Simplify your F/V converter designs with versatile V/F ICs. Starting with a basic converter circuit, you can modify it to meet almost any application requirement. You can spare vourself some hard labor when designing frequency-to-voltage (F/V) converters by using a voltage-to-frequency IC in your designs. These ICs form the basis of a series of accurate, yet economical, F/V converters suiting a variety of applications.

Figure 1 shows an LM331 IC (or LM131 for the military temperature range) in a basic F/V converter configuration (sometimes termed a stand-alone converter because it requires no op amps or other active devices other than the IC). (Comparable V/F ICs, such as RM4151, can take advantage of this and other circuits described in this article, although they might not always be pin-for-pin compatible).

This circuit accepts a pulse-train or square wave input amplitude of 3V or greater. The 470 pF coupling capacitor suits negative-going input pulses between 80  $\mu$ s and 1.5  $\mu$ s, as well as accommodating square waves or positive-going pulses (so long as the interval between pulses is at least 10  $\mu$ s).

# **IC Handles the Hard Part**

The LM331 detects an input-signal change by sensing when pin 6 goes negative relative to the threshold voltage at pin 7, which is nominally biased 2V lower than the supply voltage. When a signal change occurs, the LM331's input comparator sets an internal latch and initiates a timing cycle. During this cycle, a current equal to VRFF/Rs flows out of pin 1 for National Semiconductor Appendix C Robert A. Pease



a time t = 1.1 R<sub>t</sub>C. The 1  $\mu$ F capacitor filters this pulsating current from pin 1, and the current's average value flows through load resistor  $R_i$ . As a result, for a 10 kHz input, the circuit outputs 10  $V_{DC}$  across  $R_i$  with good (0.06% typical) nonlinearity.

Two problems remain, however: the output at V1 includes about 13 mVp-p ripple, and it also lags 0.1 second behind an input frequency step change, settling to 0.1% of fullscale in about 0.6 second. This ripple and slow response represent an inherent tradeoff that applies to almost every F/V converter.

## The Art of Compromise

Increasing the filter capacitor's value reduces ripple but also increases response time. Conversely, lowering the filter capacitor's value improves response time at the expense of larger ripple. In some cases, adding an active filter results in faster response and less ripple for high input frequencies.

Although the circuit specifies a 15V power supply, you can use any regulated supply between 4 V<sub>DC</sub> and 40 V<sub>DC</sub>. The output voltage can extend to within 3  $V_{DC}$  of the supply voltage, so choose R<sub>1</sub> to maintain that output range.

Adding a 220 k $\Omega$ /0.1  $\mu$ F postfilter to the circuit slows the response slightly, but it also reduces ripple to less than 1 mVp-p for frequencies from 200 Hz to 10 kHz. The reduction in ripple achieved by adding this passive filter, while not as good as that obtainable using an active filter, could suffice in some applications.



FIGURE 1. A Simple Stand-Alone F/V Converter Forms the **Basis for Many Other Converter-Circuit Configurations** 

# **Improving the Basic Circuit**

Further modifications and additions to the basic F/V converter shown in *Figure 1* can adapt it to specific performance requirements. Figure 2 shows one such modification. which improves the converter's nonlinearity to 0.006% typical

Reconsideration of the basic stand-alone converter shows why its nonlinearity falls short of this improved version's. At low input frequencies, the current source feeding pin 1 in the LM331 is turned off most of the time. As the input frequency increases, however, the current source stays on more of the time, and its own impedance attenuates the output signal for an increasing fraction of each cycle time. This disproportionate attenuation at higher frequencies causes a parabolic change in full-scale gain rather than the desired linear one.

In the improved circuit, on the other hand, the PNP transistor acts as a cascade, so the output impedance at pin 1 sees a constant voltage that won't modulate the gain. Also, with an alpha ranging between 0.998 and 0.990, the transistor exhibits a temperature coefficient of between 10 ppm/°C and 40 ppm/°C—a fairly minor effect. Thus, this circuit's

nonlinearity does not exceed 0.01% maximum for the 10V output range shown and is normally not worse than 0.01% for any supply voltage between 4V and 40V.

# **Add an Output Buffer**

The circuit in *Figure 3* adds an output buffer (unity-gain follower) to the basic single-supply F/V converter. Either an LM324 or LM358 op amp functions well in a single-supply circuit because these devices' common-mode ranges extend down to ground. But if a negative supply is available. you can use any op amp: types such as the LF351B or LM308A, which have low input currents, provide the best accuracy.

