

# Popular Electronics®

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14278

A **PE TESTED** FIRST!

## Now You Can Enjoy HI-FI Remote Sound

**WITHOUT RUNNING WIRES  
AROUND THE HOUSE**

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- How to Use CB Radio  
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### TEST REPORTS:

**B-I-C Belt-Drive Record Changer**  
**Pickering Stereo Phono Preamp**  
**Tram AM CB Mobile Transceiver**  
**Heath Color Organ**

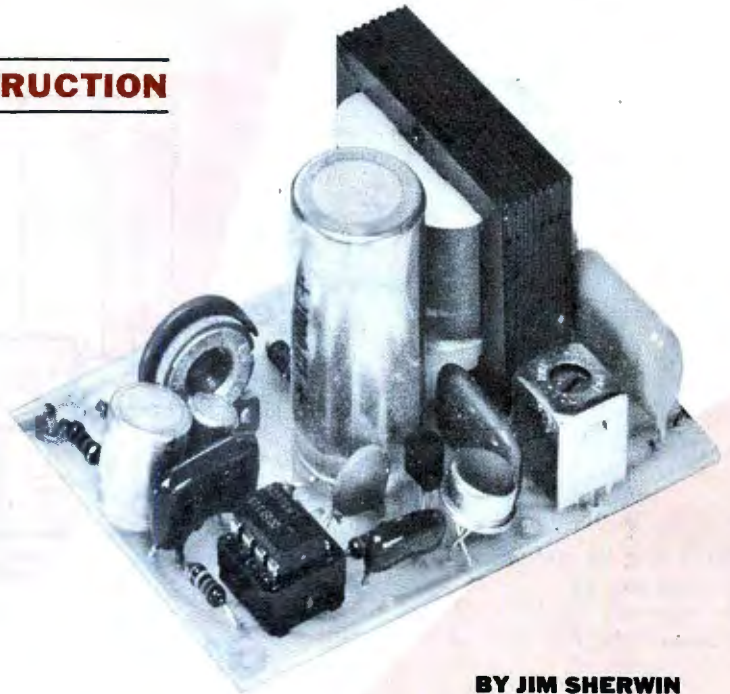


**C**ONNECTING extension speakers to one's stereo system or rear speakers for a four-channel setup is a terrifying task for many people. It often means running speaker wires through walls and floors, hiding them under a carpet, or tacking them to baseboards. Presented here is a "wireless" system that eliminates this problem!

The new system uses existing ac power lines, operating in an FM mode. The result is good-quality audio (wide frequency response and low distortion) and a high order of noise immunity.

To overcome the old obstacles of the wireless medium (noise, narrow bandwidth, etc.), as in AM wireless intercoms, the Wireless Hi-Fi System employs an FM carrier, phase-locked-loop (PLL) and voltage-controlled-oscillator (vco) IC's. The result claimed by the author is a 30-to-17,000-Hz  $\pm 0.2$  dB frequency response, 2% total harmonic distortion (THD), and excellent noise immunity for a fine signal-to-noise (S/N) ratio of -50 dB unweighted, at 2 watts power output into an 8-ohm speaker. (How to achieve higher power output is discussed later in this article.) In addition, putting the signal on the ac line will generally assure good reception at all power outlets in your home—even if the transmitter and receiver are on *opposite* sides of the line.

## CONSTRUCTION



BY JIM SHERWIN

# A Wireless Audio System for Remote Speakers

*Now you can enjoy high-quality remote sound without running wires around the house.*

As structured, the system will accommodate a monophonic signal. To transmit stereo, two systems can be used, each tuned to a different carrier frequency. The drive signals can be taken from the tape monitor and tape output jacks of your amplifier. If your hi-fi system doesn't have these jacks, the signal can be taken from across the speaker terminals, but the advantage of a fixed level will be lost.

**About the Circuit.** The transmitter, shown schematically in Fig. 1, uses voltage-controlled oscillator IC1 to drive the Q1 amplifier. The dc power bus is regulated by IC2.

Two inputs are provided so that both left and right channels from a stereo system can be combined for transmission to a remote speaker system. For a monophonic source, resistors R7 and R8 can be eliminated and



Fig. 1. Two inputs are provided for the transmitter. For a monophonic source, R7 and R8 can be eliminated.

