. During the first 30 minutes of each hour, the two middle digits signify the hour and the rightmost digits indicate minutes past the hour (like most digital clocks), and the two digits on the left are blank. Figure 1-b shows how eleven fourteen (or literally "fourteen after eleven") would look in this mode.

However, once the clock advances to beyond half-past the hour, the right digits go blank, the middle display would increment to display the next hour, and the left two digits would indicate how many minutes are left until that hour. For example, if it was thirteen minutes to twelve (11:47), the display would look like Fig. 1-c. You could read that literally as "thirteen before twelve."

On the hour, the clock just displays the hour, so the left and right minute displays remain blank (see the display for 12 o'clock shown in Fig. 1-d).

As mentioned earlier, there are 60 LEDs arranged in a circle to display seconds or an approximation of the current tide level in your locale. The LEDs act as a light chaser sweeping through the seconds of a minute. They can also show the relative tide level by moving from the 12 o'clock position at high tide to.
the 6 o'clock position at low tide. The timing of the tide-indicator mode is set to display two complete tide cycles in 24.51 hours just as it should. (For more on this subject see the sidebar entitled "A Bit About Tides.")

Fading-out the digits

The HyperClock also differs from other clocks in how it updates its display. In typical clocks, the display digits abruptly change as time passes. However, in four of HyperClock's eight modes when a digit must be updated, the LED segments representing its old value are dimmed as the segments for the new value become brighter. Likewise, each LED in the light chaser fades off rather than turns off. This animation is rather relaxing to watch.

These display effects are accomplished by a mixture of multiplexing and duty-cycle modulation. The seven-segment displays are all common-anode types, so each digit has its own anode-driver transistor. In typical multiplexed-display fashion, the cathodes of corresponding segments of each digit are connected together and share a common driver.

Like any other multiplexed display when the segment data for a digit is placed on the segment data lines, the anode driver for that digit is activated, and the seven-segment display lights to exhibit the appropriate digit. All the other digits are off at this time. After a short period of time (1.83 ms) the segments are turned off via the segment lines and the anode driver is deactivated. The anode driver for the next digit is then activated, the segment data for that digit is placed on the segment lines, and the process continues until all the digits have been lit. Since the human eye is too slow to see the digits turn on and off, it appears as though they are all on simultaneously. Multiplexing the digits in this fashion reduces the number of pins on the microcontroller needed to control the LED displays.

HyperClock's fading effect is created by modulating the duty cycle of the segment enable signals. When a seven-segment display is enabled, the segment data lines spend part of the time in states corresponding to the current digit to be displayed, and the rest of the time in states corresponding to the next or "future" digit that will be displayed. Each time the digit is enabled, the duty cycle will favor the future-digit data more and more, until only the new digit is displayed, and the process repeats each time the display must be updated.

The microcontroller

At the heart of the HyperClock is an Intel 8749 microcontroller. It is programmed to perform a variety of functions, namely: display multiplexing, time-keeping, receiving switch input, coordinating the hour and alarm chime, and initiating a power-fail mode that permits the time-keeping functions to continue while blanking the display to conserve backup-battery power.

The 8749 has 2K of EPROM, 128 bytes of RAM, 24 I/O pins, a programmable 8-bit timer, and an interrupt-control structure. The custom-program placed in the processor to create HyperClock's special effects extensively exercises all of the chip's features; Fig. 2 contains a simplified flow chart for the HyperClock program. The program is shown divided into two sections: an interrupt routine and a main loop.

The interrupt routine is mainly responsible for taking the segment data from the segment-data buffer in the microcontroller and placing it on the display-control lines in a multiplexed fashion. The interrupt that initiates the routine comes from the 8749's internal timer, which has been programmed to execute the interrupt every 1.83
ms. Using an interrupt-program segment in this way allows the display's fading effect to appear gradual because the process of updating the display occurs at regular intervals. This routine is also responsible for checking a powerline-frequency input on the microcontroller to determine if a powerline cycle has passed. If so, it informs the main loop of the program.

The main loop keeps track of the number of cycles that pass so it knows when to update the segment-data buffer or initiate the alarm or chime. If no powerline cycles are detected, the main loop assumes AC power has been terminated and puts the clock in power-fail mode. In that mode it shuts off the display and allows the microcontroller to “invisibly” keep track of time via a 6-MHz crystal. The main loop also processes input from the clock's switches (we’ll talk more about that later).

For all that goes on inside the microcontroller, the functions assigned to its pins by the HyperClock program (see Fig. 3) are relatively easy to understand. Let's take them one group at a time.

The pins labeled A through G and DP (pins 12–19) in Fig. 3 are the outputs for the display-segment data. They indirectly control the cathode drivers for the multiplexed display. Similarly, the outputs labeled DEO–DE4 (pins 21–24 and 35) control the anode drivers for the display via demultiplexer chips, which we’ll discuss later.

The pins labeled S1–S7 (pins 27–34) are used as function-switch inputs. The switches connected to those inputs (DISPLAY MODE, DISPLAY DATE, DISPLAY ALARM, INCREMENT HOUR/MONTH/MODE, INCREMENT MINUTE/DAY, SNOOZE/TIDE ADVANCE, and ALARM TOGGLE, respectively) activate various chip functions by grounding those pins. A complete explanation of the switches’ functions will be presented later.

Low-going pulses from the pin labeled CHIME (pin 34) activate the chime circuit, which is composed of discrete components. The microcontroller triggers the chime circuitry with a 12.8-ms wide low-going pulse at one-second intervals. When not in alarm mode, the chime signals the hour by chiming an appropriate number of times, and signals each half-hour by chiming once.

When the microcontroller is in the “alarm” mode, the chime circuitry is used as an alarm annunciator. In this mode the clock does not chime on the hour and half hour. That permits you to use the HyperClock as an alarm clock without the chimes disturbing you until the appointed time.
The line input (pin 6) is used by the IC to accept a 50- or 60-Hz square wave. The square wave is used for time keeping and to detect AC-power failure as mentioned earlier. The 50Hz/60Hz input (pin 1), tells the 8749 what frequency to expect at the line input. If pin 6 is low, the 8749 assumes the line signal is at 60 Hz, but if that pin is high the 8749 prepares for 50-Hz operation.

As its name implies, the reset input (pin 4) initializes the microcontroller. A low at that input will erase all modes and settings previously entered.

The xi and x2 inputs (pins 2 and 3) need to be connected to the 6-MHz crystal, XTL1, mentioned previously. Last, but certainly not least, are the 5-volt power inputs (pins 5, 26, and 40) and the grounds (pins 7 and 20). Those inputs of course, are self explanatory.

**The display circuit**

While the microcontroller does a great deal, the HyperClock requires some additional circuitry to make it a complete timepiece. For example, the microcontroller cannot provide nearly enough current to drive the LED display. For that reason additional anode and cathode drivers were included in the design. They are shown along with the other display components in Fig. 4.

The control signals for the LED cathodes originate from the A-G and DP pins of the microcontroller. The A-G signals are sent to a ULN2003 buffer/driver (IC7) which contains seven high-current drivers. Each output is capable of providing 500 mA of peak drive current. Since the ULN2003 contains only seven of the eight cathode drivers required, a Darlington driver was made out of two 2N2222 transistors to drive the DP line.

The DE0 through DE4 outputs generated by the microcontroller are decoded by two 74LS145 decoder/driver IC's (IC5 and IC6). Only one decoder output is driven low at any time. Each 74LS145 output supplies current to a 2N2907 drive transistor that sources current for the anodes of a group of LEDs or a display digit.

Note that there are additional LEDs to provide an AM/PM in-
dication (LED61), tell you if the alarm-mode is active (LED62), and provide a winking second display (LED63 to LED65). That's all there is to the display circuit.

The remainder

The most noteworthy of the clock's remaining circuitry (shown in Fig. 5) is the chime circuit. When the microcontroller generates a low-going pulse on pin 34, it activates Q19. That transistor then provides sufficient current to drive Q1 into saturation.

With Q1 on, the negative side of C13 is effectively grounded, which causes it to charge. When Q1 is turned off, C13 discharges through a 470K resistor (R13). The resistor/capacitor combination has a time constant of 0.47 seconds. The exponentially decaying signal produced by the discharge is buffered through a unity-gain amplifier (IC3-a) to a 1N914 diode (D3).

The cathode of D3 is connected to the output of a 50% duty cycle 5-kHz square-wave oscillator consisting of IC2-a, R7, C10, and Q2. Transistor Q2 serves to provide a dynamic pull-up for that oscillator since the LM393 is an open-collector type comparator. The 5-kHz
square wave present at the emitter of Q2 is clamped in amplitude by the buffered exponential waveform from IC3-a, so the 5-kHz signal decays in amplitude in step with the discharge of C13.

The decaying 5-kHz signal is fed through C14 to a second-order low-pass filter tuned to approximately 5 kHz. The filter is composed of a LM324 op-amp (IC3-d), C11, C12, C14, R10, and R11. It removes the high-frequency components contained in the decaying square wave to smooth it out. From there the signal is passed to two more op-amps (IC3-b and IC3-c) that form a push-pull amplifier, which provides the piezo transducer with a 10-volt peak-to-peak drive signal.

While the output signal is not exactly a pure sine wave, the audible result sounds pretty much like a small bell. If you feel the chime is too loud, you can eliminate half of the push-pull amplifier by jumpering one side of the transducer to the 5-volt supply, which is available via JU2.

The entire circuit receives power from a 9-VAC wall-mount transformer. The 9-VAC supply is fed to a full-wave bridge rectifier and filtered by C3 to act as an unregulated 12-volt DC supply. The 12-volt supply powers the display circuitry: op-amps, the comparators, and an LM340-5 (IC1) 5-volt regulator. The regulator in turn powers the 5-volt supply line.

If AC power is interrupted, a 9-volt battery connected to J1 sources current to the regulator to keep the HyperClock functioning. If you plan to unplug the clock for any length of time, the battery should be disconnected to conserve its life.

A BIT ABOUT TIDES
Predicting the tides in any given locale is not a simple job. Tides are affected by many cyclic astronomical forces: the declination in the orbits of the moon and sun relative to a point on the Earth, and the local geography of the coast line in the area in which you live, to name a few. HyperClock predicts the tidal levels from the most predominate of these forces, the moon. The moon requires 29.53 days to orbit Earth, and that combined with the 24-hour solar day causes a high and low tide every 12 hour and 25.5 minute interval. HyperClock tracks the moons primary affect on the tide. You can find out about the level of local tides in your area from your newspaper or library. The information can be used to initially set the tide indication on your HyperClock to a low or high point. From there the graphic display will be an aid in the determination of the relative level of the tides in your locality.