The output buffer in Figure 3 also acts as an active filter, furnishing a 2-pole response from a single op amp. This filter provides the general response

$$
V_{OUT}/I_{OUT} = R_L/(1 + K1p + K2p^2)
$$
.

(p is the differential operator  $d/dt$ .) As shown.  $R_1$  controls the filter's DC gain. The high frequency response rolls off at 12 dB/octave. Near the circuit's natural resonant frequency. you can choose the damping to give a little overshoot-or none, as desired.



#### FIGURE 2. Adding a Cascade Transistor to the LM331's Output Improves Nonlinearity to 0.006%



# Dealing with F/V Converter Ripple

Voltage ripple on the output of F/V converters can present a problem, and the chart shown in Figure A indicates exactly how big a problem it is. A simple, slow. RC filter exhibits low ripple at all frequencies. Two-pole filters offer the lowest ripple at high frequencies and provide a 30-times-faster step response than RC devices.

To reduce a circuit's ripple at moderate frequencies, however, you can cascade a second active-filter stage on the F/V converter's output. That circuit's response also appears in Figure A and shows a significant improvement in low-ripple bandwidth over the single-active-filter configuration, with only a 30% degradation of step response.

Figures  $B$  and  $C$  show filter circuits suitable for cascading. The inverting filter in Figure B requires closely matched resistors with a low TC over their temperature range for best accuracy. For lowest DC error, choose R5 = R2 +  $(R_{\rm IN}|R_{\rm F})$ . This circuit's response is

 $-V_{OUT}/V_{IN} = n/(1 + (R_F + R2 + nR2)C4p + R_F R2C3C4p^2).$ 

where 
$$
n = DC
$$
 gain. If  $R_{IN} = R_F$  and  $n = 1$ ,

 $-V_{OUT}/V_{IN} = 1/(1+(R_F+2R2)C4p+R_FR2C3C4p^2).$ 



TL/H/8741-4





TL/H/8741-5





TL/H/8741-6

# FIGURE C. This 2-Pole Noninverting Filter Suits **Cascade Requirements on F/V Converter Outputs**

The circuit shown in *Figure C* does not require precision passive components, but for best accuracy, choosing an A1 with a high CMRR is critical. An LM308A op amp's 96 dB minimum CMRR suits this circuit well, but an LM358B's 85 dB typical figure also proves adequate for many applications. Circuit response is

 $V_{\text{OUT}}/V_{\text{IN}} = 1/(1 + (R1 + R2) C2p + R1R2C1C2p^2).$ For best results, choose  $R3 = R1 + R2$ .

# **Components Determine Response**

The specific response of the circuit in Figure 3 is

 $V_{\text{OUT}}/I_{\text{OUT}} = R_1/(1 + (R_1 + R_2)C_2p +$ R<sub>1</sub> R2C1C2C2p<sup>2</sup>).

Making C2 relatively large eliminates overshoot and sine peaking. Alternatively, making C2 a suitable fraction of C1 (as is done in Figure 3) produces both a sine response with 0 dB to 1 dB of peaking and a quick real-time response having only 10% to 30% overshoot for a step response. By maintaining Figure 3's ratio of C1:C2 and R2:RL, you can adapt its 2-pole filter to a wide frequency range without tedious computations.

This filter settles to within 1% of a 5V step's final value in about 20 ms. By contrast, the circuit with the simple RC filter shown in *Figure 1* takes about 900 ms to achieve the same response, vet offers no less ripple than Figure 3's op amp approach.

As for the other component in the 2-pole filter, any capacitance between 100 pF and 0.05 µF suits C3 because it serves only as a bypass for the 200 k $\Omega$  resistor. C4 helps reduce output ripple in single positive power-supply systems when  $V_{\text{OUT}}$  approaches so close to ground that the op amp's output impedance suffers. In this circuit, using a tantalum capacitor of between 0.1  $\mu$ F and 2.2  $\mu$ F for C4 usually helps keep the filter's output much quieter without degrading the op amp's stability.

