### Engineer's notebook

## Discriminator computes frequency differential

by T. J. John

Meerut, Uttar Pradesh, India

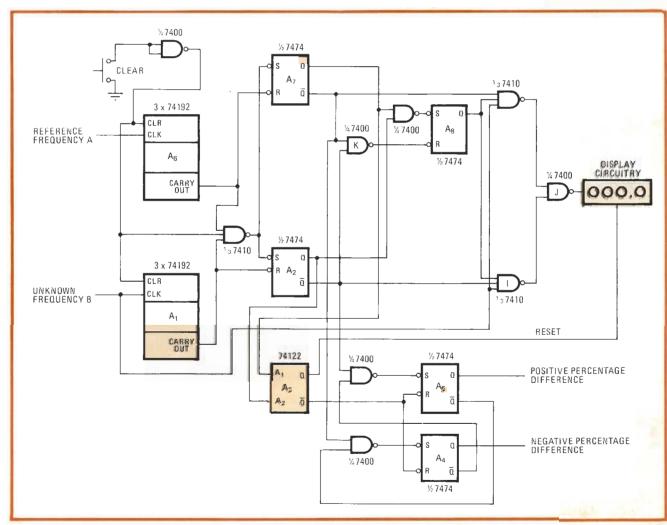
This circuit computes the percent difference of two frequencies to 0.1% without needing an accurate, crystal-controlled timebase. It is useful in industrial applications that convert a process variable into a train of pulses in order to monitor that unknown quantity.

The circuit finds  $\pm d = 100(p_x - p_r)/p_r$ , where d is the percent difference, and  $p_x$  and  $p_r$  are the number of pulses counted from an unknown and a reference source, respectively, in a given time. By making  $p_r$  equal to a decade multiplier of  $10^n$  for any general n, the first

equation becomes  $\pm d = (p_x - 10^n)(10^{2-n})$ , where  $p_x - 10^n$  is the difference between the number of pulses generated by the unknown source and the  $10^n$  pulses generated by the reference, and  $10^{2-n}$  indicates the position of the decimal point. Thus the equation is reduced to a form where it can be solved in hardware with off-the-shelf logic elements.

Note that for an accuracy of 0.1%, the value of n must be 3. Therefore, three cascaded decade counters must be used at each input for counting up to  $10^3 = 1,000$ , as shown in the figure.

A system clear initializes the counters to 0 and presets several circuit flip-flops. The carry output of the last 74192 in each chain then generates a negative-going pulse each time it counts to 1,000. Assuming a positive difference between unknown frequency B and reference frequency A,  $A_1$  reaches 1,000 first and resets flip-flop  $A_2$ .  $A_2$ 's  $\overline{Q}$  output then fires one-shot  $A_3$ , whose on-time must be small compared to the period of the unknown



**Comparison.** The circuit for finding  $\pm$  (A – B)/A, in percent, needs no frequency standard and it can be built with standard logic elements. Results are directly displayed. Accuracy of measurement is 0.1%, obtained by using three cascaded decade counters at each input.

frequency measured. Flip-flops A<sub>4</sub> and A<sub>5</sub>, the polaritydifference indicators, are then reset.

The preceding actions permit the 1,001st pulse

counted by A<sub>1</sub> to be passed directly from the input to the display-counting circuit through enabled gates I and J. For each pulse appearing at the output of J, the counter is advanced by 1, starting from the least significant bit displayed. When the reference-frequency counter, A6. reaches 1,000, A7 is reset. A8 in turn is reset through gate K and gate I is disabled. At that time, the display will directly indicate the magnitude of the difference between frequencies A and B, in percent. Because A<sub>1</sub> reaches 1,000 before A<sub>2</sub>, A<sub>5</sub> is set and will indicate there

exists a positive frequency difference between B and A. In the event a negative percentage difference is measured, A2 reaches 1,000 first. A sequence of events similar to those discussed previously then occurs, but with the upper portion of the circuit becoming active first, and with the end result that the negative-difference

indicator, A4, is set.

# One-shot/flip-flop pairs detect frequency bands

by Edward E. Pearson Opelousas, La.

A retriggerable monostable multivibrator and a type D flip-flop can form a simple reliable frequency comparator that senses if an input frequency is greater than or less than a predetermined reference. Connecting additional comparators in parallel, together with AND logic, permits the detection of input frequencies that fall within selected bands.

Both the one-shot and the flip-flop are wired for positive edge triggering. Each input pulse causes the monostable's output to go high for the period of its preset timing interval. The flip-flop is triggered simultaneously, but its output is determined by the state of its D input at the time of trigger threshold.

If the period of the input frequency is shorter than the preset timing of the monostable, a constant high level will be present at the D input, forcing the flip-flop's Q output to remain high. If the input frequency period becomes greater than that of the monostable, the D input will go low prior to the next incoming trigger. The flip-flop's Q output then goes low and remains low

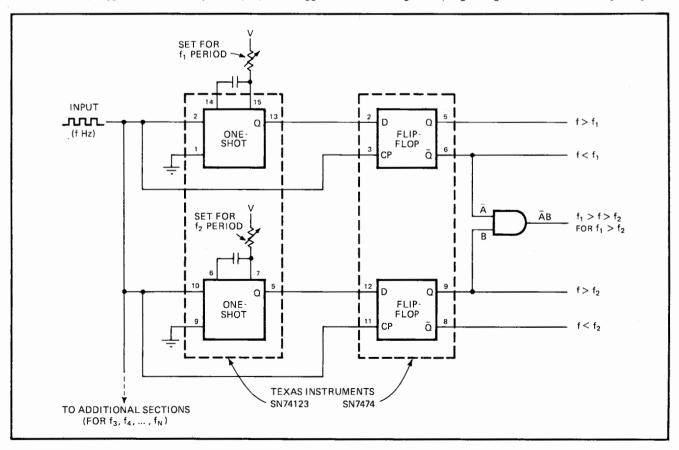
until the input period becomes shorter than that of the monostable.