The line input (pin 6) of the microcontroller cannot be driven directly from the 12-VAC wall transformer. So the transformer signal is conditioned by a Schmitt-trigger circuit to generate a suitable square wave. First the signals amplitude is reduced by a voltage divider consisting of R1 and R58, and its positive excursions are limited to about 5 volts by D1. The limited signal is then sent to the inverting input of the LM393 comparator. Positive feedback is applied to the comparator’s non-inverting input by R4 to prevent it from generating false signals. The comparator drives the microcontroller’s line input with the resulting square wave.

Construction
In order to build a HyperClock, you’ll need a programmed 8749 microcontroller. A preprogrammed and tested microcontroller is available from the supplier mentioned in the parts list. The executable code to program an 8749 is available from the RE-BBS (516-293-2283, 1200/2400, 8N1), as a file called HYPER.HEX. The file is supplied in Intel's Hex format, which is directly compatible with most EPROM program-
FIG. 6—DISPLAY-SIDE PARTS-PLACEMENT diagram can be used to locate most of the HyperClock’s components.

mers. The software is also available on floppy disk from the source mentioned in the parts list.

A 6.5- x 6.5-inch octagon-shaped PC board is also available from the supplier to help you assemble a HyperClock of your own. If you wish, you can make your own double-sided printed-circuit board from the foil patterns included in this article or using artwork from the supplier listed in the parts list. Of course, you could even use a point-to-point wiring technique, so we’ll discuss that briefly later on.

Figures 6 and 7 show the parts-placement diagrams for the HyperClock (6 shows the display side and 7 shows the solder side) for those of you that will use a PC board. All components, except for the wall-mount transformer, are shown mounted on the circuit board. Note that the switches can be placed on either side of the board, depending on the cabinet you wish to place the clock in. There are some additional connector pads on the board so you can run wires to the switches should your cabinet design require that they be located off the PC board. Figure 8 shows a completed HyperClock board.

The design readily lends itself to many different project cases. However, make sure that the cabinet you choose has some openings in the back to allow a little cool air to flow around the clock’s 5-volt regulator and heat sink. If you like the case used for the prototype, you can build one out of a length of wood molding as the author did.
The PC board was designed to work with many different dual-digit displays, so you don't necessarily have to restrict yourself to the Panasonic units mentioned in the parts list. Just make sure that whatever you use is a pin-for-pin same-size replacement. If you do use the recommended units, be sure to raise the two minute displays up from the board so that their viewing surfaces are flush with that of the hour display.

Lastly, the PC board provides some holes for wire ties to hold the 9-volt battery and the wall-mount transformer leads. You should take advantage of them. Remember to install the 9-volt backup battery and connect JU1 and JU2 to select 50- or 60-Hz operation and the volume of the chime, respectively.

When you connect the clock to power it should come up at 12:30 AM and will be ready to accept the current time, alarm time, date, and the tide level if desired. If you run into any difficulty, you may find some of the troubleshooting tips provided in the following section useful. However, if all is well, you can proceed to the "operating" section to prepare the clock for use.

**Point-to-point wiring**

Working with perforated construction board and point-to-point wiring gives you the freedom to design your own display layout. One nice alternate design would be to place the 60 LED's in the form of a full cycle of a sine wave, especially if you set the clock to display the tide level.

An early prototype of the clock was built using wire wrap. Regardless of the wiring technique, you should invest in a large enough piece of perforated construction board (at least 6 x 6 inches) so that you can make your custom display with plenty of room to spare for all the support electronics.

Furthermore, when you are shopping for the hour and minute displays, select minute displays that are somehow distinctly different than the hour display. That will make the display more easily readable in the "minutes before the hour" mode.

It is also suggested that you don't place the display components on the same side of the board as the heatsink/regulator assembly, C3, and the 9-volt battery. Doing so would increase the profile of the display side of the board, forcing you to place the LED lens at an undesirable distance from the display components.

Aside from those suggestions, layout is not critical, so if you observe good construction and wiring techniques you should not have any problem getting the clock to function. However, if you should encounter some difficulty, the following hints ought to help:

- If no LED's are illuminated when you apply power, first check the unregulated supply for a minimum of 10 volts DC. While only 2 to 3 volts of overhead are required to operate the 5-volt regulator, at least 10 volts is required to drive the LED's sufficiently.
- If the power-supply circuitry is okay, check the wiring around the LM393 (IC2) from the bridge rectifier and going to IC4 pin 6; the microcontroller looks at that line and if there is no line frequency at that input, the clock will shut down the LED drivers (as we explained earlier).

Since most of the wiring in the clock runs between the LED drivers and the LED's, it is likely that you could have made an error in one or more of those connections. If you observe that any active LED segments do not form numbers, then you should check the connections from IC4 to IC7 and the corresponding connections to the cathodes of the LED displays. A mistake between IC4 to IC5 and IC6 will make the displayed digits and/or seconds appear out of order. An error in wiring from the outputs of the 74LS145's to the...
FIG. 8—A FINISHED HYPERCLOCK. The PC board makes assembly neat and straightforward.

Table 1—Modes and Their Features

<table>
<thead>
<tr>
<th>Feature</th>
<th>Mode</th>
</tr>
</thead>
<tbody>
<tr>
<td>Fading Updates</td>
<td>0</td>
</tr>
<tr>
<td>Plain Updates</td>
<td>1, 2</td>
</tr>
<tr>
<td>Before-The-Hour Display</td>
<td>3</td>
</tr>
<tr>
<td>Plain Display</td>
<td>4, 5</td>
</tr>
<tr>
<td>Tide Light Chaser</td>
<td>6</td>
</tr>
<tr>
<td>Seconds Light Chaser</td>
<td>7</td>
</tr>
</tbody>
</table>

The functions performed by S4 and S5 in the HyperClock depend on the state of the three display switches (S1 through S3). If none of the display switches are depressed, pressing S4 advances the current hour displayed and pressing S5 advances the minute display. That is how you set the time.

Pressing and holding down S1 (the Display-Mode button) causes the clock to display the number of the current operating mode, which can be altered by pressing S4 (S5 will do nothing). The various operating modes and the features they support are listed in Table 1.

If you depress and hold switch S2 (the Display-Date button) the current month and day are displayed. With that switch depressed, the month and day can be advanced by pressing S4 and S5, respectively.

Activating switch S3 will cause the clock to display the time the alarm is set for. By pressing S3 along with S4 you can alter the hour setting, and by pressing S3 and S5 simultaneously you can change the minute setting.

Moving on to the last two switches, the Alarm Toggle switch (S7) determines the state of the alarm and the hourly chime of the clock. For example, on power-up the hour chime is enabled and the alarm is disabled. If S7 is pressed once, the alarm is turned on and the hour chime is disabled. Depressing the switch again will turn off both the alarm and the hourly chime.

The Snooze/Tide Switch (S6) has a dual purpose. If the HyperClock's alarm was armed and goes off, that button will silence the alarm for an additional ten minutes. You can forestall the alarm in this way as many times as you like. If the alarm is disabled, and the clock is in a mode that supports the tide-level display (modes 4 through 7), pressing S6 will advance the tide indication on the 60-LED display to set its position.

Driver transistors, which connect to the LED anodes, will cause the same effect.

Operation

The functions performed by S4 and S5 in the HyperClock depend on the state of the three display switches (S1 through S3). If none of the display switches are depressed, pressing S4 advances the current hour displayed and pressing S5 advances the minute display. That is how you set the time.

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LINE POWER FROM 12 VOLTS

Build a 40-watt DC-to-AC inverter, and power AC appliances from your automobile cigarette lighter.

WOULD YOU BELIEVE THAT THIS ARTICLE was written on an electric typewriter while the author was sitting next to a stream on a camping trip? The typewriter was powered from our 40-watt inverter that can be plugged into an automobile’s cigarette lighter socket. The unit has enough power for many items that normally don’t go on camping trips, such as a TV, a stereo, an electric razor, or a desk lamp. However, it also has some uses that may not be as obvious; it can be used to power items such as an oscilloscope or soldering iron when doing electronics work in the field. On road trips, the inverter can be used to power a camcorder battery charger.

The inverter draws a maximum of 5 amps, which is completely safe for an automobile cigarette lighter socket, and the no-load current is only half an amp. The output voltage is regulated and remains fairly constant from no-load to full-load. Figure 1 shows the output-voltage waveform superimposed over a sine wave. The rectangular output waveform has the same RMS and peak voltage as the sine wave, so the device being powered will never know the difference. The rectangle-wave operation greatly increases efficiency. The waveform would look similar if displayed on an oscilloscope.

Operation

The inverter, the schematic of which is shown in Fig. 2, is actually a push-pull audio amplifier. The “input,” or reference signal, is a 5-volt square wave. The output is 340-volt peak-to-peak AC signal. The feedback signal is rectified in order to match the DC reference signal. On one half of the AC waveform, the upper three FETs are gated on, and on the other half the lower three FETs are on.

Normally, 120-volt AC outlets have one side at ground and one side that’s “hot.” The hot side alternates from -170 to +170 volts. The inverter output is a little different. On one half of the AC cycle, one side is near ground and the other is at +170 volts. During the other half of the cycle the situation is reversed.

Op-amp IC1-a and its associated components form a 300-Hz clock oscillator, and counter IC2 divides the clock signal by four to obtain a 75-Hz inverter frequency. The 75 Hz, rather than 60 Hz, is used to avoid transformer saturation. Some electric clocks will run fast with that frequency, but most electronic gear will work just fine. Decade counter IC2 controls the timing of the reference signal and the gating-on of the error-amp signal to the proper set of FETs.

Figure 3 shows the timing relationships in the inverter. When IC2 pin 3 goes high, the output of buffer IC1-c is high. That reverse biases D1 and allows the error amp signal to reach Q1, Q2, and Q3. At the same time, IC2 pin 4 is low, which causes the output of buffer IC1-d to be low. That grounds the gates of Q4, Q5, and Q6 thereby turning them off. Pins 2 and 7 of IC2 are also low, so Q7 is off. A 5-volt reference from regulator IC3 is now present at the error-amp’s (IC1-b) non-inverting input. The reference-signal rise time is slowed by R12 and C2 in order to avoid output overshoot, and the gain and frequency response of the error amp is set by R15, R25, and C3.