# **Avoid Low-Leakage Limitations**

Note that in most ordinary applications. this 2-pole filter performs as well with 0.1  $\mu$ F and 0.02  $\mu$ F capacitors as the passive filter in *Figure 1* does with 1 uF. Thus, if you require a 100 Hz F/V converter, the circuit in Figure 3 furnishes good filtering with C1 = 10  $\mu$ F and C2 = 2  $\mu$ F, and eliminates the 100 uF low-leakage capacitor needed in a passive filter

Note also that because C1 alwavs has zero DC voltage across it, you can use a tantalum or aluminum electrolytic capacitor for C1 with no leakage-related problems: C2. however, must be a low-leakage type. At room temperature, typical 1 uF tantalum components allow only a few nanoamperes of leakage, but leakage this low usually cannot be quaranteed

## **Compensating for Temperature Coefficients**

F/V converters often encounter temperature-related problems usually resulting from the temperature coefficients of passive components. Following some simple design and manufacturing quidelines can help immunize your circuits against loss of accuracy when the temperature changes.

Capacitors fabricated from Teflon or polystyrene usually exhibit a TC of  $-110 \pm 30$  ppm/°C. When you use such a component for the timing capacitor in an F/V converter (such as C<sub>t</sub> in the *figure*) the circuit's output voltage—or the gain in terms of volts per kilohertz-also exhibits  $a - 110$ ppm/°C TC.

But the resistor-diode network (Rx, D1, D2) connected from pin 2 to ground in the figure can cancel the effect of the timing capacitor's large TC. When  $R_X = 240 \text{ k}\Omega$ , the current flowing through pin 1 will then have an overall TC of 110 ppm/°C, effectively canceling a polystyrene timing capacitor's TC to a first approximation. Thus, you needn't find a zero-TC capacitor for  $C_t$ , so long as its temperature coefficient is stable and well established. As an additional advantage, the resistor-diode network nearly compensates to zero the TC of the rest of the circuit.

## **Bake it for a While**

After the circuit has been built and checked out at room temperature, a brief oven test will indicate the sign and the size of the TC for the complete F/V converter. Then you can add resistance in series with  $R<sub>X</sub>$ , or add conductance in parallel with it, to greatly diminish the TC previously observed and vield a complete circuit with a lower TC than you could obtain simply by buying low TC parts.

For example, if the circuit increases its full-scale output by 0.1% per 30°C (33 ppm/°C) during the oven test, adding 120 k $\Omega$  in series with R<sub>Y</sub> = 240 k $\Omega$  cancels the temperature-caused deviation. Or, if the full-scale output decreases bv -0.04% per 20°C (-20 ppm/°C), just add 1.2 M $\Omega$  in parallel with R<sub>v</sub>.

Note that to allow trimming in both directions, you must start with a finite *fixed* TC (such as the  $-110$  ppm/°C of C<sub>t</sub>), which then nominally cancels out by the addition of a finite adjustable TC. Only by using this procedure can you compensate for whatever polarity of TC is found by the oven test

You can utilize this technique to obtain TCs as low as 20 ppm/°C, or perhaps even 10 ppm/°C, if you take a few passes to zero-in on the best value for Ry. For optimum results, consider the following guidelines:

- Use a good capacitor for C<sub>t</sub>; the cheapest polystyrene capacitors can shift value by 0.05% or more per temperature cycle. In that case, you would not be able to distinquish the actual temperature sensitivity from the hysteresis, and you would also never achieve a stable circuit.
- After soldering, bake or temperature-cycle the circuit (at a temperature not exceeding 75°C in the case of polystyrene) for a few hours to stabilize all components and to relieve the strains of soldering.
- Do not rush the trimming. Recheck the room temperature value before and after you take the high temperature data to ensure a reasonably low hysteresis per cycle
- Do not expect a perfect TC at  $-25^{\circ}$ C if you trim for  $\pm 5$ ppm/°C at temperatures from  $+25^{\circ}$ C to 60°C. None of the components in the *figure's* circuit offer linearity much better than 5 ppm/°C or 10 ppm/°C cold, if trimmed for a zero TC at warm temperatures. Even so, using these techniques you can obtain a data converter with better than 0.02% accuracy and 0.003% linearity, for a  $\pm$  20°C range around room temperature.
- Start out the trimming with  $R_X$  installed and its value near the design-center value (e.g., 240 k $\Omega$  or 270 k $\Omega$ ), so you



Two Diodes and a Resistor Help Decrease an F/V Converter's Temperature Coefficient

will be reasonably close to zero TC; you will usually find the process slower if you start without any resistor, because the trimming converges more slowly.