To determine whether an input frequency (f) falls between two known frequencies,  $f_1$  and  $f_2$ , two one-shot/flip-flop combinations are required, as shown. The top pair of devices detects an input greater or less than  $f_1$ , while the bottom pair detects an input greater or less than  $f_2$ . The AND gate provides a high output when the input frequency lies inside the preset band (less than  $f_1$  or greater than  $f_2$ , if  $f_1$  is greater than  $f_2$ ). This detection scheme can be expanded to include any desired number of segments within the operating passband.

The frequency band detector also has an inherent memory function that could be particularly useful in control applications. When the input signal terminates, for example, with a tone burst, no trigger is available to the flip-flops, and all outputs remain static until the input signal returns.

Although the detector responds only to the period of the input signal and does not require the input to maintain a specific duty cycle, input pulses must have a rapid rise time. All trigger thresholds must be reached within an interval that is appreciably less than the monostable's propagation delay time. Circuit speed is limited only by the setup and hold performance of the components being used.

**Sensing frequency.** Retriggerable one-shot and flip-flop compare frequency of input to preset reference frequency. To form frequency-band detector, two frequency comparators and AND gate are needed. Depending on period of input pulse train, each one-shot output is high or low. Each flip-flop triggers to level seen by its D input prior to trigger threshold. AND gate output goes high when f falls between f<sub>1</sub> and f<sub>2</sub>.



## In-range frequency detector has jitter-free response

by A. J. Nicoll Instromedix Inc., Beaverton, Ore.

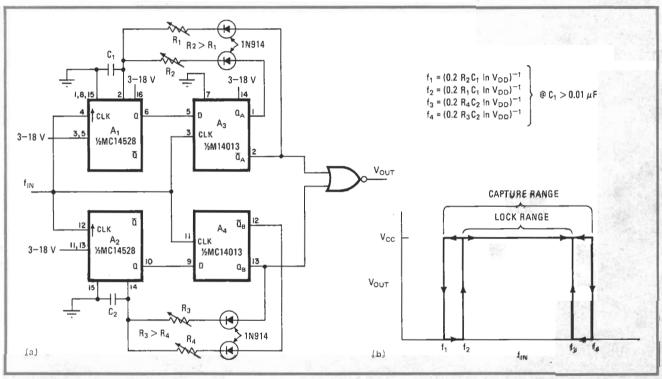
This simple circuit will detect when an input signal falls within a specified frequency range and is thus ideal for use as an out-of-tolerance alarm or as a rudimentary phase-locked loop. It could also be called unusual, since it uses hysteresis to provide separate lock and capture ranges that eliminate the jitter of the circuit's logic-level output.

The diagram shown in (a) and the hysteresis curve shown in (b) help make the circuit's operation clear.  $A_1$ 

and  $A_2$  are two retriggerable one-shots. Their pulse widths, and therefore their maximum frequency of operation, are controlled by  $R_1$ - $R_4$ . Whether  $R_1$  or  $R_2$  controls the width of  $A_1$  and  $R_3$  or  $R_4$  controls the width of  $A_2$  depends upon the state of the  $A_3$  or  $A_4$  D-type flip-flops.

Assume  $R_1$  and  $R_4$  are the controlling elements as an input signal of arbitrary frequency,  $f_{\rm in}$ , arrives to trigger both one-shots simultaneously. The positive transition of  $f_{\rm in}$  then fires  $A_1$  and  $A_2$ , as shown. The next positive-going transition will trigger both  $A_1$  and  $A_2$  again while clocking the previous output states, which were generated before retriggering, into  $A_3$  and  $A_4$ .

If this second transition occurs before either one-shot has returned to its time-out state, a logic 1 will be clocked into its respective flip-flop, changing the state of that flip-flop. Once the flip-flop moves from a 0 to a 1, the pulse width of the one-shot will be controlled by one



Within limits. Circuit (a) detects whether input signal is within user-set frequency range  $f_2$ – $f_3$  (b). Flip-flops enable selectable hysteresis so that circuit, once locked, will not change state until  $f_{in}$  moves below  $f_1$  or moves above  $f_4$ . Lock and capture ranges are controlled by  $R_1$ – $R_4$ . Hysteresis eliminates jitter that would normally occur at output if  $f_{in}$  were near  $f_2$ 's or  $f_3$ 's edges.

120

Electronics/June 8, 1978

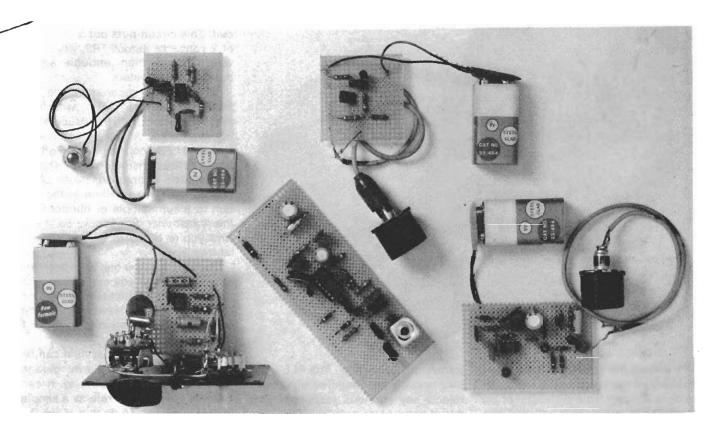
of the two timing elements, R<sub>2</sub> and R<sub>3</sub>.