Next, pin 2 of IC2 goes high, which turns Q7 on and the reference signal is pulled to ground. Pins 3 and 4 of IC2 are now low.
and the FET gates are grounded, turning them off. Pin 4 of IC2 now goes high and the other three FET's are gated on. The reference signal now rises to 5 volts, and the other half of the AC output waveform is generated. The next clock pulse causes IC2 pin 7 to go high; all FET's are now off and the reference is set to zero. The following clock pulse resets IC2 and another cycle begins.

A filter that protects the CMOS circuitry against alternator spikes and reversed input polarity is formed by R7, C8, and D7. Components R9 and C4 filter output spikes, and R18–R21 are pre-load resistors to stabilize the inverter when no load is connected. Although the FET's have no current-equalizing source resistors, they still share current fairly equally. (When a FET "hogs" current it heats up more and its on resistance increases, causing it to draw less current.)

Construction

The inverter circuit was built on a perforated construction board. Transistors Q1, Q2, and Q3 share a 1.5- by 4-inch heatsink, and Q4, Q5, and Q6 share another; the heatsinks are made of aluminum sheet. Figure 4 shows an internal view of the inverter. In the prototype, the FET's were not insulated from the heatsinks because the heatsinks are isolated from ground and all other circuitry. If you use any other heatsinking configuration, the FET's should be insulated.

PARTS LIST

All resistors are 1/4-watt, 5%, unless otherwise noted.

- R1–R7–100 ohms
- R8–1000 ohms
- R9–1000 ohms, 1/2-watt
- R10, R11–4700 ohms
- R12–R16–10,000 ohms
- R17–10,000-ohm potentiometer
- R18–R21–22,000 ohms, 1/2-watt
- R22–R26–100,000 ohms
- R27, R28–470,000 ohms
- R29–1 megohm

Capacitors

- C1–0.001 µF, ceramic disc
- C2–0.01 µF, ceramic disc
- C3–0.0047 µF, ceramic disc
- C4–0.05 µF, 200 volts, ceramic disc or metal film
- C5–C7–0.1 µF, ceramic disc
- C8, C9–470 µF, 35 volts, electrolytic

Semiconductors

- IC1–LM334 quad op-amp
- IC2–4017 CMOS decade counter
- IC3–LM7805 or LM340-5 +5-volt regulator
- D1–D7–1N4003 diode
- Q1–Q6–IRF511 60-volt 3.5-amp MOSFET
- Q7–2N2222 or 2N3904 NPN transistor

Other components

- T1–120/12.6 volt center-tapped 3-amp power transformer
- J1–banana jack, red
- J2–banana jack, black
- J3–AC power receptacle
- F1–5-amp slow-blow fuse
- S1–SPST 6-amp switch
- NE1–neon indicator light with series resistor

Miscellaneous: fuse holder, perforated construction board, enclosure, aluminum for heatsinks, standoffs for mounting circuit board, wire, solder, etc.
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Parts placement isn't critical except for the 100-ohm gate resistors. They prevent VHF oscillations and should be placed within half an inch of the FET's. Just make sure that everything is securely mounted inside the cabinet to prevent shorting. Also, the prototype's metal cabinet has had several half-inch holes drilled in the bottom and rear for ventilation.

Power up

To safely test the inverter, it should really be operated with a 1-amp current-limited power supply. If you don't have one, simply connect it to approximately 12-volts DC, and keep a look out for smoke or sudden failures.

Connect an oscilloscope ground to chassis ground and the probe to the junction of D3 and D6; you will see an alternating DC signal. The frequency should be between 70 and 90 Hz. If it isn't you can adjust it by changing the value of R27. Adjust trimmer R17 for 180-volts peak. If you use a DVM or a VOM, connect it across the inverter's AC outlet, and adjust R17 for 120-volts AC.

Now it's time for a full-power test. You will need a 12.6 volt, 10-amp power supply or a car battery. A 120-volt, 40-watt light bulb makes a good load for testing. With a 12.6 volt input, the inverter will deliver 150 volts output, which will read about 105 volts on a DVM. With a 14.2 volt input, which is what an automobile alternator supplies, the output will be 115-volts AC.

We're sure you'll find many uses for your inverter at home or on the road from powering low-power AC equipment on a camping trip to re-charge your camcorder batteries as you drive to your next vacation spot!

- R-E

FIG. 3—THE TIMING RELATIONSHIPS in the inverter. When IC2 pin 3 goes high, the output of buffer IC1-c (pin 8) is high. That reverse biases D1 and allows the error amp signal to reach Q1, Q2, and Q3. At the same time, IC2 pin 4 is low, which causes the output of buffer IC1-d to be low. That grounds the gates of Q4, Q5, and Q6 thereby turning them off.

Parts placement isn't critical except for the 100-ohm gate resistors. They prevent VHF oscillations and should be placed within half an inch of the FET's. Just make sure that everything is securely mounted inside the cabinet to prevent shorting. Also, the prototype's metal cabinet has had several half-inch holes drilled in the bottom and rear for ventilation.

Power up

To safely test the inverter, it should really be operated with a 1-amp current-limited power supply. If you don't have one, simply connect it to approximately 12-volts DC, and keep a look out for smoke or sudden failures.

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- R-E

FIG. 4—IN THE PROTOTYPE, the FET's are not insulated from the heatsinks because the heatsinks are isolated from ground and all other circuitry.

- R-E

FIG. 4—IN THE PROTOTYPE, the FET's are not insulated from the heatsinks because the heatsinks are isolated from ground and all other circuitry.

- R-E
FOR ANYONE WHO HAS EVER TRIED TO repair a television with defective signals throughout, or changed a horizontal output transistor only to have it fail once more, we have a technique that can save hours of work and needless replacement of parts. All you have to do is check the "HOT pulse," or the signal at the collector of the horizontal output transistor. Let's see why this waveform is so important, and some key procedures for measuring the signal.

The HOT pulse is important because it performs many functions other than just sweeping the CRT beam horizontally. Some of the key functions of the horizontal output waveform are:

- It generates 0.7 amps of horizontal deflection current every 63.5 microseconds.
- It generates 15,000–30,000 volts DC for the picture tube.
- It generates 3,000–8,000 volts DC for the focus circuit.
- It delivers "trace-derived" high-current DC power from 16 to 30 volts to operate most circuits.
- It delivers "retrace-derived" low-current DC power of 185 to 220 volts.
- It provides 6.3 volts for the pulse current of CRT filaments.
- It is a critical safety feature.
- It provides accurate pulse voltages for the tuner's frequency-synthesis power source.

What to look for

The horizontal output pulse supplies operating voltages for the entire TV. It is therefore the most important waveform to check on every TV before and after changing parts. **Note:** In order to make any of the following measurements, your scope must be capable of measuring, and have input protection up to 2 kilovolts or more. Also, a digital-readout oscilloscope, although not essential to troubleshooting, will make it easier to make the measurements.

The first thing to check when analyzing the horizontal output pulse is the waveform shape; it should look like the one shown in Fig. 1, and be symmetrical in shape during the retrace time. A wide peak at the top of the retrace, or deep saddle conditions, can be caused by an off frequency or glitch in the horizontal transistor base-drive signal. Such problems are often caused by a change in the value of the output-transistor timing capacitors, or by an excessive load on a B+ supply.

Any excessive ringing or noise is a clear indication of deflection-system problems, such as a cracked integrated high voltage...
an amplitude reaching 1500 volts peak-to-peak. The noise could cause symptoms from drive lines in the video picture to faint noise throughout the TV's circuits.

First measure the DC voltage level of the horizontal waveform. In Figs. 1 and 2 you can see that the digital display shows approximately 118 VDC, which is the regulated B+ voltage. Next measure the peak-to-peak voltage of the waveform. As you can see from Fig. 3, the display shows 905 volts peak-to-peak.

The frequency of the waveform must also be measured. That's as simple as pushing a button on a digital scope. Figure 4 shows the frequency to be 15.7343 kHz. If everything checks out so far, you know the condition of the regulated B+ supply, that the TV is not in the shut-down mode, and that the horizontal oscillator is locked to the composite video sync pulse.

The duty cycle of the horizontal transistor output waveform is helpful in troubleshooting. The manufacturer specified that the retrace time should be from 11–14 microseconds, and the trace time should be 49–50 microseconds. Those recommended duty cycles should be observed when troubleshooting.

The time-duration measurement of the retrace pulse should be made between the 10% levels of the waveform. Some digital scopes are equipped to measure portions of a waveform with a delta-time feature. To make that measurement on a digital scope, align the pulse so that the top of it is at the 100% graticule marking and the bottom is at the 0% marking, using the volts/division and calibration knobs. (Make sure your scope will allow accurate measurement of the retrace pulse should be between 11 and 14 microseconds; 12.83 microseconds in this case.)
BUILD THIS
MAGNETIC FIELD METER

Determine your exposure to line-frequency magnetic-fields with our easy-to-build portable ELF gaussmeter.

REINHARD METZ

IF YOU ARE ONE IN A GROWING number of people who are concerned about the potentially harmful effects of exposure to magnetic fields, you will be interested in this important construction project. Now you can build your own gaussmeter, and determine the magnitude of magnetic flux densities in and around your home. Our handheld, battery-operated magnetic-field meter is sensitive from 0.1 microtesla (µT) to 20 milliteslas (mT), and has a frequency range from 50 Hz to 20 kHz.

Why all the worry?
Magnetic fields are all around us. They occur from the generation, distribution, and use of 50 and 60-Hz electricity, electronic equipment, and even from Earth’s magnetic field, which has always been present throughout Man’s evolution. Man has been “tuned” into Earth’s steady magnetic field of about 30 µT (at sea level) for millions of years. Some sources of excessive magnetic fields that have caused the greatest public concern include power-distribution substations, power lines, CRT terminals, and use of appliances.

Magnetic field intensities can vary greatly, depending on the exposure source and the distance from that source. The rate at which the field intensity falls off with distance can vary from one source to another, depending on how well the current-carrying lines are balanced, or how well the opposing lines of magnetic flux cancel each other out. Fields from coils, magnets, or transformers drop off rapidly with distance by a factor of 1/r³. In power lines, if currents flow in opposite directions, the drop-off is 1/r² because of partial field canceling. When unbalanced current exists, the field intensity falls off less rapidly as 1/r.