- . If you change R<sub>X</sub> from 240 k $\Omega$  to 220 k $\Omega$ , do not pull out the 240 k $\Omega$  part and put in a new 220 k $\Omega$  resistor--you will get much more consistent results by adding a 2.4  $M\Omega$ resistor in parallel. The same admonition holds true for adding resistance in series with  $R_X$ .
- . Use reasonably stable components. If you use an LM331A (±50 ppm/°C maximum) and RN55D film resistors (each  $\pm$  100 ppm/°C) for R<sub>1</sub>, R<sub>t</sub> and R<sub>S</sub>, you probably won't be able to trim out the resulting  $\pm 350$  ppm/°C worst-case TC. Resistors with a TC specification of 25 ppm/°C usually work well. Finally, use the same resistor value (e.g., 12.1  $k\Omega \pm 1\%$  for both R<sub>S</sub> and R<sub>t</sub>; when these resistors come from the same manufacturer's batch, their TC tracking will usually rate at better than 20 ppm/°C.

Whenever an op amp is used as a buffer (as in Figure 3), its offset voltage and current ( $\pm 7.5$  mV maximum and  $\pm 100$ nA, respectively, for most inexpensive devices) can cause a  $\pm$  17.5 mV worst-case output offset. If both plus and minus supplies are available, however, you can easily provide a symmetrical offset adjustment. With only one supply, you can add a small positive current to each op amp input and also trim one of the inputs.

## **Need a Negative Output?**

If your F/V converter application requires a negative output voltage, the circuit shown in Figure 4 provides a solution with excellent linearity ( $\pm 0.003$ % typical,  $\pm 0.01$ % maximum). And because pin 1 of the LM331 always remains at 0 V<sub>DC</sub>, this circuit needs no cascade transistor. (Note, however, that while the circuit's nonlinearity error is negligible, its ripple is not.)

The circuit in *Figure 4* offers a significant advantage over some other designs because the offset adjust voltage derives from the stable 1.9 V<sub>DC</sub> reference voltage at pin 2 of the LM331; thus any supply voltage shifts cause no output shifts. The offset pot can have any value between 200 k $\Omega$ and 2  $M\Omega$ .

An optional bypass capacitor (C2) connected from the op amp's positive input to ground prevents output noise arising from stray noise pickup at that point; the capacitance value is not critical.

#### **A Familiar Response**

The circuit in *Figure 4* exhibits the same 2-pole responsewith heavy output ripple attenuation-as the noninverting filter in Figure 3. Specifically,

 $V_{\text{OUT}}/I_{\text{OUT}} = R_F/(1 + (R4 + R_F)C4p + R4R_FC3C4p^2).$ Here also, R5 = R4 + R<sub>F</sub> = 200 k $\Omega$  provides the best bias current compensation.

The LM331 can handle frequencies up to 100 kHz by utilizing smaller-value capacitors as shown in *Figure 5*. This circuit increases the current at pin 2 to facilitate high-speed switching, but, despite these speed-ups, the LM331's 500 ppm/°C TC at 100 kHz causes problems because of switching speed shifts resulting from temperature changes.

To compensate for the device's positive TC, the LM334 temperature sensor feeds pin 2 a current that decreases linearly with temperature and provides a low overall temperature coefficient. An R<sub>v</sub> value of 30 k $\Omega$  provides first-order compensation, but you can trim it higher or lower if you need more precise TC correction.



low temperature coefficients

TL/H/8741-8

FIGURE 4. In This F/V Circuit, the Output-Buffer Op Amp Derives its Offset Voltage from the Precision Voltage Source at Pin 2 of the LM331

# **Detect Frequencies Accurately**

Using an F/V converter combined with a comparator as a frequency detector is an obvious application for these devices. But when the F/V converter is utilized in this way, its output ripple hampers accurate frequency detection, and the slow filter frequency response causes delays.

If a quick response is not important, though, you can effectively utilize an LM331-based F/V converter to feed one or more comparators, as shown in *Figure 6*. For an input frequency drop from 1.1 kHz to 0.5 kHz, the converter's output

responds within about 20 ms. When the input falls from 9 kHz to 0.9 kHz, however, the output responds only after a 600 ms lag, so utilize this circuit only in applications that can tolerate F/V circuits' inherent delays and ripples.