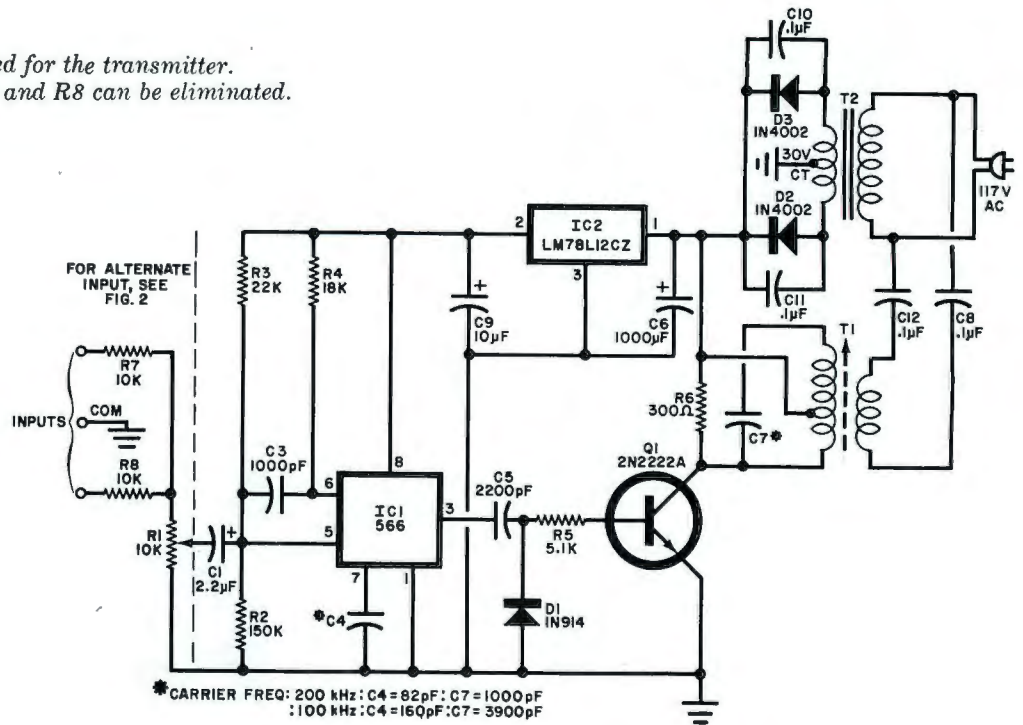
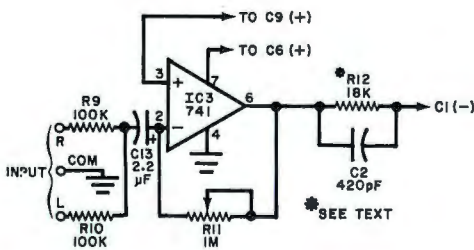


Fig. 2. Use the circuit below to provide pre-emphasis for the transmitter in Fig. 1.



POPULAR ELECTRONICS' performance tests verified that the S/N ratio measured  $-65$  dB, exceeding the author's claimed  $-50$  dB. With "A" weighting it was  $-93$  dB. At  $0.9$  W, distortion was  $0.98\%$ ; at  $2.5$  W, it was  $4.9\%$ . Frequency response was  $30$  to  $17,000$  Hz  $\pm 0.5$  dB—Ed.

the input signal fed directly to the top of potentiometer R1. If you want to use the system to feed one channel from a stereo system to the other side of the room, for example, omit R8 and install a  $470$ -pF capacitor across R7. This will improve the frequency response so that it goes out to about  $20,000$  Hz. The alternate input circuit shown in Fig. 2 should be used in either of two situations. First, if you are tapping the signal(s) from a high-impedance source, the  $10,000$ -ohm resistor(s) could load down the driving signal in a mono circuit, or introduce unacceptable crosstalk between the two channels of a stereo system (as applied to the main power amplifier). If so, the circuit of Fig. 2 will provide the higher

**TRANSMITTER PARTS LIST**

- C1, C13\*— $2.2$ - $\mu$ F, 50-volt electrolytic capacitor
- C2\*— $420$ -pF mica capacitor
- C3— $1000$ -pF disc capacitor
- C4—See Fig. 1 for value
- C5— $2200$ -pF disc capacitor
- C6— $1000$ - $\mu$ F, 25-volt electrolytic capacitor
- C7—See Fig. 1 for value
- C8, C10, C11, C12— $0.1$ - $\mu$ F, 200-volt film capacitor
- C9— $10$ - $\mu$ F, 16-volt electrolytic capacitor
- D1—1N914 silicon diode
- D2, D3—1N4002 rectifier diode
- IC1—566C voltage-controlled oscillator integrated circuit
- IC2—78L12CZ—12-volt regulator integrated circuit
- IC3\*—741C operational amplifier integrated circuit.

