

# Use transconductance amplifiers to make programmable active filters. The OTAs offer a 10-octave tuning range by using predictable $V_{BE}$ -versus- $I_C$ relationships.

Operational transconductance amplifiers (OTAs) can provide control of many active filter parameters over a 60-dB dynamic range. Ranges this wide are possible because the controlling signals are currents and not voltages. And currents can represent filter variables—such as corner frequency, center frequency,  $Q$  or bandwidth—better than a voltage, since the relationship between the collector current and the base-to-emitter voltage drop of a silicon transistor is predictable over extremely wide ranges.

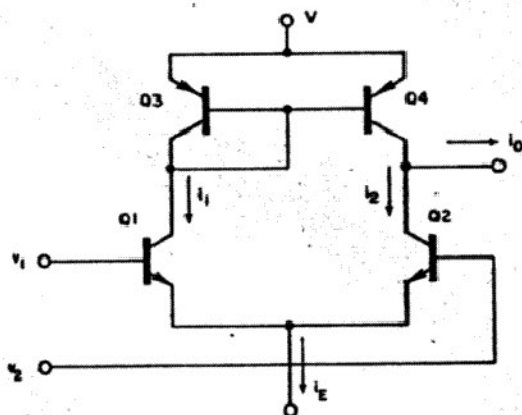
Over the last few years state-variable and biquad active filter circuits have been adopted by many companies as universal filter building blocks. Analog multipliers scale the impedance elements of these filters, thereby achieving parameter control.<sup>1</sup> Commercially available multipliers, however, are limited when it comes to voltage range, temperature drift and linearity. Because of these drawbacks, programmable filters based on analog voltage multipliers are generally restricted to operation over a 3-to-6-octave range. But by exploiting the transistor's  $V_{BE}$  vs  $I_C$  relationship in an OTA, you can use current signals to provide filter control over many decades.

The operational transconductance amplifier consists of a matched pair of npn transistors, connected as a variable transconductance multiplier, and a matched pair of pnp transistors, connected as a current mirror (Fig. 1). The small-signal relationship between the collector difference currents  $i_1 - i_2$  and the base-emitter difference voltages  $v_1 - v_2$  can be expressed by

$$i_1 - i_2 = (i_E / 2V_T) (v_1 - v_2),$$

in which  $V_T$  is the thermal voltage and is typically 26 mV at room temperature, and  $i_E$  is the combined emitter current.

Due to the current-mirror action of the pnp pair, the collector current of  $Q_4$  equals  $i_1$ , the collector current of  $Q_3$ . The difference current,  $i_1 - i_2$ —shown as  $i_o$ —is proportional to the product of the emitter current and the differential voltage applied to the npn pair. This product



1. The basic operational transconductance amplifier uses two pnp transistors and two npn's to provide a linear conversion of voltage to current.

holds up over a very wide range of common-emitter current values—a property that makes the OTA very powerful in programmable filter synthesis and wide-range gain control.<sup>2,3</sup> Let's see how this can be put to use by examining an integrator—the basic building block of active filters.

## Programmable integrators from OTAs

A programmable integrator can be built with an op amp and a transistor array (Fig. 2). To ensure the validity of a small-signal approximation for  $Q_1$  and  $Q_2$ , the differential input signal is scaled down with resistors  $R_1$  and  $R_2$ . For the values given in Fig. 2, the difference voltage becomes

$$v_1 - v_2 = (V^- - V^+) R_2 / (R_1 + R_2) \\ = 3.2 \times 10^{-3} (V^- - V^+).$$

Thus an input signal of  $\pm 5$  V, pk-pk, is attenuated into a  $\pm 16$ -mV swing at the bases of  $Q_1$  and  $Q_2$ . Transistor  $Q_3$ , connected in diode form, provides emitter bias for the current-mirror transistors, while the 100-k $\Omega$  potentiometer compensates for the over-all voltage offset of the OTA. The pot should be adjusted so that the difference current is zero for  $V^- = V^+ = 0$ .