# **Author's Biography**

Bob Pease is a staff scientist in the Advanced Linear Integrated Circuit Group at National Semiconductor Corp., Santa Clara, CA. Holder of four patents, he earned a BSEE from MIT. Bob lists tracking abandoned railroad roadbeds and designing V/F converters as hobbies.



FIGURE 5. An LM334 Temperature Sensor Compensates for the F/V Circult's Temperature Coefficient



FIGURE 6. Combining a V/F IC with Two Comparators Produces a Slow-Response Frequency Detector

National Semiconductor Appendix D Robert A. Pease



loperatile Monolithic V/Fs Can Compute as Well as Convert with High Accuracy

The best of the monolithic voltage-to-frequency (V/F) converters have performance that's so good it equals or exceeds that of modular types. Some of these ICs can be designed into quite a variety of circuits because they're notably versatile. Along with versatility and high performance come the advantages that are characteristic of all V/F converters, including good linearity, excellent resolution, wide dynamic range, and an output signal that's easy to transmit as well as couple through an isolator.

One of the recently introduced monolithic types, the LM131, has both high performance and a design that's rather flexible. For instance, it can compute and convert at the same time; the computation is a part of the conversion. Among other functions, it can provide the product, ratio and square root of analog inputs.

This IC has an internal reference for its conversion circuitry that's also brought out to a pin, so it's available to external circuits associated with the converter. Not surprisingly, it turns out that any deviations of the reference, due to process variations and temperature changes have equal and opposite effects on the scale factors of the converter and the external circuitry. (This presumes, of course, that the scale factor of the external circuitry is a linear function of voltage.)

## PRECISION RELAXATION OSCILLATOR

Before looking at some applications, quickly take a look at the basic circuit of an LM131 V/F converter (Figure 1). Basically, this IC, like any V/F converter, is a precision relaxation oscillator that generates a frequency linearly proportional to

the input voltage. As might be expected, the circuit has a capacitor, CL, with a sawtooth voltage on it. Generally speaking, the circuit is a feedback loop that keeps this capacitor charged to a voltage very slightly higher than the input voltage, V<sub>IN</sub>. If V<sub>IN</sub> is high, C<sub>L</sub> discharges relatively quickly through RL, and the circuit generates a high frequency. If V<sub>IN</sub> is low, C<sub>L</sub> discharges slowly, and the converter puts out a low frequency.

When C<sub>I</sub> discharges to a voltage equal to the input, the comparator triggers the one-shot. The one-shot closes the current switch and also turns on the output transistor. With the switch closed, current from the current source recharges C<sub>i</sub> to a voltage somewhat higher than the input. Charging continues for a period determined by  $R<sub>T</sub>$  and  $C<sub>T</sub>$ . At the end of this period, the one-shot returns to its quiescent state and CL resumes discharging.

Resistor R<sub>S</sub> sets the amount of current put out by the current source. In fact, the current in pin 1, with the switch on. is identical to the current in pin 2. The latter pin is at a constant voltage (nominally 1.90V), so a given resistor value can set the operating currents. When connected to a high impedance buffer, this pin provides a stable reference for external circuits.

The open-collector output at pin 3 permits the output swing to be different from the converter's supply voltage, if the load circuit requires. The supplies don't have to be separate, however, and both the converter and its load can use the same voltage.



FIGURE 1. A voltage-to-frequency converter such as this is a relaxation oscillator with a frequency proportional to the input voltage. Current pulses keep C<sub>L</sub>'s average voltage slightly greater than the input voltage.

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# **STEADY AS SHE GOES**

By far the simplest of the circuits that make use of the reference output voltage from the LM131 is one that simply ties this output pin right back to the signal input. This connection is iust a V/F converter with a constant input, which makes it a constant-frequency oscillator. Even with this simple circuit (Figure 2), variations in the reference voltage have two opposite effects that cancel each other out, so the circuit is particularly stable. In this type of circuit, the temperature-dependent internal delays tend to cancel as well, which isn't true of relaxation oscillators based on op amps or comparators.