- Q1—2N2222A transistor
- R1— $10,000$ -ohm pc-type potentiometer
- R11\*— $1$ -megohm pc-type potentiometer
- The following resistors are  $\frac{1}{2}$  watt:
- R2— $150,000$  ohms,  $10\%$
- R3— $22,000$  ohms,  $10\%$
- R4, R12\*— $18,000$  ohms,  $5\%$
- R5— $5100$  ohms,  $10\%$
- R6— $300$  ohms,  $5\%$
- R7, R8— $10,000$  ohms,  $10\%$
- R9\*, R10\*— $100,000$  ohms,  $10\%$
- T1—Line-coupling transformer (see text)
- T2—Power transformer with 30-volt center-tapped, 50-mA secondary (Triad No. F133P or similar)
- Misc.—Printed circuit board; line cord; machine hardware; hookup wire; solder; etc.
- \*Optional preemphasis circuit components.
- Note—See Fig. 3 Parts List for availability of parts.

degree of isolation required. The parallel combination R12-C2 will smooth out the frequency response of the system. To make up for losses in the RC combination, gain is provided (between unity and 10) by op amp IC1. Gain is controlled by feedback resistor R11. Because the system's noise immunity is good, no preemphasis/deemphasis is required. However, if you wish to experiment, the alternate input circuit should be used, with a higher RC time constant. For standard  $75$ - $\mu$ s preemphasis, R12 should be changed to  $180,000$  ohms. Then, all program material above  $2120$  Hz will be emphasized before it enters the transmitter. As above, op amp IC1 provides gain (set by R11) to balance out losses in the preemphasis net-

work. Of course, if preemphasis is introduced in the transmitter, deemphasis must then be incorporated into the receiver. The free-running frequency,  $f_c$ , of voltage-controlled oscillator IC1 in Fig. 1 is determined by the values of R4 and C4. The sensitivity of the vco with the biasing shown and a  $V_{cc}$  of  $+12$  volts is about  $\pm 0.66f_c/V$ . For minimum distortion, the deviation should be limited to  $\pm 10\%$ , which means that the maximum input signal level at pin 5 of IC1 should be  $0.3$  volt peak-to-peak. The output voltage from the tape monitor or tape output jack of an audio system may range from  $0.4$  to  $1.5$  volts rms; therefore, R1 is provided so that an appropriate modulation level can be set. The frequency-modulated output at

pin 3 of *IC1* is roughly a 6-volt peak-to-peak square wave. This signal is used to modulate the *Q1* r-f oscillator, which uses tuned transformer *T1* as its collector load. Because *T1* is tuned to *f<sub>c</sub>*, by adjusting its slug and *C7*, it serves as a high-impedance collector load, which eliminates the need for additional current limiting for *Q1*. Because the collector signal can have as much as a 50-volt peak-to-peak amplitude, *Q1*'s breakdown point must be high. The transistor specified for *Q1* in the Parts List has a 60-volt breakdown to provide a margin of safety. The modulated r-f output from the transmitter

is coupled to the power line via *C8* and *C12*.

Note that Fig. 1 specifies values for *C4* and *C7* for either 100- or 200-kHz operation. Also note that capacitors *C10* and *C11* are connected across rectifier diodes *D2* and *D3*. These capacitors reduce the small step transient that might be present across the silicon diodes whose upper harmonics fall within the frequency of interest.