The curve (b) shows more clearly how the lock and capture ranges are controlled by  $R_1$ – $R_4$ , where  $f_1$ – $f_4$  are equal to the reciprocals of the pulse widths determined by  $C_1$ – $R_1$  or  $-R_2$  and  $C_2$ – $R_3$  or  $-R_4$ .  $A_3$  will move high when  $f_{in}$  rises above  $f_2$ , and it will not move back to its initial state until  $f_{in}$  falls below  $f_1$ . Similarly,  $A_4$  will change from a 0 to a 1 when  $f_{in}$  rises above  $f_4$ , and it will change back to a 0 only when  $f_{in}$  falls below  $f_3$ . The amount of hysteresis acting upon  $f_1$ – $f_2$  and  $f_3$ – $f_4$  can be

chosen by simply selecting the appropriate resistance values for  $R_1-R_4$ .

The NOR gate output moves high when  $f_{in}$  is within the set limits of  $f_2-f_3$ . It will not move low again until the input frequency falls below  $f_1$  or above  $f_4$ . If desired, an OR gate can be used instead of a NOR gate, since both the Q and  $\overline{Q}$  outputs are available in D-type flip-flops.  $\square$ 

Designer's casebook is a regular feature in *Electronics*. We invite readers to submit original and unpublished circuit ideas and solutions to design problems. Explain briefly but thoroughly the circuit's operating principle and purpose. We'll pay \$50 for each item published.



# EXPERIMENTING WITH PHASE-LOCKED LOOPS

Four simple but useful circuits that will increase your understanding of these versatile 565 and 567 IC's.

BY HERB COHEN

HEN phase-locked-loop (PLL) integrated circuits began to appear on the hobby market, the experimenter was faced with the same problem always encountered with new devices: what to do with it. If he tried to understand the theory—usually available in abundance—he was clouted with terms like rads/second, capture ratio, lock range, and lag networks. Needless to say, there was an urge to slam shut the book and treat the IC as a "black box"—not really such a bad idea.

Although there have been a number of articles published in various magazines explaining the basics of PLL theory (see "How Phase-Locked Loops Work," February 1975), a "hands-on" session with these IC's will tell you more about them than all the reading you're likely to do. As an example, the 565 and 567 PLL's are so simple to work with and require so few external components that you don't

need a stage-by-stage understanding of what goes on inside them to put them to use.

In this article, we describe four simple projects you can build around a pair of commonly available PLL's to give a "feel" for how they perform. Each project illustrates a different aspect of phase-locked-loop technology, and each is a practical circuit you can put to immediate use. Before proceeding to the projects themselves, however, let us first discuss the specific PLL's used in our projects.

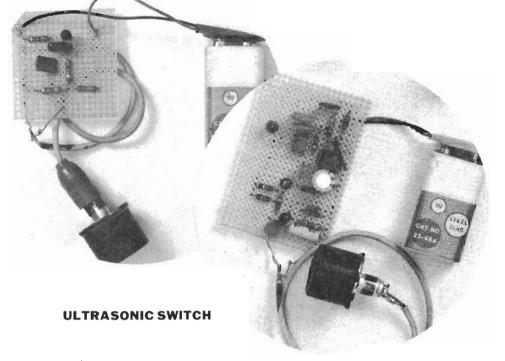
**Two PLL's.** Of the many phase-locked-loop IC's available, the 567 is the only one that is designed primarily for switching applications. When this PLL goes into lock, its output transistor, driven from a quadrature detector, is capable of passing 100 mA of current. This makes the 567 ideal as an SCR or relay driver.

Another unique feature of the 567

PLL is its ability to be driven from a low-voltage (4.5 to 10 volts) power supply at a nominal 10-mA drain. This means that the IC can be powered by a 9-volt transistor battery with good life expectancy. So, although it doesn't have the bandwidth or sensitivity of other PLL's, the 567 is ideal for hundreds of tone-decoder applications in the range from 1 Hz to 100 kHz.

The 565 is a general-purpose PLL IC and is by far the most popular now being used. It exhibits a very wide ±60% locking range and a 1-mV input sensitivity. This PLL is ideal for use as an SCA decoder, which will let you receive the hidden subchannels on FM.

The following four projects are examples of the simplicity and versatility of the PLL IC. The first three are built around the 567, while the fourth—an SCA decoder—employs the 565 PLL. The circuits can be assembled easily on perforated boards or pc boards.



Ultrasonic Switch. The simple transmit/receive system shown in Fig. 1 can receive a signal from distances of up to 40' (12 m) and more in hallways and enclosed areas where the acoustical properties are good. The transmitter is shown in B, while the receiver is shown in A.

Transducers TR1 and TR2 in the receiver and transmitter are 40-kHz barium-titanate ultrasonic transducers. Transducer TR1 in the receiver picks up the sound waves from transmitter transducer TR2 and passes them to the amplifier consisting of transistors Q1 and Q2. The PLL (IC1) then accepts the amplified signal and rejects any spurious responses and out-of-band noise pulses.

Light-emitting diode LED1 and limiting resistor R9 are installed in the circuit only temporarily to assist in tuning the system. Once the system has been properly tuned, these components are removed and replaced with the load to be driven (relay, lamp, etc.).