FIGURE 2. A V/F converter is a stable-frequency oscillator if its input is connected to its reference output. If the reference voltage changes, the effects of the change cancel out, so the frequency doesn't change. With low tempco components for Rand C<sub>T</sub>, frequency stability vs temperature can be as good as  $\pm 25$  ppm/°C.

Resistors  $R_L$  and  $R_S$  are best taken from the same batch. (R<sub>L</sub> must be larger than R<sub>S</sub>, so it's made up of two resistors.) By doing this, the tempco tracking, which is the critical parameter, is five to ten times better than it would be if Ri were a single 30.1 k $\Omega$  resistor.

Although the reference output, pin 2, can't be loaded without affecting the converter's sensitivity, the comparator input, pin 7, has a high impedance so this connection does no harm

Frequency stability is typically  $\pm 25$  ppm/°C, even with an LM331, which as a V/F converter is specified only to 150 ppm/°C maximum. From 20 Hz to 20 kHz, stability is excellent, and the circuit can generate frequencies up to 120 kHz.

Although the simplest way of using the reference output is to tie it back to the input, the reference can also be buffered and amplified to supply such external circuitry as a resistive transducer, which might be a strain gauge or a pot (Figure 3). As in the stable oscillator already described, deviations of the internal reference voltage from the ideal cause the transducer's and the converter's sensitivities to change equally in opposite directions, so the effects cancel.

In this circuit, op amp A2 buffers and amplifies the constant voltage at pin 2 of the converter to provide the 5V excitation for the strain gauge. Amplifier A1, connected as an instrumentation amplifier, raises the output of the strain gauge to a usable level while rejecting common-mode pickup.

A potentiometer-type transducer works just as well with this circuit. Its wiper output takes the place of A1's output as shown at the X.

The reference terminal is both a constant voltage output and a current programming input. So far, it's been shown simply with one or two resistors going to ground. It is, however, a full-fledged signal input that accepts a signal from a current source quite well.



FIGURE 3. In this strain-to-frequency converter, the converter's reference excites the strain gauge (or the optional pot) through buffer amp A2. This makes the circuit insensitive to changes in the reference voltage.

Appendix D: Versatile Monolithic V/Fs Can Compute as Well as Convert with High Accuracy

This extra input is what enables the LM131 to compute while converting. For instance, it will convert the ratio of two voltages to a frequency proportional to the ratio (Figure 4). The circuit is still a V/F converter, but has two signal inputs. both of them going to rather unorthodox places at that. The inputs, shown as voltages, are converted to currents by two current pumps (voltage-to-current converters). Of course. if currents of the proper ranges are available, the current pumps aren't needed. The left current pump, which includes  $Q1$  and A1, determines how fast capacitor  $C_1$  discharges between output pulses. The other pump sets the current in the reference circuit to control the amount of recharge current when the one-shot fires. Tying the comparator input, pin 7, to the reference pin sets the comparator's trip point at a constant voltage.

#### \*Stable components with low tempco



TL/H/8742-4

# FIGURE 4. This circuit converts the ratio of two voltages to an equivalent frequency without a separate analog divider. Full-scale output is 15 kHz. The two op amp circuits convert the inputs to proportional currents.

To get an idea of how the circuit works, consider first the effect of, for instance, tripling the input voltage, V1. This make C<sub>1</sub> discharge to the comparator trip point three times as fast, so the frequency triples. Next, consider a given change, such as doubling the voltage at the other input. V2. This doubles the recharge current to CL during the fixedwidth output pulse, which means CL's voltage increases twice as much during recharging. Since the discharge into Q1 is linear (for V1 constant), it takes twice as long for CL to discharge-the frequency becomes half of what it was before.

Although the current pumps in Figure 4 must have negative inputs, rearranging the op amps according to Figure 5 makes them accept positive inputs instead. Trimming out the offset in the op amp gives the ratio converter better

linearity and accuracy. The trim circuit in Figure 5a needs stable positive and negative supplies for the offset trimmer, while the one in *Figure 5b* needs only a stable positive supply. Unmarked components in Figure 5b are the same as in Figure 5a.



R1, R2, R3: Stable components with low tempco  $Q1: B \geq 330$ 



TL/H/8742-6

# FIGURE 5. These current pumps adapt the converter circuits in Figures 4 and  $6$  to positive input voltages. Optional offset trimming improves linearity and accuracy, especially with input signals that have a wide dynamic range.

h

Note that the full-scale range of the current pumps can be changed by varying the value of the input resistor(s). If either of these pump circuits is used with a single positive supply, the op amp should be a type such as 1/2 LM358 or 1/4 LM324, which has a common-mode range that includes the negative-supply bus.