The transfer function of the circuit can easily be found if you use Laplace transforms:

$$V_o = -I_o / sC = (i_E / 2V_T) (v_2 - v_1) / sC,$$

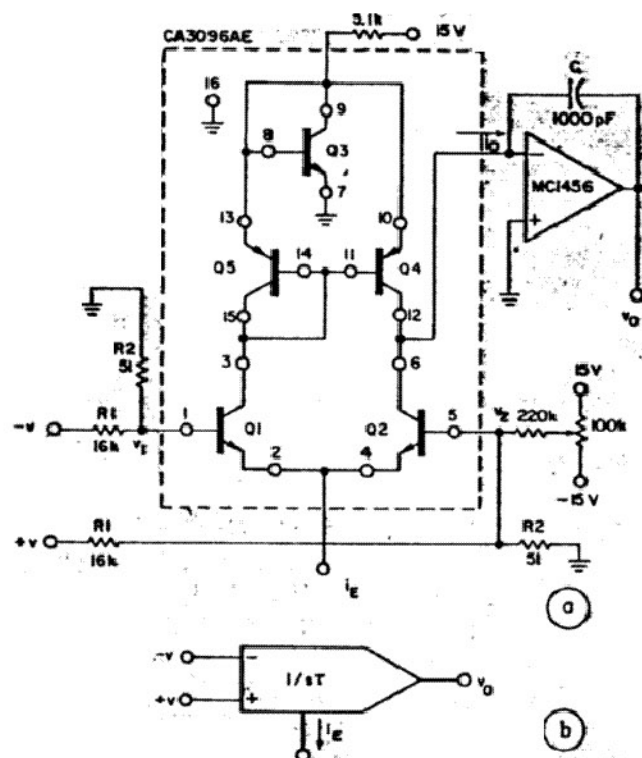
which can be simplified to

$$V_o = (V^+ - V^-) / s\tau,$$

where

$$\tau = [2 V_T C (R_1 + R_2) / R_2] / i_E.$$

These equations show that the circuit integrates the differential input signal and that the constant of integration,  $\tau$ , is inversely proportional to  $i_E$ . The constant,  $\tau$ , can be programmed externally if  $i_E$  is controlled with a variable current generator. When this is all put together, you have a programmable differential integrator.



2. To build a programmable integrator, the OTA can be combined with an op amp and some resistors to form the basic building block of an active filter.

Of course, if one input is grounded, the circuit is an inverting or noninverting integrator. To minimize any error introduced by the op amp at low current levels, an op amp with very low input bias current, such as one with a FET input or super-beta transistor input, should be used.

### Combined integrators make a filter

Two integrators can be combined to make a programmable filter that exhibits simultaneous second-order, low-pass and bandpass responses (Fig. 3). To build a circuit using this block diagram, you also need a dual-output current generator to control both integrators. The actual circuit is shown in Fig. 4 (offset-adjustment pots have not been included, for the sake of simplicity). Now let's see how it works.

If we start with the block diagram of Fig. 3 and use Laplace transforms to find the bandpass and low-pass outputs, we get:

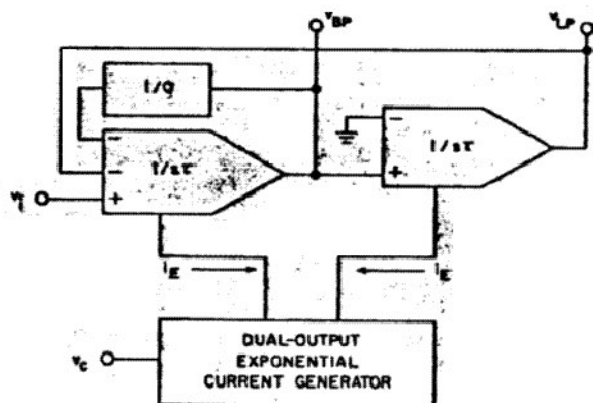
$$V_{BP} = (V_i - V_{LP} - V_{BP}/Q) / s\tau \text{ and}$$

$$V_{LP} = V_{BP} / s\tau,$$

where  $1/Q$  represents the fraction of  $V_{BP}$  fed back to the input of the first integrator. After combining these equations and rearranging terms, we get:

$$V_{BP}/V_i = s\omega_c / (s^2 + s\omega_c/Q + \omega_c^2) \text{ and}$$

$$V_{LP}/V_i = \omega_c^2 / (s^2 + s\omega_c/Q + \omega_c^2),$$



3. This block diagram of a second-order integrator shows how you can obtain both bandpass and low-pass outputs from a single programmable filter.

where

$$\omega_c = 1/\tau \approx 62 \times 10^6 (i_E).$$

The center frequency of the bandpass can now be calculated:

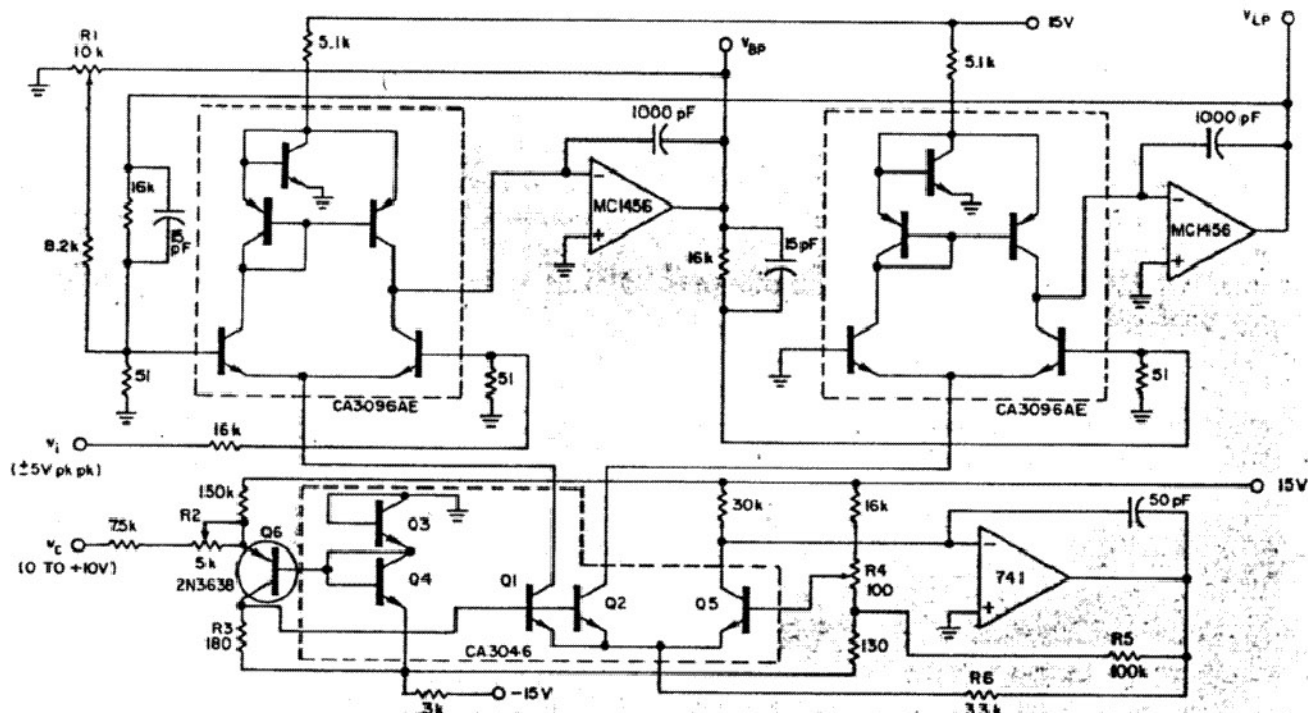
$$f_c = \omega_c / 2\pi \approx 10^7 (i_E).$$

This shows that  $f_c$  is directly proportional to the common-emitter current that feeds the OTAs. Thus if  $i_E$  changes from  $1 \mu A$  to  $1 mA$ ,  $f_c$  varies accordingly from  $10 Hz$  to  $10 kHz$ —a three-decade or 10-octave range.

When you have ranges this wide, exponential parameter control is usually more convenient than linear control, since changes can then be expressed in decibels rather than linear steps. The exponential control over a 10-Hz-to-10-kHz range can be accomplished by use of a dual-output exponential current generator. Let's specify the circuit parameters so that a 1-V increment in the control voltage,  $V_c$ , results in a one-octave change in  $i_E$  and hence in  $f_c$ . The relationship between  $f_c$  and  $V_c$  can then be written as

$$f_c = 10 \times 2^{V_c} \text{ Hz } (0 \leq V_c \leq 10).$$

The circuit of Fig. 4 provides this wide range. Its  $Q$  factor is set by  $R_1$  and can range from 0.5 to several hundred. As the  $Q$  increases, the circuit will eventually break into oscillation at a frequency of  $f_c$ . When used in this mode, the circuit can operate as a voltage-controlled quadrature oscillator, since the low-pass output has the



4. The complete schematic of the programmable active filter uses but three transistor arrays and three op amps,

yet offers a 10-octave tuning range. The filter delivers bandpass and low-pass outputs.

bandpass output by  $90^\circ$ . For good sine purity, though, a nonlinear limiting element should be added in the feedback loop to ensure output amplitude stabilization. As an added bonus, the  $Q$  and  $f_c$  are totally independent, and any adjustment of either has no effect on the other.