Figure 1-a, -b, and -c show drop-off rates of 1/r, 1/r², and 1/r³, respectively. Figure 2 lists some of the many sources of magnetic field exposure, with their range of intensities and drop-off rates.

Although a great deal of controversy still prevails, many people in the scientific community believe that exposure to magnetic fields of extremely-low frequency (ELF fields of 1-100 Hz) may pose a risk to human health. Some disturbing findings of exposure to ELF fields include a significant increase in serum triglycerides (a possible stress indicator) in humans, disorientation of chicks (a result suggesting that bird migration could be affected), and a slowed reaction time in monkeys.

A study conducted by epidemiologist Nancy Wertheimer and physicist Ed Leeper, found that exposures to magnetic fields as small as 0.25 µT correlated with a rise in cancer rates. In the study, the researchers examined wiring and transformers in the neighborhood of birth homes of children who had died of leukemia between 1950 and 1975, along with those of a control group of children who did not have the disease. The results of their studies were published in The American Journal of Epidemiology (March, 1979). Some experts argue that other factors, such as pollution and exposure to chemical carcinogens, make interpretation of those findings very difficult.

Standards for acceptable exposure to ELF fields are emerging, as are results of studies
FIG. 1—MAGNETIC FIELD drop-offs. A fast drop-off of $1/r^3$ (a), $1/r^2$ (b), and a slow drop-off of $1/r$ (c) is typical of many sources of magnetic fields.

describing possible hazard levels. If you are more interested a detailed account of scientific findings and the political history of the effects of magnetic-field radiation, we suggest a three-part series of articles by Paul Brodeur, *The New Yorker* (June 12, 19, and 26, 1989). "60-Hz and The Human Body", *IEEE Spectrum*, Parts 1-3, Volume 27, Number 9, pages 22-35 (August, 1990) is also a good source for technical information. The Environmental Protection Agency (EPA) has published a report titled "The Evaluation of the Potential Carcinogenicity of Electromagnetic Fields", publication number EPA/600/6-90/005B. This report contains analyses of 64 scientific studies, and is currently under review by the Scientific Advisory Committee.

Well, that’s enough background for now. Let’s examine some of the theory behind how the ELF meter works.

Theory

The quantity of magnetic flux density, $B$, is in units of webers/meter$^2$, or tesla (T). The magnetic flux, $\phi$, is defined by the integral $\phi = \int B \, ds = B \times A$ where $ds$ is the differential surface area and $A$ is the area that the coil encloses.

For a coil immersed in a field, the induced open-circuit voltage, $E$, is equal to the number of turns of a coil, $N$, times the rate of change of flux through it.

$$E = N \times \frac{d\phi}{dt}$$

Note that the value of $N \times \frac{d\phi}{dt}$ is actually negative with respect to the induced voltage value, but for our purposes we will just consider the magnitude of the product. The direction of the induced current is such that its own magnetic field opposes the changes in flux responsible for producing it.

If we substitute for $\phi$ we get

$$E = N \times A \frac{dB}{dt}$$

If the magnetic field of a sine wave is $B = a \sin(\omega t)$, $a$ is the amplitude in teslas and $\omega$ is the angular velocity ($2\pi f$), then

$$dB = a \omega \cos(\omega t) \, dt, \quad E = N \times A a \omega \cos(\omega t)$$

Since $\cos(\omega t)$ varies from +1 to -1, the peak magnetic field is defined as

$$E = N \omega a$$

For a frequency of 60 Hz, $\omega$ equals

$$2\pi \times 60 = 377$$

For a coil size of $3\frac{1}{2}'' \times 3''$, the area is .0068 m$^2$, and therefore

$$E = 2.56 \times 10^{-2} N a$$

For the 12-turn pickup coil that we’ll use, the sensitivity is 30 $\mu$V per $\mu$T.

Circuit description

The meter’s 12-turn field pickup is integrated into the unit’s circuit board. For remote sens-
FIG. 2—HERE ARE SOME PRIMARY SOURCES of magnetic field exposure with the range of field intensity in teslas, and drop-off rates.
FIG. 3—SCHEMATIC OF THE MAGNETIC FIELD METER. The magnetic field is picked up by L1 and appears as a voltage that is proportional to the field strength at the input of IC3-a, which amplifies the signal to 100 µV per µT. The signal is then further amplified by IC3-b and IC3-c to achieve the three tesla ranges.

All resistors are 1/4-watt, 1%, unless otherwise indicated.
- R1, R3, R12—10,000 ohms
- R2, R11, R15—33,200 ohms
- R4—10 ohms
- R5—R7, R22—R24, R27—1 megohm
- R8, R29—464,000 ohms
- R9, R13, R28—100,000 ohms
- R10, R14—1000 ohms
- R16, R20—42,200 ohms
- R17—4.7 meghms
- R18—51,100 ohms
- R19—46,400 ohms
- R21—1-megohm potentiometer, 5%
- R25—22,100 ohms
- R26—20,000-ohm potentiometer, 5%

Capacitors
- C1, C6—4.7 µF, 10 volts, electrolytic
- C2, C14—0.1 µF, electrolytic or polyester
- C3, C7, C15—0.1 µF, polyester
- C4, C6, C10—10 µF, electrolytic
- C5, C9—6.5 pF, ceramic disc or mica
- C11—100 µF, 10 volts, electrolytic
- C12—22 µF, 10 volts, electrolytic
- C13—330 pF, polyester
- C16—0.047 µF, polyester or ceramic disc
- C17—0.68 µF, polyester

Semiconductors
- D1, D2—1N4148 switching diode
- Q1, Q2—2N4124 NPN transistor
- IC1—IICL 7106 A/D converter
- IC2—4070 or 4030 quad 2-input exclusive-OR gate

Other components
- S1—MSS1200, SPST (Alco)
- S2—MSS4300, SPDT (Alco)

L1—18 turns, 3” diameter remote-sensing coil (optional, see text)
B1—9-volt alkaline battery, with connector
Case—Pac-Tec, HPS-9VB

NOTE: The following items are available from A & T Labs, P.O. Box 4884, Wheaton, IL 60187: A kit of all parts including PC board and case, without battery, $79.00; an etched, drilled and plated through PC board with solder mask and silk-screened parts placement, $15.00; a fully assembled and tested unit, $109.00. Add 6.75% sales tax for Illinois residents, 5% shipping and handling in U.S., 12% shipping and handling in Canada. Check or Money order (UPS COD in contiguous U.S. only) is accepted.
ing, an external field coil probe can be used. Figure 3 shows the complete schematic of the circuit. The magnetic field picked up by the coil appears as a voltage, which is proportional to field strength and frequency at the input of a cascaded amplifier IC3-a, -b, and -c. With a first stage amplifier gain of 3.3 set by R12-R10, the overall sensitivity is 100 µV per µT, or 100 mV per mT. The meter sensitivity is nominally 2 volts full scale, leading to the lowest level sensitivity of 20 mT full scale.

Op-amp IC3-a amplifies the signal to a normalized level of 100 µV per 1 µT. That voltage is further amplified by 1, 100, or 10,000 by IC3-b and -c. The three amplifier stages provide the three magnetic field ranges of 2 mT, 200 µT, and 2 µT (full scale). Components R3-C3 and R12-C7 establish a frequency roll-off characteristic that compensates for the frequency-proportional sensitivity of the pickup coil, and set the 20-kHz cut-off point.

Finally, IC3-d is a precision rectifier and peak detector. Its output drives IC1, a combination analog-to-digital (A/D) converter and LCD driver. Components R25-R29 and C13-C17 are used by IC1 to set display-update times, clock generation, and reference voltages. The decimal points are driven by IC2, as determined by the range-select switch S2. Transistors Q1 and Q2 serve as a low-battery detector, and turn on the battery annunciator in the LCD when the battery voltage drops below 7 volts.

Assembly and checkout

The finished unit shown in Fig. 4 uses a double-sided PC board, which is available from the source mentioned in the parts list. We also show the component side and solder side of the PC board if you choose to make it yourself. You can, however, build the circuit on a perforated construction board if you like, but remember to include the 18-turn remote sensing coil, L1, as indicated in the Parts List. Mount all parts below the LCD display first. It's easier to fix assembly problems if a socket is used with the LCD. Install all parts as shown in Fig. 5 paying attention to component valves and capacitor polarities. If you are using the internal sensing coil, install jumpers between L1-TP3 and L1-TP4.

If you are using the case specified in the parts list, raise and angle the display as necessary with wire-wrap IC sockets. Make holes in the front panel for S1 and S2. Mount the finished PC board in the case using a spacer for the single screw holding the center bottom of the board, and attach the battery connector. You are now ready for power-up and checkout.

With power on, adjust R26 for 1,000 volt between TP1 and TP2. Then, select the 20 mT range and short the pickup coil with a very short lead between TP3 and TP4. Adjust offset-null potentiometer R7 for a display of 0.00. Remove the jumper, and the meter is complete.
Calibration

Calibration of the meter is basically determined by the pick-up-coil characteristics, amplifier gains, and meter reference-voltage setting. The amplifier gains, as we previously discussed, are chosen to match the coil characteristics as closely as possible.

If you desire to calibrate your meter more exactly, you will need to generate a known magnetic field intensity. One way to do that is to pass a known current through a coil configuration whose field pattern characteristics are known. Figure 6 shows such a calibration setup. A good controllable signal source is a sine-wave generator and an audio amplifier, whose output is coupled to the coil by an 8-ohm resistor. Measure the voltage across the resistor, and use the calculations shown.

Measurement interpretation

A great deal of controversy exists in the emerging understanding of potential health hazards of low-frequency magnetic fields. The International Radiation Protection Association (IRPA) has set some interim standards based on 1984 World Health Organization guidelines. Those IRPA standards specify a continuous maximum magnetic field exposure for the general public of 100 µT and 500 µT as the maximum occupational exposure allowed over the entire working day.

Some European countries have already adopted strict magnetic field emission requirements for video display terminals, but the United States is taking a more cautious approach about developing and enforcing such guidelines.

Whatever studies and data you think are accurate, now you have a way to measure your own exposure and take whatever action you believe is prudent.

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**HOT TROUBLESHOOTING**

continued from page 108

rate digital readings when it is unscaled.) Select the dual-channel mode, couple channel B to ground, and align the trace so it lies on the 10% graticule marking. Select the time-measurement mode, and set the “begin” and “end” knobs so that the intensified trace section is as shown in Fig. 5. The digital display should show between 11 and 14 microseconds. (The display in Fig. 5 shows 12.83 microseconds.)