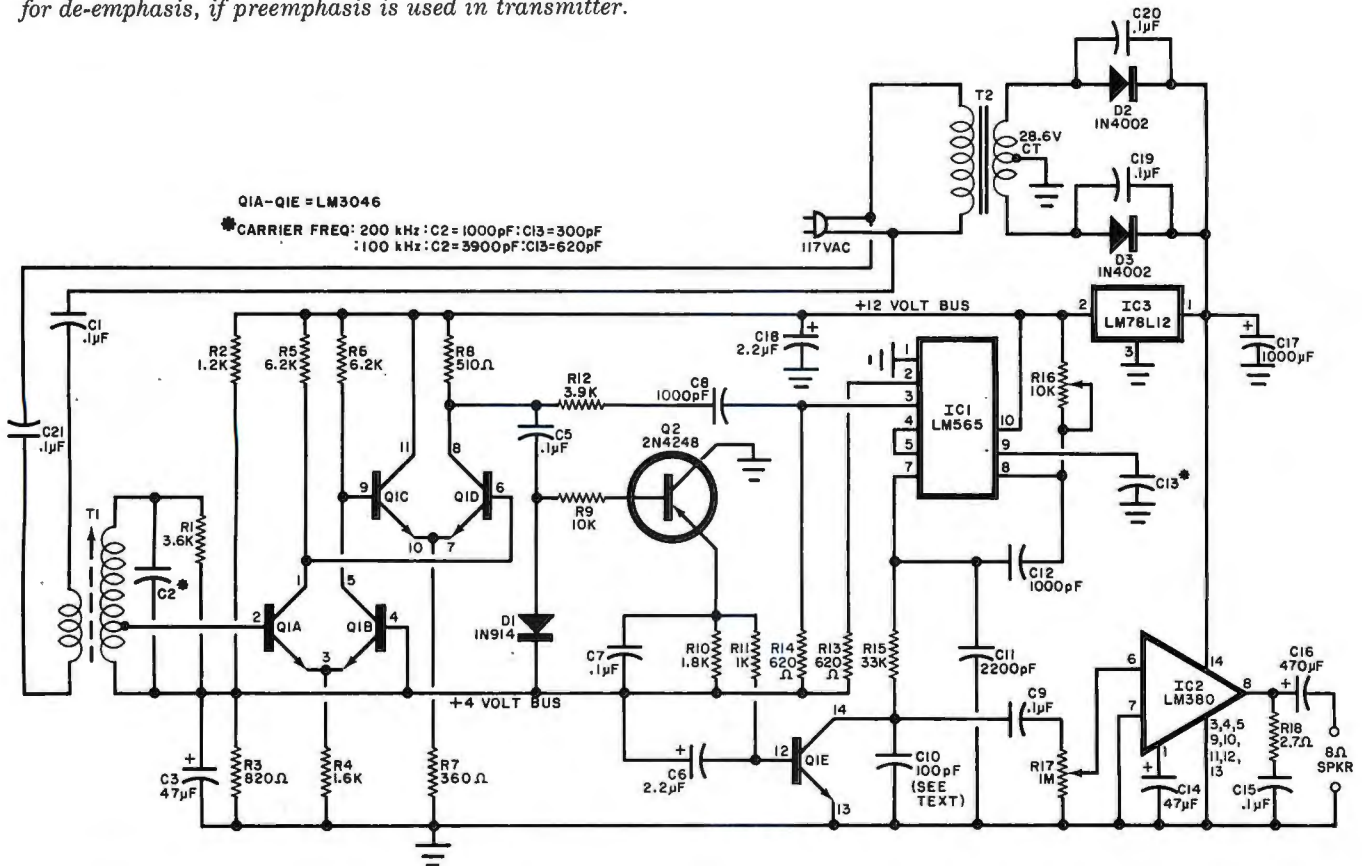
The job of the receiver is to amplify, limit, and demodulate the received FM signal riding on the ac power line which can have high-amplitude noise

transients. In addition, the receiver must also provide an audio mute in the absence of a carrier.

The circuit shown in Fig. 3 picks up the incoming FM signal and tunes to the carrier frequency via the *T1/C2* circuit. This signal is then applied to a two-stage limiting amplifier (composed of elements inside transistor array *Q1*), phase-locked loop *IC1*, audio amplifier *IC2*, and a mute circuit made up of *Q1E* and discrete transistor *Q2*.

The FM carrier is coupled to the primary of tuned transformer *T1* through *C1* and *C21*. The secondary of

Fig. 3. In the receiver, the value of *C10* must be increased for de-emphasis, if preemphasis is used in transmitter.



## RECEIVER PARTS LIST

C1, C9, C19, C20, C21—0.1- $\mu$ F, 200-volt film capacitor  
 C2—See Fig. 3 for value  
 C3, C14—47- $\mu$ F, 16-volt electrolytic capacitor  
 C4—Not used  
 C5, C7, C15—0.1- $\mu$ F, 25-volt disc capacitor  
 C6, C18—2.2- $\mu$ F, 25-volt electrolytic capacitor  
 C8, C12—1000-pF disc capacitor  
 C10—100-pF disc capacitor (see text)  
 C11—2200-pF disc capacitor  
 C13—See Fig. 3 for value  
 C16—470- $\mu$ F, 25-volt electrolytic capacitor  
 C17—1000- $\mu$ F, 25-volt electrolytic capacitor  
 D1—1N914 silicon diode  
 D2, D3—1N4002 rectifier diode  
 IC1—565C phase-locked loop integrated circuit

IC2—LM380 audio amplifier integrated circuit (National)  
 IC3—78L12CZ 12-volt regulator integrated circuit

Q1 (A to E)—LM3046 transistor array (National)  
 Q2—2N4248 transistor.