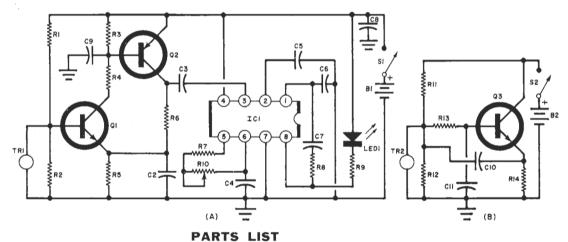
The transmitter shown in B is a Colpitts oscillator configuration that uses transducer TR2 in the resonance circuit. This circuit puts out a minimum of 2 volts rms across TR2, which develops more than enough sound power for the system.

To tune the receiver, place the transmitter about 5' (1.5 m) away from the receiver with both transducers facing each other. Adjust potentiometer R10 until LED1 comes on. Turn off the power to the transmitter; LED1 should immediately extinguish. (Note LED1 may light up when R10 tunes the system to a submultiple or harmonic of the transducer frequency, so make sure you're tuned to the fundamental frequency.)

Once the receiver has been properly tuned, the range you can obtain with this system is dependent mainly on room acoustics. However, you should be able to obtain a minimum of 20' (6.1 m) of range

The ultrasonic relay system can be used as an intruder alarm, garage door opener, or remote relay. It can even be made to operate as a simple motion detector. To do this, place the transmitter and receiver about 10' (3 m) apart and with their transducers facing the same wall in a room. The LED in the receiver should come on; if it doesn't, move the transmitter closer to the receiver until it does. Then, move it away until the LED just extinguishes. Now, walk along the side of the room that the transmitter and receiver are facing. As you move, the LED will blink on and off.

Fig. 1. A simple transmit/receive ultrasonic relay.



B1.B2-9-volt battery

C1,C6-0.04-µF disc capacitor

-5-µF, 10-volt electrolytic capacitor

C3,C4--0.0047-µF disc capacitor

C5—0.1-µF disc capacitor C7,C10—0.02-µF disc capacitor

C8—100-µF, 10-volt electrolytic capacitor C9—0.001-µF disc capacitor

C11—330-pF disc capacitor IC1—567 PLL IC

LED1—Light-emitting diode TR1,TR2—Ultrasonic transducer (Detectron No. HC1)

Q1-2N4946 pnp transitor

-2N4917 npn transistor

Q3—HEP S0007 npn transistor

The following are ¼-watt, 10% resistors:

R1—180,000 ohms

43,000 ohms R3,R4-2200 ohms R5, R13, R14-1000 ohms R8, R7-2700 ohms

R9—330 ohms

R11—22,000 ohms R12—47,000 ohms

R10-10.000-ohm miniature potentiome-

S1,S2—Spst switch

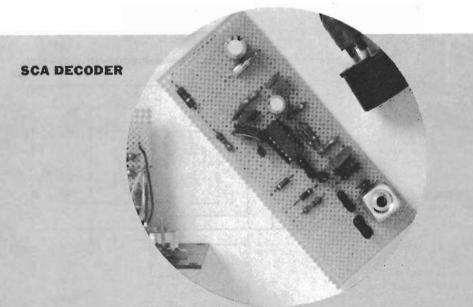
Misc.—Battery clips; hookup wire; sol-

der; etc.

shown in Fig. 3, is hard to beat. Just by touching the plate, body capacitance unbalances the circuit and pulls IC1 into lock. An external signal is normally fed into the PLL (IC1) via pin 3 and C1 and R1 are used to tune the voltage-controlled oscillator to the input frequency. When lock occurs, the vco adjusts itself to be 90° outof-phase with the input signal.

With the input signal at pin 3 coming from the output of the vco (pin 5 of IC1) through C4, R2, and R3, the two signals are in-phase with each other and the circuit cannot lock. However, when the touch plate is approached, enough capacitive phase shift is introduced to allow the circuit to lock.

Wiring the jumper into the circuit as shown allows the system to latch in the on position even after your hand is removed from the touch plate. To reset the system to off requires S1 to be closed momentarily.



SCA Decoder. Our final PLL project is an SCA decoder built around the 565 phase-locked-loop IC. This is es-

sentially a 67-kHz FM detector. However, a PLL is a better detector for FM than any of the traditional detector de-

R6

Fig. 4. SCA decoder.

#### **PARTS LIST**

-220-pF disc capacitor

C2—0.002-µF disc capacitor
C3—330-pF disc capacitor
C4—560-pF disc capacitor
C5.C9,C10—0.04-µF disc capacitor -0.1-µF disc capacitor

C7—0.001-µF disc capacitor C8—0.001-µF disc capacitor C11—30-µF, 15-volt electrolytic capacitor

-12-volt zener diode

-565 PLL IC -Phono jack

-10-mH slug-tuned inductor (Miller No. 9060)

Q1-2N2926 npn transistor

The following are 14-watt, 10% resistors:

-100,000 ohms -22,000 ohms

-8200 ohms -1500 ohms

-15,000 ohms

R6, R7, R11-4700 ohms

-6800 ohms

-1000 ohms R12-47,000 ohms

R10-10,000-ohm. linear-taper potentiometer

47,000-ohm, 1/2-watt. 10% resistor R13-

-Spst switch Misc .--Battery clip; hookup wire; solder; signs because it has the ability to dive 6 dB below the noise level and still lock onto a signal.

In the case of an SCA subchannel where the information is only 10% of the total program power (most of that lost in the audio filtering), the 565 IC's ability to reject noise is an important factor in building a simple and effective SCA decoder.

Capacitors C1, C2, and C3 and coil L1 (Fig. 4) form a bandpass filter that peaks at 67 kHz and rejects all lowfrequency components of the audio signal in an FM tuner. Transistor Q1 amplifies this signal and passes it to IC1. The PLLIC is tuned by C7, R6, and R10. Since the tuning frequency is also a function of the supply voltage, the IC should be zener-diode regulated.