If you measure, say, 9 microseconds, instead of 11–14 microseconds, even though the peak-to-peak value, the DC voltage, the wave shape, and the frequency are correct, the TV will work for awhile, but will more than likely fail at some point. That’s because the horizontal output system sees a 35.7% reduction in retrace time—meaning that retrace is faster and generates higher voltage. Therefore, the horizontal output transistor is on longer at full-scan conduction, producing increased heat, increased scan-derived power supply levels, and higher voltages throughout the set. All the circuits are now stressed working at the higher voltages. That, in time, will cause components to fail.

Your scope can also be used to watch for an instantaneous start-up pulse. Simply connect the scope and preset it to view the HOT pulse. Then, watch the CRT as you apply power to the TV’s circuitry. If you see a pulse appear and then disappear, your start-up circuitry is operating and the set is in the shut-down mode.

If that happens, service the chassis in a “powered-down” condition by either halving the normal B+ level separately, or reducing the AC input power to 60–90 VAC. Then monitor the collector of the horizontal output transistor with your scope.

Many underlying performance problems can be uncovered by examining specific characteristics of the “HOT pulse.” The waveform shape, symmetry, and duty cycle of horizontal output transistor is critical in diagnosing and troubleshooting electrical malfunctions in your TV set.
This solid-state electronic compass uses Hall-effect sensors to keep you heading in the right direction.

ANTHONY J. CARISTI

MOST OF US HAVE AT ONE TIME OR another used a common magnetic compass, which often consists of a light-weight balanced magnet suspended on a pivot. The magnet, free to rotate, is affected by Earth's magnetic field, and assumes a position in which its north-seeking pole points to Earth's magnetic north pole. The geographical north pole of Earth is offset from the magnetic north pole by about 10 or 15 degrees in most areas of the United States.

Many low-cost compasses leave something to be desired in their performance, which can be affected by any tilt of the case or friction in the pivot. However, with the development of solid-state magnetic detecting devices, called Hall-effect generators, it is possible to construct a low-cost, reliable magnetic compass which has no moving parts and eliminates the disadvantages of inexpensive mechanical types. Because the project contains no moving or mechanically sensitive parts, it is an extremely rugged device that can tolerate all potential stresses encountered when hiking or traveling through rough terrain. Taking a reading on the compass is quick, easy, and very reliable.

This solid-state compass uses a unique detection system that produces two sharply defined points centered on the direction of magnetic north, as indicated by an LED. That permits a quick, accurate reading. The project, housed in a plastic enclosure, is small and lightweight, and is powered by a common 9-volt battery. Since the compass circuit is energized only when it is used to take a reading, the battery's useful life approaches that of its shelf life.

BUILD AN ELECTRONIC COMPASS

About the circuit

Development of a magnetically sensitive solid-state compass is made possible through a phenomenon called Hall effect, which was discovered in 1879 by Edwin Hall; he observed that a small voltage was developed at the edges of a current-carrying gold foil when the foil was exposed to a magnetic field. Solid-state technology now provides small, low-cost Hall-effect devices, which are very sensitive and able to detect Earth's extremely weak magnetic field.

The basic Hall-effect sensor, shown in Fig. 1, is a small sheet of semiconductor material in which a bias current flows. The Hall-effect output of the sensor takes the form of a voltage measured...
across the width of the conducting material, and will be negligible in the absence of a magnetic field. If the biased Hall sensor is placed in a magnetic field with the flux at right angles to the flow of current, a voltage output directly proportional to the intensity of the magnetic field is produced. Additionally, the voltage will be a function of the angle between the lines of force and the plane of the sensor. Maximum Hall-effect output voltage occurs when the face of the sensor is at right angles to the lines of force, and zero voltage is produced when the lines of force are parallel to the face of the sensor.

The Hall-effect sensor is further enhanced by using integrated-circuit technology to add a stable high-quality DC amplifier to the device. It then provides a usable linear output voltage which is sensitive enough to react to Earth's magnetic field (about 1/2 Gauss).

Referring to the schematic in Fig. 2, The Hall-effect generators (IC3 and IC4) are three-terminal linear devices which are driven by a regulated 5-volt supply provided by fixed-voltage regulator IC1. The output of each of the sensors is a DC voltage that varies linearly from a quiescent value of 2.5 volts as their position with respect to the lines of force of the magnetic field changes. A typical sensor has an output-voltage sensitivity of about 1.3 millivolts per Gauss.

Two Hall-effect generators are used in the circuit to provide twice the sensitivity of a single sensor. The two devices are oriented in opposite directions so that the change in output voltage of one sensor will be positive while that of the other will be negative as the compass is rotated.

The sensitivity control (R9) allows adjustment of the width of the output voltage of one sensor. The two devices are physically oriented in opposite directions so that the change in output voltage of one sensor will be positive while that of the other will be negative as the compass is rotated.

As shown in Fig. 3, the LED will be illuminated over a small arc as the compass is rotated full circle, and will remain off over the rest of the 360-degree span. The sensitivity control (R9) allows adjustment of the width of the input of the comparator exceeds the 3.4-volt reference level, the output of IC2-d (pin 14) goes high, applying forward bias to Q1. That in turn illuminates LED1 to indicate that a voltage exceeding the reference exists at IC2-b pin 7. The use of a voltage comparator to detect the change in output voltage of IC2-b (pin 7) produces two sharply defined points and allows more accurate determination of the magnetic north pole.

As shown in Fig. 3, the LED will be illuminated over a small arc as the compass is rotated full circle. True magnetic north is the position of the center of the arc.
the arc. Once the two LED switching points are determined, true magnetic north is then the position at the center of the arc.

Power is provided by a common 9-volt battery. The circuit draws about 25 mA and, since it’s usually powered for only a few seconds at a time, battery life is extremely long; several hours of continuous compass operation is also possible. Circuit stability with a falling battery voltage is ensured by the 5-volt regulator, IC5. When the battery is exhausted and cannot deliver sufficient current to operate the circuit, the LED will appear dim or will not illuminate at all.

Construction

The circuit, when built on the printed circuit board (for which we have provided the foil pattern), is very compact; the prototype is housed in a 2½-inch square by 1-inch high plastic enclosure, that has sufficient room to accommodate both the board and the 9-volt battery. A metal enclosure must not be used for this project—it can attenuate or distort Earth’s weak magnetic field. The power switch and sensitivity control are mounted on the side of the enclosure to allow easy operation of the compass.

Figure 4 shows the parts layout. The position of all polarized components (especially the Hall sensors) must be followed exactly as shown. The operation of the project depends upon the Hall generators being placed in opposite directions and exactly parallel as shown in Fig. 4. Note that the orientation of the sensors is determined by the marked face of the device, with pin 1 being on the left side when looking at the markings on the face of the device. The sensitivity control, R9, may be placed on the side of the enclosure to allow circuit adjustment when necessary. You should use a battery clip for B1. If desired, a suitable clip can be obtained from a discarded 9-volt battery (just peel away the metal case and rip the top off). Be very careful to wire the battery clip with the correct polarity.

When the circuit board is completed, examine it very carefully for shorts, opens, and cold solder joints. It is much easier to correct problems at this stage rather than later on if you discover that your project does not operate. A photo of the finished board is shown in Fig. 5.

Use a photocopy of the artwork in Fig. 6 for the top of the compass; you can simply glue it in place. Indicator LED1 is placed at the north indication of the compass by drilling a suitable size hole in the plastic top where the letter N would be. Be very careful when drilling; some plastics will shatter if subjected to excessive stress. Be sure to properly orient the top of the enclosure in accordance with the final position of the PC board.

Checkout

When you are satisfied that all wiring is complete and correct, the arc. Once the two LED switching points are determined, true magnetic north is then the position at the center of the arc.

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Checkout

When you are satisfied that all wiring is complete and correct,
the checkout procedure must be performed, and be sure to use a fresh 9-volt battery. Checkout requires a DC voltmeter connected to ground and the output terminal of IC1. Apply power to the circuit check for +4.75 to +5.25 volts. Measure the resistance between the 5-volt bus and ground; a normal reading is about 600 ohms. Measure the terminal voltage of the battery to be sure that it is delivering at least 7 volts under load to IC1. Replace a weak battery if necessary.

Next, measure the output voltage of IC2-a pin 1, and verify the voltage range of potentiometer R9. (Compass orientation is not important at this time.) The voltage should be about 2 to 3 volts DC. Measure and record the DC voltage that you observe at IC2-a pin 1.

Measure the voltage change at IC2-c pin 8 as the sensitivity control is rotated over its entire range. The difference between the highest and lowest readings should be about 0.45 volts. Ideally, the center of the measured voltage range should be close to the voltage recorded earlier at IC2-a pin 1.

If necessary, change the value of R8 and/or R10 so that the voltage range obtained at IC2-c pin 8 is somewhat centered about the voltage reading at IC2-a pin 1. This ensures proper adjustment range of the sensitivity control for the particular pair of Hall generators that are used in your compass project.

Once the sensitivity range is correct, rotate R9 over its range while observing the LED. At one end of the setting, the LED should be extinguished, and at the other end it should be illuminated; if not, check the polarity of LED1 and the orientation of Q1. Check pin 14 of IC2-d to be certain it swings from about zero to battery voltage as R9 is rotated over its range. Check pin 13 of IC2-d for a voltage of about 3.4 volts as set by R11 and R12. Problems in this area may warrant replacing IC2 if everything else checks out alright—check your soldering before changing the IC.

When the LED operates as described, the project is ready to be tested under actual operating conditions. Before you start, make sure that there are no magnetic fields nearby, and the project is not shielded by a large mass of iron or steel.

While holding the unit horizontally in any direction, apply power and carefully adjust R9 so that the LED is at the switch-over point between on and off; allow at least 10 seconds for the circuit to stabilize. Flicker of the LED is normal as the circuit switches back and forth. Once R9 is set, rotate the compass over a 360-degree arc (full circle) and note that the LED will be on over part of the arc, and off over the rest. If necessary, readjust potentiometer R9 very slightly to obtain this result. The optimum setting for R9 will be at the point where the arc of illumination is as small as possible.

As the compass is rotated over the illuminated arc, note the two on/off points. When the compass is positioned halfway between those points, it is facing the magnetic north pole, and the scale indications on its face indicate all other directions.