The following resistors are 1/2 watt:

R1—3600 ohms, 5%  
 R2—1200 ohms, 5%  
 R3—820 ohms, 5%  
 R4—1600 ohms, 5%  
 R5, R6—6200 ohms, 5%  
 R7—360 ohms, 5%  
 R8—510 ohms, 5%  
 R9—10,000 ohms, 10%  
 R10—1800 ohms, 5%  
 R11—1000 ohms, 5%  
 R12—3900 ohms, 10%  
 R13, R14—620 ohms, 5%  
 R15—33,000 ohms, 10%  
 R18—2.7 ohms, 10%  
 R16—10,000-ohm pc-type potentiometer

R17—1-megohm potentiometer  
 T1—Line-coupling transformer (see text)  
 T2—Power transformer with 26.8-volt center-tapped, 1-ampere secondary (Triad No. F40X or similar)  
 Misc.—Printed circuit board; line cord; machine hardware; hookup wire; solder; etc.

Note—The following items are available from American Scientific Corp., 1957 Old Middlefield Rd., Mountain View, CA 94043: etched and drilled transmitter pc board for \$3; etched and drilled receiver pc board for \$5.50; transmitter IC's and transistors and Toko line-coupling transformer for \$6.00; receiver IC's and transistors and Toko transformer for \$10.50; complete transmitter kit for \$18.95, plus \$1.00 shipping; complete electronics receiver kit for \$27.95, plus \$1.00 shipping. National Semiconductor IC's are available, singly or in quantity from OEMorsco, 2403 Charleston Rd., Mountain View, CA 94043.

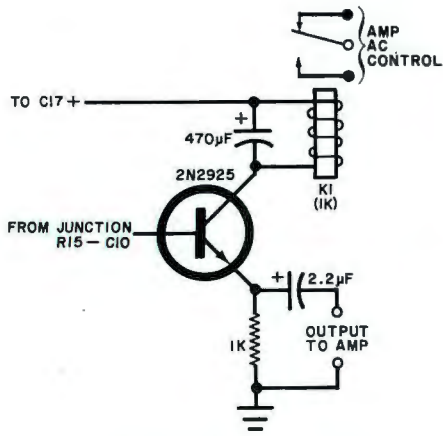


Fig. 4. Use this circuit to change receiver output for more power.

T1 is tuned by C2, while R1 lowers the Q of the circuit to permit it to accept the  $\pm 10\%$  modulation and to prevent excessive ringing on noise spikes. Such ringing would cause the mute circuit to deactivate. The secondary of T1 is tapped to match the input impedance of Q1A. The recovered carrier at the secondary of T1 can have an amplitude of between 0.2 and 45 volts peak-to-peak. The 17:1 turns ratio from the full secondary to the tap reduces the amplitude of the recovered carrier so that the base of Q1A "sees" a signal that ranges from 12 mV to 2.6 volts peak-to-peak.

The two-stage limiter amplifier composed of Q1A through Q1D operates as a fairly fast comparator whose slew rate is about 70 volts/ $\mu$ s and gain is about 3000. The output from this limiter is a roughly 7-volt peak-to-peak square wave that has rise and fall times of 100 ns. (Any fairly fast comparator that delivers more than 5 volts output could be used as the limiting amplifier. However, the transistor array specified is a low-cost item that has the advantage of containing an extra-high beta transistor, Q1E, which is used as the mute switch.) The 7-volt peak-to-peak output of the limiting

amplifier is applied directly to the mute detector, Q2, but is reduced to about 1 volt by the attenuator consisting of R12 and R14 before being applied to PLL IC1.