The demodulated audio signal comes out of the decoder at a 50-mV level. It has a 7,000-Hz audio bandwidth that can hardly be considered hi-fi. This bandwidth, however, is more than sufficient for background music.

The tuning procedure is simple. Connect the output of your FM tuner to the input of the SCA decoder and the output of the decoder to your audio amplifier. Set R10 to the center of rotation. Scan the FM dial; all you should be able to hear at this point is noise and no stations. An SCA subchannel will appear as a sharp drop in the noise level, accompanied by a distorted music program. Now, adjust R10 for the best signal-to-noise (S/N) ratio and highest fidelity.

Tune to the weakest SCA subchannel you can find. Adjust L1 for the lowest possible noise level. The SCA decoder is now ready to use.

Closing Comment. The preceding four projects illustrate only a small portion of the possible applications to which the versatile phase-locked-loop IC can be put. A couple of the projects should be able to suggest other projects of your own.



By Forrest M. Mims

#### THE 567 TONE DECODER

HERE ARE many applications for circuits that can be made to respond to a specific tone frequency while ignoring all others. Some of these include radio controlled garage door openers and model airplanes, automatic paging systems, intrusion alarms, communications systems, and highly secure electronic locks.

Over the years, several practical circuits which respond to a specific tone frequency have been designed. Two of the best known are the resonant reed relay, once very popular with radio control modelers, and narrow bandpass active filters using one or more operational amplifiers.

The arrival of sophisticated single chip phase-locked loops (PLL's) a few years ago has made possible a variety of versatile tone-detection circuits, the 567 being designed especially for this role. Supplied in either an 8-pin mini-DIP or TO-5 can, the 567 is available from several sources listed in the Electronics Market-place in this magazine for less than \$1.75.

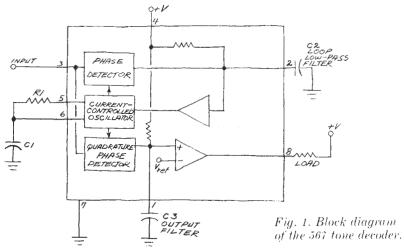
The functional block diagram of the 567 tone decoder is shown in Fig. 1. The circuit incorporates 62 transistors to provide synchronous AM lock detection and a power output stage. In operation, a current controlled oscillator (cco) operates at a frequency determined by external components R1 and C1. This frequency is called the

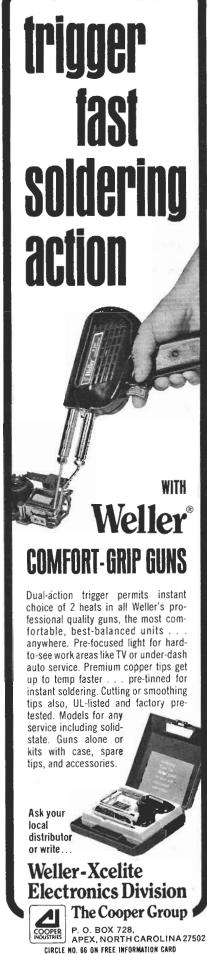
center frequency (f) and is equivalent to 1.1/R1C1. Both the input and the cco signals are fed into a pair of phase detectors. When the input frequency falls within the detection bandwidth of the circuit (0 to 14% of f), an output transistor capable of sinking up to 100 mA turns on. The output can directly control miniature lamps, relays, and LED's.

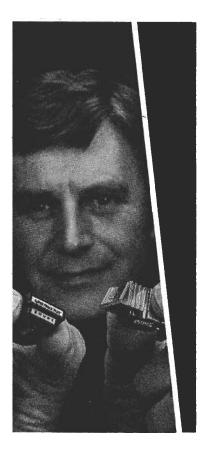
The 567 is an incredibly flexible chip with many different operating characteristics and capabilities. For example, it has a tone detection range of from 0.01 Hz to 500 kHz and will lock onto a signal with an amplitude of only 20 mV rms! Operating voltage ranges from a TTL-compatible 4.75 volts to a high of 9 volts. Standby current consumption is a reasonably low 6 to 10 mA, while activated current consumption without load is 11 to 15 mA.

**Experimenter's Circuit.** I think the best way for you to relate to the 567 PLL is to try it in an actual circuit, such as shown in Fig. 2. This circuit is a straightforward tone decoder with a built-in variable-frequency oscillator made from a single 555 timer.

The 555 is operated in its astable mode to produce square shaped output pulses. The repetition rate of the 555 is controlled by R4 and C5. Increasing the values of R4 and C5 slows the repetition rate, while decreasing







## Tune-up Time.

Your intricate stereo stylus assembly is like a high-performance sports car — it needs a little routine care to keep it in nep operating condition. Your Shure dealer will professionally inspect your stylus and clean it. Then, if your stylus tip is worn, he will select the correct genuine Shure replacement stylus to restore your cartridge to its original specifications. For more information, see your dealer or (next best), write for our booklet, "Visit to the Small World of a Stylus."

Shure Brothers Inc. 222 Hartrey Ave., Evanston, IL 60204 In Canada: A. C. Simmonds & Sons Limited

#### SHURE



Manufacturers of high fidelity components, microphones, sound systems and related circuitry.