Using the compass

Always be sure that the battery is reasonably fresh, and take along an extra one before starting out on an excursion with the compass. (A weak battery will be indicated by a dim or totally unlit LED.) Avoid taking a compass reading in any area where there may be a magnetic field from a nearby device, or where Earth’s magnetic field is shielded by a large mass of metal.

Hold the compass in a horizontal position and rotate it full circle while observing the LED. Adjustment of the sensitivity control is indicated if the LED is totally on or totally off as the compass is rotated. Always allow at least 10 seconds operating time for the circuit to stabilize. Once the sensitivity control is adjusted, it should not require readjustment unless the project is subjected to an extreme change in temperature.

Don’t forget that the electronic-compass circuit can be used for things other than a simple direction finder. It provides an electronic means of finding north, so it should be easy to interface the compass to other devices that may need to know where north is—a robot, for example.
IF YOU’VE BEEN LOOKING FOR A WAY to generate high voltage, you’ve undoubtedly run across the volt-
age doubler. Voltage doubling using diode-capacitor combina-
tions is a common practice. How-
ever, whole banks of doublers, called cascades, can also be used for producing extremely high DC voltages from moderate to high AC voltages. Such high DC volt-
ages may be needed for TV sets, lasers, air purifiers, industrial smoke-stack dust removers, negative-ion generators, and, of course, for experimenting, on which we’ll concentrate here.

Half-wave doubler

Figure 1 shows a half-wave volt-
age doubler; we’ll assume that C1 and C2 are initially discharged. During the first half-cycle shown in a, the upper input terminal is positive and the bottom negative, so D1 conducts and C1 charges to about 170 volts peak. Diode D2 can’t conduct, since it’s back-bi-
ased, so C2 discharges through RL. In the second half-cycle (b), the analysis is similar, except that D2 conducts and C2 charges.

The circuit is really a transfor-
mersless voltage amplifier. While T1 can provide isolation, as well as increase the AC voltage initially going into the doubler, the amplification due to the doubling action would occur without it. When the polarity reverses, both the input voltage and the charge across C1 are in series like two batteries, producing about 340 volts peak. One problem, though, is that a half-wave doubler can’t be used with a load that draws much current.

Full-wave doubler

Let’s see how a full-wave voltage doubler is related to and built from both positive and negative half-wave rectifiers. Figure 2-a shows a half-wave rectifier with a positive output. Fig. 2-b shows the same version with a negative output, and Fig. 2-c shows the two combined into a full-wave voltage rectifier.

The full-wave voltage doubler shown in Fig. 3 has been redrawn for greater clarity; it has better regulation than a half-wave ver-
sion, and is easier to filter. The circuit produces nearly double the peak AC voltage of 170 volts, or about 340 volts peak across RL. For the first half-cycle (a), D2 is cut off and D1 conducts, so that Vc1 equals approximately 170 volts DC. On the next half-
cycle (b), the positive voltage is replaced by a negative voltage, so D2 conducts and D1 is cut off. RL goes across C1 and C2 in series, effectively creating a doubled level of about 340 volts DC.

Unlike the half-wave voltage doubler, the full-wave version has two capacitors across RL rather than one. Whereas C1 shown in Fig. 1 is cut off and unsupplied for half of every cycle, C1 and C2 in Fig. 3 are supplied on alternate half cycles. When the capacitor corresponding to the diode that’s cut off discharges, it can only do so through the capacitor being supplied, slightly decreasing both its current and the max-
imum voltage it reaches.

Measuring high-voltage DC

Voltage measurements will be possible only to about the second or third stage of a cascaded volt-
age doubler with most volt-
meters. Beyond that, you’ll need to use either a high-voltage DC meter or an external voltage di-
vider for use with a standard high-impedance volt-
meter (10 megohms or more).

A good voltage divider that can be used for the purpose of high-
voltage measurements is the RCA SK3868/DIV-1, a high-vol-
tage DC divider; it’s used in TVs to reduce the final anode voltage going to the CRT to the level re-
quired for the focus voltage. It consists of resistors R1 (200 megohms) and R2 (40 megohms) in series, as shown in Fig. 4. There are three leads, one for the free ends of each resistor, and the other at their juncture. If you put both a 10-megohm meter (shown as ZM in Fig. 4) and a 2.7-
megohm resistor (R3) in parallel with the 40-megohm resistor (R2), you can achieve almost ex-
actly 100:1 range multiplication, for a full-scale deflection of 20 kilovolts DC.
FIG. 1—HALF-WAVE VOLTAGE DOUBLER. During the first half-cycle (a), D1 conducts, D2 cuts off, C1 charges to 170 volts peak, and C2 discharges through R_L. For the second half-cycle (b), the input polarity is reversed, and both the input and C1 are in series, producing 340 volts peak. Now D1 cuts off while D2 conducts, and the current divides between C2 and R_L; the cycle then repeats.

FIG. 2—TWo HALF-WAVE RECTIFIERS, one with a positive output (a) and one negative (b), combine to make a full-wave voltage doubler (c).

Cascaded voltage doublers

Figures 5–8 show four additional voltage doublers. The one shown in Fig. 5 is the most straightforward. If you build it, use 1N4007 diodes with peak inverse voltage (PIV) ratings of 1 kilovolt for D1–D6, and 0.068–0.1 µF capacitors with working voltages of 400 volts DC. Figure 5 is electrically identical to the one in Fig. 6, so keep that in mind if you should come across either format. Figure 7 shows an extended version that’s better

FIG. 3—FULL-WAVE VOLTAGE DOUBLER, redrawn for greater clarity. For the first half-cycle (a), D2 is cut off and D1 conducts, producing about 170 volts DC across C1. On the next half-cycle (b), D2 conducts and D1 is cut off. The output voltage is now across C1 and C2 in series, doubling the level to about 340 volts DC.

FIG. 4—TO MEASURE HIGH VOLTAGES with an ordinary 10-megohm meter, you can use the RCA SK3868/DIV-1 high-voltage divider. The circuit provides a 1:100 voltage division, allowing 20 kilovolts to be measured on a 200-volt scale.

FIG. 5—THIS CASCADED DOUBLER uses 1N4007 diodes rated at 1 kilovolt PIV, and capacitors from 0.068–0.1 µF with a 400-volt DC working voltage. Stabilized for moderate-current applications; it’s called either a Cockcroft-Walton or Greinacher cascaded voltage doubler.

You can use a sewing needle as an emitter for the doubler shown in Fig. 8 to generate “corona wind.” That will sound like a hissing noise. (We’ll show you how to demonstrate the “wind” later on.) The circuit delivers 3.75 kilovolts DC when powered from 120 volts AC, or 7.5 kilovolts DC when powered from 240 volts AC.

The output of a cascaded voltage doubler should be terminated with no less than 200 megohms, and only then be allowed to extend beyond a protective plastic case. For safety. Voltages as high as 5 megavolts DC have been generated using
cascaded voltage doublers, especially when operating in a pressurized atmosphere. The biggest advantage to using voltage doublers is that they use inexpensive low-voltage parts. Otherwise, if all the parts had to be of the high-voltage variety, you would have to use expensive and rather large capacitors like the one shown in Fig. 9.

If you have problems with the circuit in Fig. 8 (or any other high-voltage circuit), you must discharge every capacitor (we'll tell you how in a minute) before you check for malfunctions. When examining the circuit for problems, closely check the solder connections, and then the diode directions and continuity. The 1N4007's should have a resistance of 1.1K when forward biased and be open when reverse biased, while the capacitors should all have infinite resistance.

To properly discharge capacitors, build a discharging wand like the one shown in Fig. 10. Use a 2-foot wooden (or plastic) dowel, and connect a stiff wire tip (piano wire works well) to a cold water pipe as earth ground with a good electrical connection. Discharge all capacitors twice, since they generally either hold charge, or tend to recharge from other capacitors. Don't use an AC line ground or chassis ground instead of an earth grounded water pipe, or you may blow a fuse or damage parts.

Figure 11 shows a switch for high-voltage DC that you can use with any of the cascaded voltage doublers shown here; standard switches may present a shock hazard. Also, use an electromagnetic interference (EMI) line filter like the one seen at left in the photo to keep high-voltage DC out of house wiring, and to prevent shock from static charge. The EMI filter is from Corcom Corp. (1600 Winchester Road, SEWING NEEDLE AS EMITTER ELECTRODE FOR CORONA WIND
3.75/7.5kVDC

FIG. 8—THIS 25-STAGE VOLTAGE DOUBLER will generate "corona wind." It delivers 3.75 kilovolts DC when powered from 120 volts AC, or 7.5 kilovolts DC when powered from 240 volts AC.
When you build a cascaded voltage doubler, you can encase the circuit in pure paraffin oil or candle wax to reduce the chances of getting shocked. It will also minimize corona loss, so the high-voltage DC arrives where it's needed. Figures 12 and 13 show a typical ladder-type voltage doubler before and after being sealed in wax.

Experiments

There are many experiments that can produce observable effects due to the high-voltage DC produced by voltage doublers.

- With a high-voltage emitter pointed at a ground plate (used to attract ions), with a burning candle placed in between them (see Fig. 14), you'll see the candle flame deflect toward the metal plate.

- You can make a rotor for an ion motor, using a light pivot made from a rivet with thin, stiff wire (like piano wire) attached, as shown in Fig. 15. The rotor must be balanced on top of the sewing-needle emitter (much as in a compass) used for the doubler shown in Fig. 11. (We ran a similar construction project in Radio-Electronics, February, 1991.) When powered up, the rotor will spin and a hissing sound will be heard. Both ends of the wire are bent at opposite right angles, so the emitted electrons propel the wire in a circle. You should sharpen both ends of the rotor wire to provide a sharp surface good for corona generation and electron emission. The sharpened ends will have a small radius of curvature (a tight curve or bend), giving rise to a highly distorted electric field at its surface. The high electric field is what tends to ionize air molecules in the vicinity.

- Another experiment you could try involves holding a fluorescent tube near the emitter. The tube will glow, but be careful not to touch the terminals on the ends, or you'll get a shock.

- Lines of force of an electrostatic field can be demonstrated by placing the electrodes (the high-voltage DC output and ground) in a tray covered with castor oil containing some farina. The farina will produce the pattern of the electric field lines; similar to iron filings shaken lightly on a piece of paper in the presence of a bar magnet.