#### The circuit works like this . . .

A common problem with state-variable circuits is the so-called  $Q$  enhancement at high frequencies, which results from the phase lag of finite bandwidth op amps. This effect can be reduced if you use high-speed op amps and counteract the phase lag of the amps with some intentionally inserted phase lead in the interstage coupling circuitry. This is exactly what the 15-pF capacitors do when connected in parallel with 16-k $\Omega$  coupling resistors.

Since the amount of phase lag is likely to vary from one op-amp type to another, the capacitor values shown in Fig. 4 are only indicative of the order of magnitude involved. The optimum capacitor values can be determined experimentally.

The dual-output current generator used in Fig. 4 consists of three matched transistors,  $Q_1$ ,  $Q_2$  and  $Q_3$ , a level shifter,  $Q_4$ , and a regulator op amp. The op amp keeps the current through  $Q_3$  constant, and thus develops a temperature-tracking reference voltage at the emitter of  $Q_4$  for the emitters of  $Q_1$  and  $Q_2$ .

Level shifter  $Q_4$  scales the input voltage range into a nominal swing of 180 mV at the common

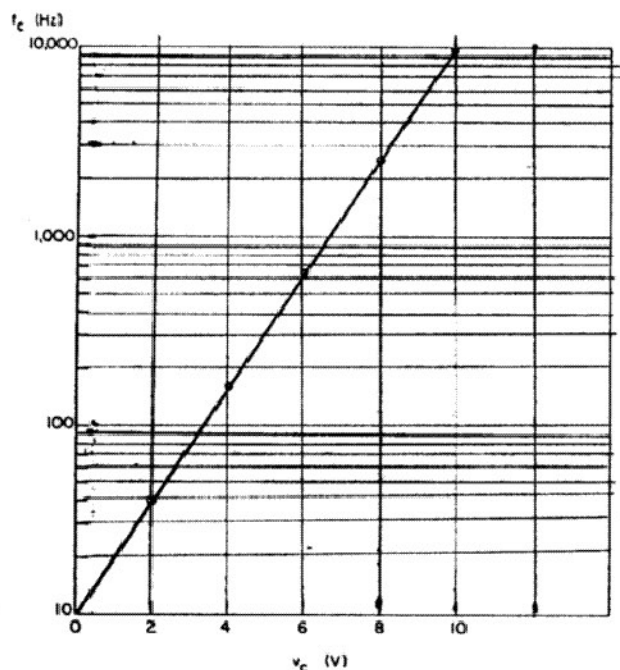
bases of  $Q_1$  and  $Q_2$ . This permits the collector currents of these transistors to undergo a three-decade change as  $V_C$  is swept from 0 to +10 V. Diode-connected transistors  $Q_3$  and  $Q_4$  translate the reference voltage of the entire exponential circuit to below ground, so that the collectors of  $Q_1$  and  $Q_2$  can feed directly to the common emitters of the OTAs.

Potentiometer  $R_2$  adjusts the width of the exponential range, and  $R_1$  shifts the entire range up or down and, in turn, controls the scale factor of the exponential function. The 50-pF capacitor in the feedback circuit of the regulator, together with  $R_6$ , prevents high-frequency oscillation of the op amp. Resistor  $R_5$  compensates for any error introduced by the emitter bulk resistances,  $r_E$ , of  $Q_1$  and  $Q_2$ . These resistances could cause the relationship between the collector current and the base-emitter voltage drop to be no longer exponential in the upper current range. The value of  $R_5$  is calculated by the formula:

$$R_5 \approx 2 \times 130 \Omega \times R_6 / r_E.$$

Since  $r_E$  is only about 10  $\Omega$  or so, the value of  $R_5$  easily compensates for the bulk-resistance error.

Nonperfect matching between  $Q_1$  and  $Q_2$  means that the collector currents of these transistors are likely to differ, and this will affect both  $f_c$  and  $Q$ . However, this doesn't create a problem, since the mismatch is automatically compensated for during the initial calibration adjustments of  $R_1$ ,  $R_2$  and  $R_1$ . The performance of the filter for a  $Q$  setting of about 50 is shown in Fig. 5.