CIRCLE NO. 58 ON FREE INFORMATION CARD

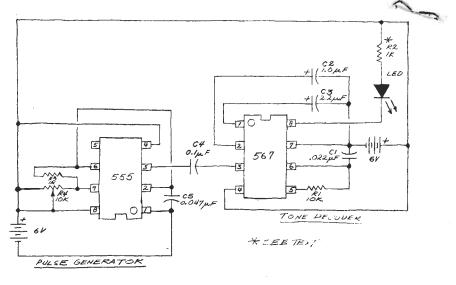


Fig. 2. Tone decoder demonstration circuit.

values speeds up the repetition rate.

With the component values given in Fig. 2,  $f_{\rm c}$  should be 5000 Hz (1.1/10,000  $\times$  0.022  $\times$  10<sup>-6</sup>. The test circuit, however, gave an fof 4480 Hz. I measured *R1* with a digital multimeter and found its actual resistance to be 10,320 ohms. This new value gave an fof 4845 Hz, which is within 3% of the predicted frequency, and any error is traceable to the tolerance of *C1*.

After you put the circuit together, experiment with the adjustment of R4 while watching the LED. The LED should turn rapidly on and off as you rotate R4's shaft past the point where the 555 oscillates at the f of the 567. If you dim the room lights, you may notice the LED flickering just before and after it turns full on and off. Resistor R2 limits LED current to about 3.5 mA. If your workbench is brightly illuminated or your LED inefficient, you can reduce the value of R2 (say, to 500 ohms) to get more current through the LED. In any event, be sure to use a regulated dc supply or fresh batteries.

Frequency Response. A graph

showing the response of the circuit in Fig. 2 in the acceptance bandwidth region is shown in Fig. 3. The graph shows the input frequency from the 555 versus the output current through the LED. Note that the bandwidth is fairly wide when the input frequency goes high to low and vice versa.

After you've tried changing the frequency of the input tone, replace R1 of the tone decoder with a 2000-ohm fixed resistor in series with a 20,000- or 25,000-ohm potentiometer and readjust R4 to provide an unknown input tone. You should easily be able to lock onto the unknown input frequency by adjusting the potentiometer. You can arrive at a rough estimate of the frequency of the unknown tone without using a scope or counter by measuring the total resistance of the potentiometer and its 2000-ohm series resistor and using the center-frequency formula.

At this point, the 567 would appear to work just fine, since it triggers an LED in response to any desired tone. But if you actually build the test circuit, you'll soon discover that the 567

#### **UNSCRAMBLERS!**

FOR PUBLIC SERVICE BAND SCANNERS
FAR EXCEEDS ANY ON THE MARKET

★ SATISFACTION GUARANTEED ★

ORDER FROM: THE ELECTRONICS

2404 SCOTSWOOD GARLAND, TEXAS 75041

UNSCRAMBLER — FOR ALL SCANNERS \$24°5
TUNES EASILY TO UNSCRAMBLE ALL SCRAMBLE
FREQUENCIES # FULL INSTRUCTIONS INCLUDED
# CONNECTS QUICKLY # 3½x3½x1½ INCHES #

POCKET UNSCRAMBLER — FOR POCKET \$34°5 SCANNERS © ONLY UNIT OF ITS KIND KNOWN © CONNECTS QUICKLY © 4x14x24 INCHES © INSTANT TUNING FULL INSTRUCTIONS POSTPAID — TEXAS RESIDENTS ADD 5% SALES TAX — NO C.O.D.'S PLEASE.

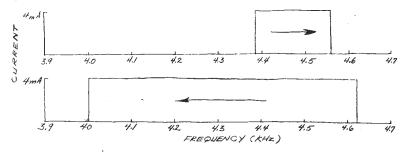


Fig. 3. Frequency response of circuit in Fig. 2. Arrows indicate positive and negative frequency changes.

has a tendency to trigger on what appear to be harmonics of the center frequency. The 567 will, in fact, lock onto frequencies corresponding to input signals near  $f_1$  (4n + 1) (where n = 0, 1, 2, 3...). Also, the square pulses from the 555 will cause an output for  $f_1/2$ 

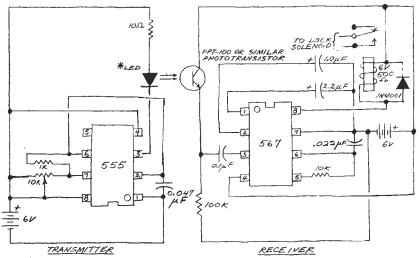
Fortunately, the response of the 567 to tones other than  $f_{\rm c}$  is usually of no consequence. And if false triggering to undesired tones becomes a problem, you can either change the center frequency of the 567 or attenuate the offending tone with a notch filter.

**Other Applications.** After you have experimented with the 567 using the basic circuit in Fig. 2, you will probably think of lots of interesting applications. One fascinating possibility is a secret photoelectric lock activated by a tone-modulated LED. A phototransistor connected to the input of the 567 can be used to receive the signal from the LED, as shown in Fig. 4. This circuit has an optical range of a few centimeters without external lenses at either the LED or phototransistor. This is all the distance you need for most lock applications, but for more range

you can add an amplifier between the phototransistor and the 567 and use a lens at each end.

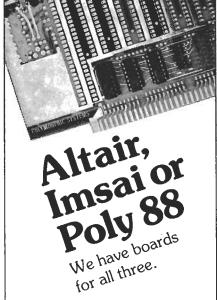
Whether checking out a potential application or just for fun, don't hesitate to experiment with the basic 567 circuit in Fig. 2. For best results, Signetics recommends that the resistance of R1 be between 200 and 20,000 ohms. Capacitor C2, the loop low-pass filter, should be selected from the Bandwidth versus Input Signal Amplitude graph given in both the National and Signetics data sheets on the 567.