- If you place two round door knobs on insulated stands made from plastic cups filled with candle wax, and then charge them, then a plastic ball suspended from a string will be drawn to and touch the positive electrode, and fall back to center when the spheres are discharged (see Fig. 16-a). A plastic ball coated with conductive lacquer swings toward the positive electrode like a pendulum; when the ball and doorknob touch, the ball becomes positively charged, so they repel one another. It then swings toward the negative side, absorbs electrons, becomes negatively charged, and is repelled back to the positive. The process repeats indefinitely as long as the high-voltage DC is present, and it will continue to operate for some time after it's shut off. The charge exchange is slow, and there'll be arcing at the positive electrode.

- A grounded metal ball alternates between both electrodes, like the conducting plastic ball. However, the arcs are smaller due to its greater weight, and should be observed at both ends, but more on the positive side.

- A light cotton ball should be drawn to the positive electrode and hang there by itself, as shown in Fig. 16-b. It's then repelled 0.5-inch toward the negative electrode, and the process should repeat indefinitely.
Build this powerful serial-bus analyzer for a fraction of the cost of commercial units—and learn about the ever-popular 68705 microcontroller in the process.

TERMINAL/MONITOR

STEVEN AVRITCH

HAVE YOU EVER BEEN FRUSTRATED by a problem with an RS-232 line? Inexpensive breakout boxes with five or six LED's suffice for solving simple problems, but they don't provide enough information to debug the serious kind. On the other hand, full-featured serial bus analyzers (SBA's) give you all the debugging information you need, but can cost close to $1000.

However, you don't really have to spend that much. Now you can build a powerful RS-232 monitor that does most of what the expensive SBA's do, yet doesn't cost much more than a quality breakout box. In addition, this project can also be used as a portable, battery-operated terminal. Features are summarized in Table 1.

Hardware design

The heart of the project is Motorola's MC68HC705C8 single-IC microcontroller, a 40-pin DIP containing built-in PROM, RAM, serial and parallel I/O ports, and clock.

The monitor requires two serial receivers: one for the TXD line and one for the RXD line. Because the microcontroller has only one built-in serial port, a second one has been implemented in software. However, the second port still requires a ±12-volt RS-232 interface. The schematic is shown in Fig. 1, and the wiring diagram is shown in Fig. 2.

A Maxim MAX232 RS-232 transceiver (IC2) provides the transmission portion of the interface; it converts TTL signals from the microcontroller to RS-232 levels. Unlike most RS-232 transceivers, which require separate +12- and −12-volt power supplies, the MAX232 has built-in charge pumps that generate the required voltages from a single 5-volt supply. A standard 1489 device (IC4) converts incoming ±12-volt signals to TTL levels.

Switch SI is a DPDT unit that selects terminal or monitor mode. In terminal mode the CPU controls the logic levels of both lines. However, in monitor mode, all lines from the primary port (J1) are directly connected to the secondary port (J2), in which case all signals pass straight through and the device simply monitors TXD and DTR.

Other notable components include XTAL1 and the keyboard. The crystal is a standard, readily available device; it must have a frequency of exactly 2.4576 MHz in order to generate the correct baud rates. The keyboard is a surplus unit from an old TI-99/4 personal computer; the keyboard is inexpensive and readily available through many suppliers.

A beeper may be connected to pin 19 (port B7) of the microcontroller. The beeper will sound whenever the monitor detects a bell character (ASCII 07 or Ctrl-G). The beeper must be TTL-compatible (meaning that it must be powered from a 5-volt supply and draw less than 3 mA).

The low power consumption of the microcontroller allows it to operate over a wide range of input voltages (8–15 volts DC); IC3 provides voltage regulation. The output of IC3 is 6 volts; diode D2 drops voltage even further, to about 5.3 volts. D1 provides reverse-polarity input protection.

Operating modes

The RS-232 Terminal/Monitor has four basic modes of operation, as shown in Table 2. You use the keyboard to select mode, as well as a variety of operational parameters. In use, you must place SI in the Terminal position when in terminal mode, and in the Monitor position when in any of the three display modes.
The terminal-mode display is similar to a dumb ASCII terminal, except that the display is limited to forty characters by two lines. The CPU converts keyboard characters to ASCII, then transmits them over the serial link. Conversely, received characters are displayed in ASCII on the LCD screen.

The Display Bits mode continuously displays the status of the six primary RS-232 signals, as shown in Fig. 3-a. The Display ASCII mode shows activity on the TXD and RXD lines. TXD data appears in ASCII on line one and RXD data on line two (Fig. 3-b). The Display Hex mode is similar, except that each character appears in hexadecimal format, as shown in Fig. 3-c. (A period "." indicates that the line was inactive when a character was received on the other line.)

**TABLE 1—FEATURES**

<table>
<thead>
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<th>Feature</th>
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<tbody>
<tr>
<td>40-character x 2-line LCD display</td>
</tr>
<tr>
<td>Full keyboard</td>
</tr>
<tr>
<td>Selectable baud rate (300 – 19,200)</td>
</tr>
<tr>
<td>Selectable protocol (number of data and parity bits)</td>
</tr>
<tr>
<td>Programmable scroll rate</td>
</tr>
<tr>
<td>Recall of last two lines displayed</td>
</tr>
<tr>
<td>8- to 15-volt DC power (9V battery is perfect)</td>
</tr>
<tr>
<td>CMOS design for low current drain and long battery life</td>
</tr>
</tbody>
</table>

**TABLE 2—MODES OF OPERATION**

<table>
<thead>
<tr>
<th>Mode</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Terminal</td>
<td>The unit acts as a simple RS-232 &quot;dumb&quot; terminal. The unit can be connected anywhere a dumb terminal is needed.</td>
</tr>
<tr>
<td>Display Bits</td>
<td>Displays status (high or low) of the six primary RS-232 lines (TXD, RXD, RTS, CTS, DSR, DTR).</td>
</tr>
<tr>
<td>Display ASCII</td>
<td>Displays TXD (line 1) and RXD (line 2) activity in ASCII format.</td>
</tr>
<tr>
<td>Display Hex</td>
<td>Like Display ASCII mode except hexadecimal display.</td>
</tr>
</tbody>
</table>

**FIG. 1—SCHEMATIC OF THE TERMINAL/MONITOR.** The keyboard is a surplus unit from a TI 99/4 computer.
During operation in ASCII or hex mode, you can press Cntl-S to halt input temporarily and read what is displayed. Then simply press Cntl-Q when you’re ready to continue.

The monitor has several keys that can produce more than one symbol. For example, the question mark symbol is located on the front of the “I” key. To produce a question mark, press I and the FCTN key simultaneously.

**Function requests**
You can change several operating characteristics by pressing special key combinations, as shown in Table 3. To set a given parameter, press FCTN and the key shown in column one of the table.

**Select Mode.** Press FCTN-M, then press 1–4 to select mode.

**Current Parameters.** Press FCTN-0 to view current settings for mode (terminal), baud rate (1200), protocol (7E1), scroll rate (0), scroll mode, linefeed status (enabled), and on/off-line status (on-line). (Default values shown in parentheses.)

**Set Baud Rate.** Press FCTN-1 to set baud rate. The monitor runs from 300 to 19,200 bits/sec, but the maximum rate in ASCII and Hex Display modes is 9600.

**Protocol.** Press FCTN-2 to define the number of data bits (7, 8), parity (Even, Odd, None), and stop bits (1, 2) in each byte sent or received. The terminal currently supports three popular formats: 8N1, 7O1, and 7E1.
PARTS LIST

All resistors are 1/4-watt, 5%
R1-10,000 ohms, PC-mount potentiometer
R2-10 megohms
R3, R13-10,000 ohms
R4-R12-10,000 ohms, SIP

Capacitors
C1-1 µF, 16 volts, radial electrolytic
C2, C3-18 pF, ceramic disk
C4, C5-100 µF, 25 volts, axial electrolytic
C6-0.01 µF, ceramic disk
C7-C10-10 µF, 16 volts, radial electrolytic

Semiconductors
IC1-MC68HC705C8 CMOS microcontroller
IC2-MAX232 5-volt RS-232 transceiver
IC3-MC7806CT 6-volt regulator, TO-220 case
IC4-MC1489A RS-232 receiver
D1, D2-1N4148 switching diode

Other Components
XTAL1-2.4576 MHz
J1-16-pin 0.1" dual-row header
J2-30-pin 0.1" dual-row header
J3-8-pin 0.1" dual-row header
J4, J5-DB25 connector, male
S1-DPDT switch
BZ1-TTL-compatible beeper (Radio Shack #273-65 or equivalent)
Display-40 x 2 line LCD display module (Hitachi LM018L or equivalent)
Keyboard-Surplus TI-99/4 (48-key, 15-pin connector)

Note: The following are available from Simple Design Implementations, P.O. Box 9303, Forestville, CT 06010. (203) 582-8526:
- Complete kit including everything in parts list, IC sockets, PC board, instructions, and schematic—$754 + $5 S/H
- Preprogrammed MC68HC705C8, instructions, and schematic—$28.00 + $2.50 S/H
- Software on 5½-inch IBM-compatible floppy disk—$15.00 + $2.50 S/H
- PC board only—$13 + $2 S/H.
CT residents please add 8% sales tax.

Scroll Rate Delay. Press FCTN-3 to vary scroll rate. Data may scroll by faster than you can read it; variable scroll rate allows you to reduce scrolling speed by disabling the DTR line for a while after receiving each carriage return. Scroll-rate delay may vary from 0 (no delay) to 9 (maximum delay—about three seconds). The device connected to the monitor must recognize DTR and stop sending data when it is low. This works only in terminal mode.

Two-line/Scroll Mode. Press FCTN-4 to switch between the continuous scrolling and two-line modes. In scroll mode the terminal scrolls incoming data at the scroll rate set by FCTN-3. In two-line mode the terminal stops the display every time both display lines are filled. You must then press the space bar to continue. This feature works only in terminal mode.

Enable/Disable LF. Press FCTN-5 to switch between responding to and ignoring incoming linefeed characters. Some terminals, modems, and host systems issue a linefeed (LF) in addition to a carriage return (CR) at the end of every line. Extra linefeeds cause a blank line to be displayed on the screen, hence make it difficult to read. Tell the monitor to ignore extra linefeeds by selecting the LF Disabled mode. This feature works only in terminal mode.

Recall Last Two Lines. Press FCTN-6 to review the last two lines that scrolled off the screen. Press any key to return to the current two lines. Note that communication is disabled via the DTR line when displaying the last two lines. This feature works only in terminal mode.