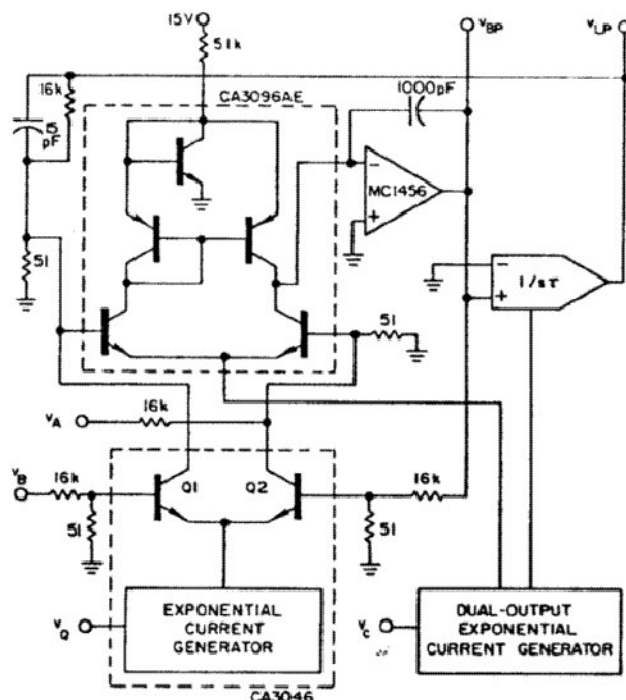
5. The response of the programmable filter is linear (in decibels) over a 10-V tuning range. It has a corner or center frequency within a 10-Hz-to-10-kHz range.

The transfer characteristics of the OTAs, as well as those of the dual-output current generator, depend on  $V_T$ , the thermal voltage. The filter thus has a certain amount of temperature sensitivity, and for applications where this is critical, you can minimize it. All you have to do is replace the  $51\text{-}\Omega$  input resistors ( $R_2$  in Fig. 2) and the  $180\text{-}\Omega$  current-generator input resistor ( $R_3$  in Fig. 4) with thermistor elements that have compensating characteristics.

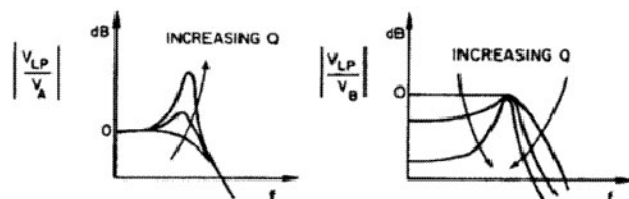
#### Automatic Q control for the filter

In the circuit of Fig. 4,  $Q$  is determined by the setting of  $R_1$ . However, the  $Q$  can also be made programmable if you add a transconductance multiplier and associated current generator. This modification is shown in Fig. 6, where, for clarity, only the circuit components directly involved in the change are shown schematically; the rest of the circuit is shown in block diagram form. For optimum performance, make provisions to zero the voltage offset between  $Q_1$  and  $Q_2$ .

The response of this circuit to input-signal levels is the same as for the circuit shown in Fig. 4, except that the value of  $Q$  is now determined by a control voltage,  $V_Q$ . The gain of the circuit at  $f_c$  equals  $Q$ , and this may be a problem for large values of  $Q$ . This is because the amplifiers in the filter will saturate for large voltage inputs, unless the input is intentionally maintained below a predetermined value. To eliminate this possible saturation problem, an additional input to the



6. You can program the  $Q$  of the tunable filter by adding a transconductance multiplier in the input stage and an exponential current generator to control it.



7. The low-pass response for the modified circuit of Fig. 6 shows that for the  $v_A$  input, gains will reach values higher than one (a), while for the  $v_B$  input, gains are adjusted for a maximum of one (b).

filter,  $V_n$ , can be designed in.

Since the signal  $V_B$  is applied to the filter through the multiplier that controls the  $Q$ , it undergoes some attenuation. This attenuation is inversely proportional to the value of  $Q$ . Thus the gain of the filter at  $f_c$  is unity, regardless of the value of  $Q$ . The typical low-pass responses to the two input lines are shown in Fig. 7. ■■

#### References

1. Sparkes, R. G. and Sedra, A. S., "Programmable Active Filters," *IEEE Journal of Solid-State Circuits*, Vol. SC-8, February, 1973, pp. 93-95.
2. Jung, W. G., "Get Gain Control of 80 to 100 dB," *Electronic Design* No. 13, June 21, 1974, pp. 94-99.
3. Franco, "Hardware Design of a Real-Time Musical System," Department of Computer Science Technical Report. University of Illinois. October. 1974.