The PLL operates as a narrow-band filter that tracks the incoming FM signal and provides a low-distortion demodulated audio output whose signal-to-noise (S/N) ratio is high. The oscillator inside the PLL is set to free-run at a frequency near that of the car-

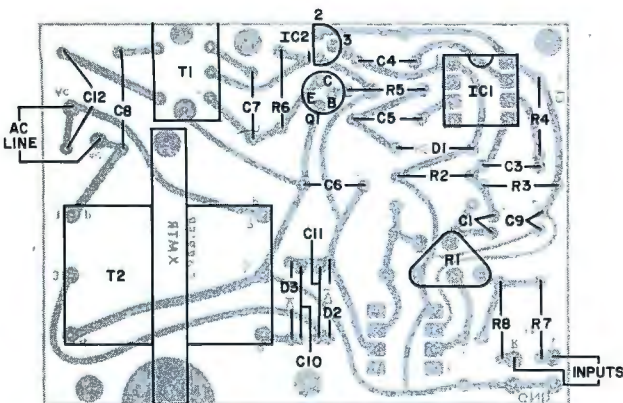
rier by the choice of value for C13 and setting of R16.

If 75-microsecond preemphasis is used, increase the value of C10 to 2200 pF. Increasing this capacitor's value will reduce the carrier level in the audio line at the expense of high-frequency audio response.

The mute circuit quiets the receiver when no carrier is present. Without this circuit, an excessive noise level would result as the PLL attempts to

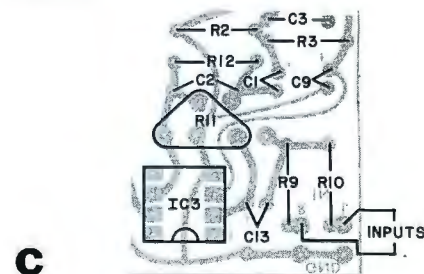


A



B

Fig. 5. Etching and drilling guide for the transmitter are shown above (A). Component placement for standard transmitter is at left (B). If pre-emphasis is used, the component placement at right is necessary (c).