At very low input frequencies, the time required for the 567 to lock onto the input tone can become relatively long. For example, I found that an input tone of 500 Hz required a full second for lock to occur. These and other eccentricities of the 567 are covered in detail in the manufacturer's (Signetics) data sheet, and I urge you to obtain this well-prepared document to assist you while experimenting with the 567. Another excellent source of information on the 567 is found in the Signetics Digital/Linear/MOS Applications book (Section 6, "Phase Locked Loop Applications").



\*USE HIGH EFFICIENCY INFRARED EMITTER SUCH AS SSL-55C, TIL 3J, ETC. 555 APPLIES 45 µS WIDE PULSES WITH PEAK CURRENT OF 120 mA TO LED

Fig. 4. Infrared-activated secret look.



Video Terminal Interface: This is one almost every owner of an Altair-compatible computer needs. There have been problems getting parts for this one (the people with the other Video board have had the same problem, since they use the same hard-to-get chip), but deliveries are improving. By the way, our video board has GRAPHICS as well as an 8 bit parallel key board input port, so compare before you buy. \$185.00 - \$210.00 kit depending on whether 32 or 64 characters per line.

Poly I/O: With this one you will save a lot of time in making prototype circuits. I/O port address is selectable with dip switch and fully buffered inputs and outputs. \$55 kit.

Analog Interface: Good for interfacing your computer to an analog world. Ten bits of resolution in and out. \$145.00 for one channel and \$195.00 for two channels (kit).

Ask about how to get a free POLY I/O or ANALOG Board.

Watch for our new products. Next time they will be on the shelf before we advertise.

Have you heard about the POLY-88? We don't have the space here to describe all the features. See it at your local computer store (except for MITS dealers, though some do carry our other products). This might be the system you're looking for.

Support your local computer store.

All prices and specifications subject to change without notice. Prices are U.S.A. only. Calif. residents add 6% sales tax. Add,5% shipping, handling, and insurance. Bank Americard and Mastercharge accepted.

#### PolyMorphic Systems

737 S. Kellogg, Goleta, Calif. 93017 (805) 967-2351

93

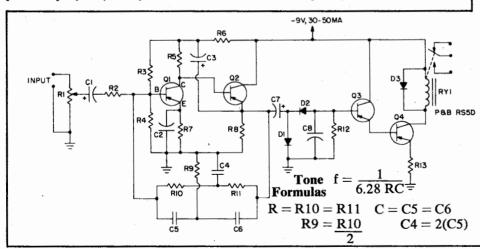
## Tone Controlled Relay



This tone controlled relay circuit is a lot more complex and more expensive than the usual tone control circuit but it's suggested for use when you need super-sensitivity and/or super-Q (the ability to respond only to the control frequency). Capacitor C8 provides a small delay of about 0.5 second so the unit can distinguish between the control input tone and random frequencies from sounds picked up by, say, a dynamic micro-

phone which can be connected to the input. In typical use potentiometer R1 is adjusted for the minimum input signal that provides reliable tripping of the relay.

The values shown provide an operating frequency of approximately 1500 Hz. The frequency is determined by R9, R10, R11, C4, C5 and C6. The relationship between these components is shown in the schematic. Frequency is calculated with



the formula  $F = \overline{2\pi RC}$ . Use 4,700ohm resistors for R10 and R11 even if it limits the range of frequencies you can use.

Virtually any general purpose PNP transistors of the type indicated in the parts list can be used. Diodes D1 and

#### PARTS LIST FOR TONE CONTROLLED RELAY

Resistors ½ watt, 10%, unless otherwise specified

R1-25,000- to 50,000-ohm linear potentiometer

R2, R10, R11-4,700-ohms

R3-100.000-ohms

R4-10.000-ohms

R5-5.600-ohms

R6-2.200-ohms

R7-1.200-ohms

R8-3.300-ohms

R9-2.400-ohms

R12-47.000-ohms

R13-22-ohms

Capacitors rated 10-VDC or higher

C1, C2, C4-0.047-uF

C3-2-uF

C5. C6-0.022-uF

C7-15-uF

C8-50-uF

Q1. Q2. Q3. Q4-2SB22. (Radio Shack 276-2003

or equiv.)

D1. D2-Germanium diode type 1N60 or equiv.

D3-Silicon rectifier type 1N4001 or equiv.

RY1-Relay, 200- to 400-ohm coil,

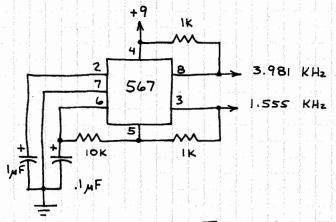
6-VDC, P&B RS5D-6 or equiv.

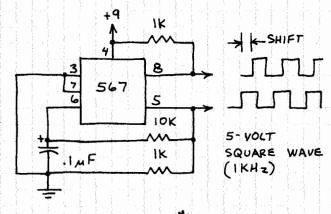
D2 should be the germanium 1N34/ 1N60 type. Be very careful about all power supply and capacitor polarities.