Local/On-line Mode. Press FCTN-L to switch between local and on-line modes. In local mode, RS-232 communications are disabled; characters typed on the keyboard appear on the LCD display immediately. In on-line mode, RS-232 communications are enabled. Characters typed on the keyboard are transmitted over the serial link and are not displayed until they are echoed by the host system. This feature works only when used in the terminal mode.

Interfacing
To use the device as a dumb
ASCII terminal, be sure to place switch S1 in Terminal position, then make sure that the unit is operating in terminal mode, and then enter the correct operating characteristics (baud rate, protocol, etc.). Last, connect the unit to the host system via a standard 25-conductor cable. If the unit doesn’t seem to work, you may need to reverse pins 2 and 3 using a null-modem cable or adapter.

To use the device as a monitor, connect it in series with the two devices (host and terminal, host and modem, etc.), place S1 in Monitor position, and choose one of the terminal modes. Then all you have to do is set protocol, baud rate, etc.

Software
The MC68HC705C8 software consists of an assembly-language program; unfortunately the listing is too long for publication. The program must be assembled and the resultant object code must be burned into the microcontroller’s built-in EPROM. If you don’t have facilities for assembling the program and burning it in, don’t worry—you can purchase a pre-programmed MC68HC705C8 from the author; take a look in the parts list for details.

If you wish to modify the source code (or maybe just look it over), you can download it from the RE-BBS (516-293-2283, 1200/2400, 8NI) as a file called SMART232.SRC, or you can otherwise order it directly from the author.

Construction
You can build this project using wire-wrap or a PC board. A foil pattern is provided if you’d like to make your own board; you can also purchase a ready-to-use board from the source mentioned in the parts list if you don’t want to make your own.

Using Fig. 4 as a guide, assemble the PC board starting with the passive components. Then mount sockets for the IC’s, connectors J1–J3, and don’t forget the two jumpers. Check your work carefully, correct any mistakes, and then insert the IC’s into the sockets. Connect the cables from the keyboard and LCD module. Then make the connections between J1, J2, and the PC board. A photograph of the author’s prototype unit is shown in Fig. 5.

After verifying all wiring, apply power to the unit; a brief sign-on message should appear on the display. Adjust Contrast potentiometer R1 for best effect. Now set switch S1 to the appropriate position, and set the desired mode and operating characteristics using the Function keys. That’s all there is to it!

Troubleshooting hints
If the sign-on message does not appear on the display on power-up, the most likely cause of the problem is a wiring error. The error is probably in the power or ground lines (IC1, pins 20 and 40), the oscillator lines (IC1, pins 38 and 39), the reset line (IC1, pin 1) or the display interface lines (pins 1–6 and 11–14 of the LCD display).

If the terminal comes up with the initialization message but does not seem to communicate, check the keyboard and RS-232 driver. Check the keyboard by pressing FCTN-0 to display system parameters. If they don’t appear, check the keyboard wiring. If the terminal does display system parameters, press FCTN-L to put the terminal in local mode and then test the keyboard by pressing every key and checking for the correct letter on the display. If the keyboard checks out, the problem is probably located somewhere in the RS-232 driver, in the cabling, or in the RS-232 line itself.

Remove the connector from the terminal’s primary port, set the DPDT switch to terminal mode, and check the voltages on the terminal’s primary port. They should be close to the values shown in Table 4. If any of the values do not agree with Table 4, check the wiring of the RS-232 line itself.

Wrapping up
The author housed his prototype in a custom-built enclosure that resembles a miniature RS-232 terminal. You could use an off-the-shelf metal or plastic case just as well. Just keep in mind that, however you package it, this terminal/monitor is a useful and fun project.
**THD ANALYZER**

we drive the input of the amplifier with a 1-volt pure sine wave and we obtain the 20 volts required at the output. Since 20 volts RMS is 56 volts peak-to-peak, S2 must be set to >20V. With S5 in "THD" position a reading of 1.4 mV (0.0014V) DC is obtained. In the "REF" position we read 2.00V. Always read the range on the DMM that affords the best resolution. Now make the calculation: 0.0014/2.00 x 100 equals 0.07 percent THD.

Voltage amplifiers are measured in the same manner, but without the requirement of a load resistor. Such amplifiers will probably show lower THD voltage readings. If the THD voltage is too low on the DMMs mV range, set S4 to "x 10." In this case, however, divide the displayed THD voltage by 10. For example, 0.8 mV on the meter is read as 0.08 mV, since it was amplified 10 times to 0.8 mV.

If the measured THD percent is three times or more greater than the analyzer's measurement "floor," no correction of the measured value is required. If it's less than three times, a good approximation of the true THD percent is given by the formula D (DUT) = \[ D_{DUT} = \frac{V_{THD}}{V_{REF}} \times 10^{\frac{S4}{10}} \]

That means if you measure a preamp's THD at 0.005%, and the measurement floor is 0.004%, the true THD of the preamp is 0.003%.

Depending upon the depth of the notch and how closely the RC's in the oscillator's frequency selective network are matched, the "floor" should typically be 0.003 to 0.005%. You can check the "floor" value in the same way you measured the 1% THD calibrator, explained earlier. Instead of using the calibrator signal, feed in a maximum signal level from output J1. If you use the \( x10 \) position, it may be necessary to connect J8 to an external earth ground.

Finally, you should know that THD measurements above 10 percent are less accurate than those below 10 percent, since the reference includes distortion plus the fundamental—not just the fundamental.

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SPEAKER PROTECTOR
continued from page 83
for the protector circuit is shown in Fig. 5. The PC board is configured for on-board mounting of the 5-amp relay that the author used.

A compatible relay with a different pinout can be used. However, it may have to be mounted on the edge of the board using double-sided tape; it will then have to be hardwired to the board. The parts-placement diagram for the power-supply board is shown in Fig. 6.

After mounting all components, solder leads of adequate length to the boards, and use different colors for the speaker and power ground leads to avoid connecting them together. Next, connect the power leads between boards, and connect an AC line cord to the power supply. The author’s completed unit, containing a protector circuit for both the left and right channels and one power supply, is shown in Fig. 7.

The completed boards can be mounted in a case like the one pictured in Fig. 7. When drilling holes in plastic cases, start with a small drill bit and work your way up.

As a final word, the circuit is designed to protect speakers from excessive DC levels caused by amplifier failure. However, it will not protect a speaker that’s rated at power levels much less than the driving amplifier can supply—only your own common sense in keeping the volume down will protect your speakers in a situation like that.

COLOR BAR GENERATOR
continued from page 55
chroma on the white bar (the first bar after horizontal sync) with R10 and R9.

Adjustment with Monitor

If you have a monitor, hook up the video output of the video generator to the monitor’s video input and do all the adjustments looking at the white color bar on the left side of the screen. If the red, green, and blue video levels are set up properly the white bar should be full brightness and white. If it is not, adjust R8 for the overall brightness and then adjust R10 and R9 for a pure white bar.

We hope you’ve learned something about color video from this project. Everything is neatly broken into fairly simple blocks so you can be assured of success. If you have an IBM Clone computer with a CGA board (or other video card with NTSC-compatible sync) you can use just the encoder section to generate NTSC-compatible sync and RGB video outputs.

FIG. 7—HERE’S THE AUTHOR’S PROTOTYPE. It contains a protector circuit for both the left and right channels and one power supply. The entire unit is only about 4 inches wide by 6 inches long.

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Wake up! You may be the victim of stolen words—precious ideas that would have made you very wealthy! Yes, professionals, even rank amateurs, may be listening to your most private conversations, even rank amateurs, may be listening to your most private conversations, even rank amateurs, may be listening to your most private conversations. Destroy this video carefully and understand its contents, you have taken the first important step in either acquiring professional help with your surveillance problems, or you may very well consider a career as a countersurveillance professional.

Wake up! If you are not the victim, then you are surrounded by countless victims who need your help if you know how to discover telephone taps, locate bugs, or "sweep" a room clean.

There is a thriving professional service steeped in high-tech techniques that you can become a part of! But first, you must know and understand Countersurveillance Technology. Your very first insight into this highly rewarding field is made possible by a video VHS presentation that you cannot view on broadcast television, satellite, or cable. It presents an informative program prepared by professionals in the field who know their industry, its techniques, kinks and loopholes. Men who can tell you more in 45 minutes in a straightforward, exclusive talk than was ever attempted before.

Fooling Information Thieves

Discover the targets professional snoopers seek out! The prey are stock brokers, arbitrage firms, manufacturers, high-tech companies, any competitive industry, or even small businesses in the same community. The valuable information they siphon off may be marketing strategies, customer lists, product formulas, manufacturing techniques, even advertising plans. Information thieves cavedrop on court decisions, bidding information, financial data. The list is unlimited in the mind of man—especially if he is a thief!

You know that the Russians secretly installed countless microphones in the concrete work of the American Embassy building in Moscow. They converted to voice scramblers, midget radio-frequency transmitters, and other bugs, plus when to use disinformation to confuse the unwanted listener, and the technology of voice scrambling telephone communications. In fact, do you know how to look for a bug, where to look for a bug, and what to do when you find it?

Bugs of a very small size are easy to build and they can be placed quickly in a matter of seconds, in any object or room. Today you may have used a telephone handset that was bugged. It probably contained three bugs. One was a phony bug to fool you into believing you found a bug and secured the telephone. The second bug placates the investigator when he finds the real thing! And the third bug is found only by the professional, who continued to search just in case there were more bugs.

The professional is not without his tools. Special equipment has been designed so that the professional can sweep a room so that he can detect voice-activated (VOX) and remote-activated bugs. Some of this equipment can be operated by novices, others require a trained countersurveillance professional.

The professionals viewed on your television screen reveal information on the latest technological advances like laser-beam snoopers that are installed hundreds of feet away from the room they snoop on. The professionals disclose that computers yield information too easily. This advertisement was not written by a countersurveillance professional, but by a beginner whose only experience came from viewing the video tape in the privacy of his home. After you review the video carefully and understand its contents, you have taken the first important step in either acquiring professional help with your surveillance problems, or you may very well consider a career as a countersurveillance professional.

The Dollars You Save

To obtain the information contained in the video VHS cassette, you would attend a professional seminar costing $350-750 and possibly pay hundreds of dollars more if you had to travel to a distant city to attend. Now, for only $49.95 (plus $4.00 P&H) you can view Countersurveillance Techniques at home and take refresher views often. To obtain your copy, complete the coupon or call.