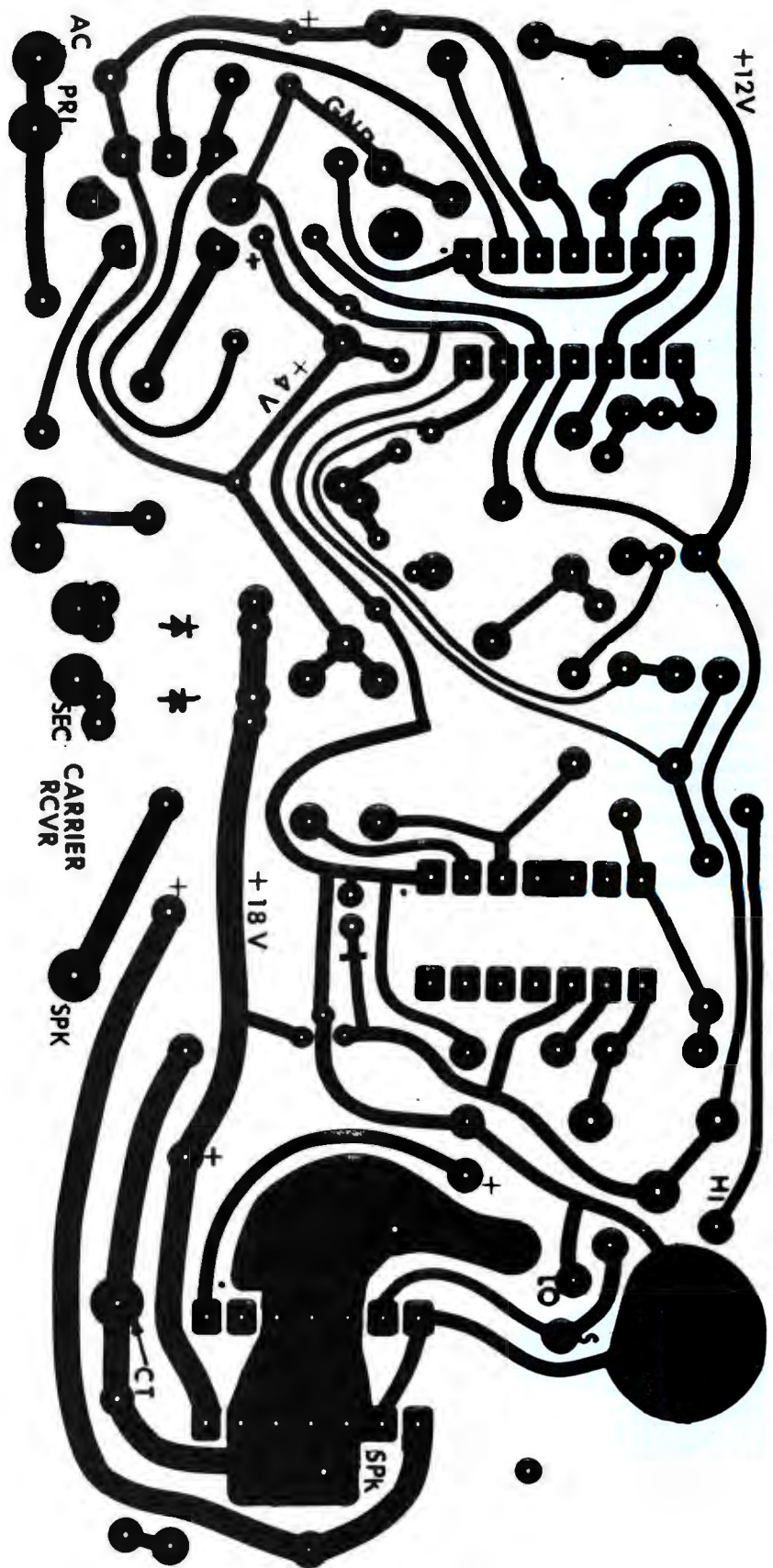
C

lock onto noise spikes when the transmitter is shut off. The mute detector is composed of  $C7$ ,  $D1$ , and  $Q2$ . Note that an emitter-follower transistor stage is used in place of a diode to give the peak detector a high input and a low output impedance. This permits the peak detector to shunt 1 to 2 mA of bias away from  $Q1E$ , without loading the limiter amplifier.

With no carrier present, the +4-volt line biases  $Q1E$  into conduction via  $R10$  and  $R11$ , shorting the audio signal to ground. When a carrier is present, the 7-volt square wave from the limiter amplifier is peak detected, producing a negative output that is integrated by  $R9$  and  $C7$ , averaged by  $R10$  across  $C7$ , and further integrated by  $C6$  and  $R11$ . The resultant -4-volt output subtracts from the +4-volt bias to deprive  $Q1E$  of base current. This transistor cuts off and allows the demodulated audio signal to pass to the audio amplifier. Peak detector integration and averaging prevents random noise spikes from deactivating the mute when no carrier is present.

Audio amplifier  $IC2$  supplies about 2 watts of audio power to an 8-ohm speaker connected across the output of the receiver. Although this amount of power is adequate for casual and background-music listening, if you want to drive relatively inefficient hi-fi speaker systems, you'll have to modify the output of the receiver to drive a hi-fi amplifier. The circuit shown in Fig. 4 does this. To use it, audio power amplifier  $IC2$  is removed from the receiver circuit and the Fig. 4 circuit goes into the receiver as specified. This switching circuit operates from the mute signal so that when the transmitter is switched on, the relay will energize to apply power to the external power amplifier. In this manner, the power amplifier will automatically switch on when a signal is received. Standby power is less than 1 watt.

**Construction.** The transmitter is best assembled on a printed circuit board, the etching and drilling guide for which is shown actual size in Fig. 5A. In Fig. 5B is the component placement diagram for the board, while the detail in Fig. 5C shows the additions and changes to be made when preemphasis is used in the transmitter. When comparing B and C, note that  $R1$  is eliminated from the former and  $R11$  is installed in the latter, displaced one pad space to the left but oriented the same.



**A**  
 Fig. 6A. Etching and drilling guide for the receiver is shown above. The component placement diagram (Fig. 6B) is on next page.

# THE SILENCER

Your ears are burning with amplified noise. Even though your system is delivering sound accurately, it's also doing an efficient job of pumping out noise . . . accurately. Ideally, music should be recreated against a dead silent background. The Phase Linear 1000 accomplishes just that with two unique systems: The Auto Correlator Noise Reduction and the Dynamic Range Recovery Systems.

★ It improves the overall effective dynamic range and signal/noise ratio 17.5 dB in any stereo system with any stereo source.

★ The Auto Correlator reduces hiss and noise 10 dB without the loss of high frequencies and without pre-encoding.

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★ Plus, it removes hum, rumble and low frequency noises, without the loss of low frequency music.