# TONE DECODER (CONTINUED) 567

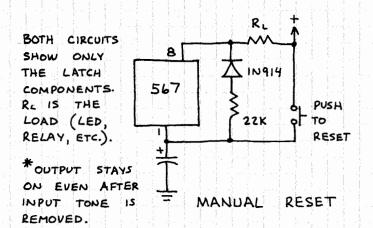
#### 2-FREQUENCY OSCILLATOR

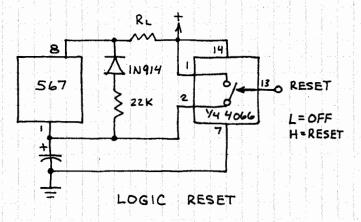
#### 2-PHASE OSCILLATOR



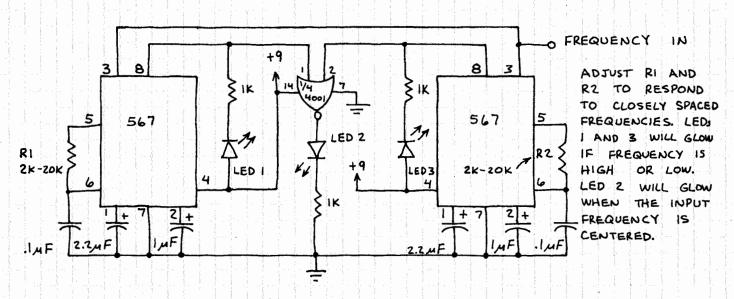


#### LATCHING THE 567 OUTPUT

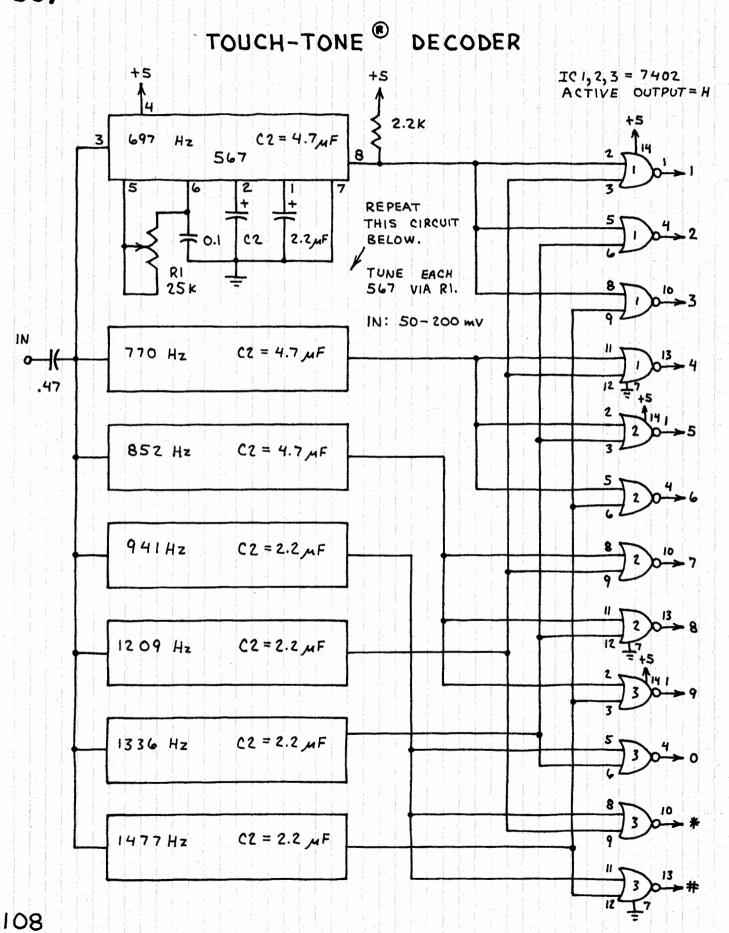




#### NARROW BAND FREQUENCY DETECTOR



# TONE DECODER (CONTINUED) 567



#### Missing-pulse detector handles variable frequencies

by Joe Lyle and Jerry Titsworth
Bendix Corp., Aircraft Brake and Strut Division, South Bend, Ind.

Virtually all missing-pulse detectors require an input signal of fixed frequency in order to operate satisfactorily. They malfunction when the input frequency varies because their circuits employ detection networks that have a fixed time constant. Through the implementation of inexpensive voltage-to-frequency and frequency-to-voltage converters to derive an average, or reference, frequency that tracks the input signal, this circuit can pinpoint missing pulses without being affected by input frequency variations.

As shown in the figure,  $A_1$  and  $A_2$  establish the reference frequency,  $f_{ref}$ , using input frequency  $f_{in}$ . Miss-

ing pulses do not change the reference because of the integrating capacitors within the converters. Meanwhile, the three NAND gates comprising the one-shot produce pulses of 10 microseconds in duration, with a frequency determined by the input signal.

The chip labeled  $A_4$  is clocked by  $f_{ref}$  and  $A_3$  through a NAND gate. As long as the input train is continuous, the Q output of  $A_4$  is low. If a missing pulse is detected, however, the one-shot will not generate a pulse to the reset pin of  $A_3$ , and the Q output of  $A_3$  (which is also clocked by  $f_{ref}$ ) will go high to clock  $A_4$ .  $A_4$  and  $f_{ref}$  will then switch  $A_5$ 's Q output to high.

This turns on transistor  $Q_1$  and the pilot lamp glows. Switch  $S_1$  is used to reset the circuit after a missing pulse has been detected. Note that circuit operation remains independent of the input frequency, since the arrival of  $f_{ref}$  and the  $10-\mu s$  pulse at  $A_3$  is synchronized to  $f_{in}$ .

The circuit should be calibrated by setting A<sub>1</sub> for an output voltage of 10 when a 10-kilohertz input signal is applied. Similarly, A<sub>2</sub> should be set to generate a 10-kHz signal for a 10-v input.

**Synchronous.** The circuit detects the missing pulse independently of the pulse train frequency. Voltage-to-frequency and frequency-to-voltage converters derive a reference frequency whose average remains the same for small anomalies occurring in the pulse train: converters' integrating capacitors hold f<sub>ref</sub> steady despite missing pulses. The reference in this way serves as a synchronous clock.