★ WARRANTY: 3 years, parts and labor

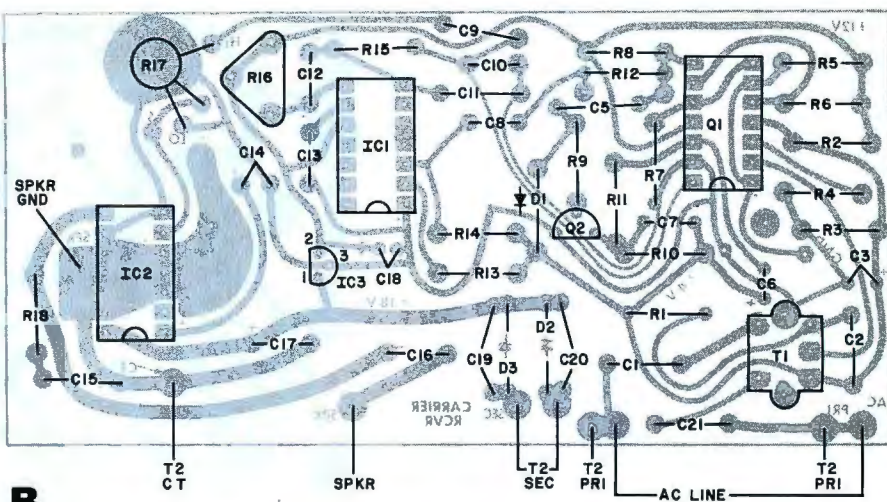
Even the finest stereo systems are limited in performance by the quality and nature of the recording. With the Phase Linear 1000, these limitations are overcome. Added to any receiver or preamplifier, it gives you the most significant improvement in sound reproduction for the money . . . more than any other single piece of equipment you could add to your system. Ask your dealer for an audition. The silence is deafening.



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**B**

Fig. 6B. Component placement diagram for the receiver is shown above. Foil pattern (Fig. 6A) is on previous page.

Line-coupling transformer *T1* can be made from a 455-kHz AM i-f transformer. In rewinding the turns, the secondary is made up of four turns, while 154 turns are used in the primary, tapped 29 turns from one end of the coil. You can use the wire originally wound on the transformer or 41 AWG enameled wire for the turns.

As with the transmitter, the receiver is best assembled on a printed circuit board. The etching and drilling guide and components placement diagram are shown in Fig. 6A & B.

Line-coupling transformer *T1* is constructed in the same manner as described for *T1* in the transmitter, except that primary is composed of 4 turns the secondary 154 turns, tapped 9 turns from one end.

**Adjustments.** Make sure that the transmitter and receiver are operating at the same frequency. (See Figs. 1 and 3 for values of frequency-determining components.) Plug the transmitter and receiver into ac outlets and use an oscilloscope or ac voltmeter to check the amplitude of the waveform across the secondary of *T1* in the receiver. There is no need at this time to have the carrier modulated.

Adjust the slugs in the *T1*'s in both the transmitter and the receiver for maximum signal amplitude across the secondary of *T1* in the receiver. If the receiver is to be located some distance from the transmitter or on the opposite side of a 117/220-volt service line, *T1* will have to be readjusted in the receiver to maximize rejection of ac line noise.

The free-running frequency of the PLL in the receiver is trimmed by adjusting *R16*, which should initially be

set near the center of its range. Rotate *R16*'s wiper slowly in both directions until the PLL loses lock, indicated by a sharp increase in noise and a distorted output from the receiver. Note the two positions where drop-out occurs; center *R16* between these two points. A finer adjustment may be required to obtain minimum noise if an SCR light dimmer is in operation.

The final adjustment is for modulation amplitude in the transmitter. Connect an audio signal source to the input of the transmitter and adjust *R1* for a signal amplitude of about 0.1 volt rms maximum at the input (pin 5) of *IC1*. This can also be accomplished by carefully listening to the receiver while adjusting *R1* in the transmitter until the sound just distorts and then backing off a little.

Because a single transmitter/receiver system will suffice for a monophonic sound system or for one channel of a stereo system, a second receiver and transmitter will be required for stereo sound over the wireless medium. In this case, one transmit/receive system would be tuned to 100 kHz and the other to 200 kHz. You can also use a pair of transmit/receive systems for the rear channels in a 4-channel system to avoid running cables all over your listening room. Again, you would use a different carrier frequency for the different channels.

One final note: Although we have mentioned using carrier frequencies of 100 kHz and 200 kHz for a two transmit/receive system, this spacing needn't be 100 kHz apart. For most purposes, a 40-kHz spacing will do, while still providing maximum frequency response from the system. ♦