#### **COMPUTING SQUARE ROOTS IMPLICITLY**

An analog divider computes the square root of a signal when the signal is fed to the divider's numerator input, and the output is fed back to the divider's denominator input.



This type of computation is called implicit, because the end result of the computation is only implied, not explicitly stated by the equation that defines the computation.

In the implicit square root computing loop described in the text. a V/F converter serves as a divider. Since it's a converter, its inputs are voltages (or currents), but its output is a frequency. To connect its output back to one of its inputs so it will compute a square root means that its output frequency must be converted back to a voltage. This is taken care of by the frequency-to-voltage converter.



Doing some algebraic substitution shows that:

$$
f_{\text{OUT}} = k3 \times \sqrt{V_{\text{IN}}}
$$

where

$$
k3 = \sqrt{k1/k2}
$$

# **IT'LL TAKE RECIPROCALS**

Taking the ratio of two inputs-in other words, doing division-is only one of the mathematical operations that can be combined with converting. Another one is a special case of division, which is taking reciprocals. In this instance, the numerator (V1 in Figure 4) is held constant, and the denominator, V2, changes over a wide range such as one or two decades. In this case, since the frequency is the reciprocal of the input, the period of the output is proportional to the input. When operated this way, the V2 current pump should have an offset trimmer. A constant current circuit is still needed to discharge capacitor C<sub>1</sub>.

Nonlinearity (that is, deviation from the ideal law) with an LM331 is a little better than 1% for 10 kHz full-scale. Increasing  $C_T$  to 0.1  $\mu$ F reduces the nonlinearity to below 0.2% while decreasing full-scale output to 1 kHz.

Two inputs can also be multiplied while converting to a frequency. The multiplying converter circuit (Figure 6) that

does this has a more elaborate current pump than the ratio circuit of *Figure 4*. This pump is really two cascaded circuits; it includes op amps A2 and A3 as well as transistors Q2 and Q3. Current from this pump goes to pin 5 to control the oneshot's pulse width. (This current ranges from 13.3  $\mu$ A to  $1.33$  mA.)

As in the ratio circuit, the left current pump controls the discharge rate of C<sub>L</sub>. The other pump, however, controls the one-shot's pulse width to vary the amount that CL charges during the pulse. If the V2 input is close to zero, the current from the pump into pin 5 is small, and the one-shot develops a wide pulse. This allows C<sub>L</sub> to charge quite a bit. It takes a relatively long time for C<sub>L</sub> to discharge to the comparator threshold, so the resulting frequency is low. As V2 goes negative (a greater absolute magnitude), the output frequency rises. Op amp A3 must have a common-mode range that extends to the positive supply voltage, which the specified types do.

Multiplying, dividing and converting can all be done at the same time by combining the V2 input current pump of Figure 4 with the circuit of Figure 6. If a scale-factor trimmer is needed, R4 in Figure 6 is a good choice, better than input resistors such as R1 or R2. Using the latter as trimmers would make the input impedance of the circuit change with trim setting.

Two V/F converter ICs along with some extra circuitry will take the square root of a voltage input. Square root functions are used mostly to simulate natural laws, but also to linearize functions that have a natural square-law relationship. One of the latter is converting diffential pressure to flow, where flow is proportional to the square root of differential pressure.



TI /H/R742-9

FIGURE 6. The product of two input voltages becomes an equivalent frequency in this converter. A current pump that includes op amps A2 and A3 controls the pulse duration of the converter's internal one-shot.

# **VERSATILE PIN FUNCTIONS GIVE DESIGN FLEXIBILITY**

Two features-the reference and the one-shot-of the LM131/LM331 V/F converter deserve a closer look because they are the key to its versatility. The simplified schematic of the chip, shown here along with a transducer and the components needed for a basic V/F converter, will help to illustrate how these features work.

The reference circuit, connected to pin 2, is both a constant voltage output and a current setting, scale-factor control input. The constant voltage can supply external circuitry, such as the transducer, that feeds the converter's input.

One great advantage of using the converter's internal reference to supply the external circuitry is that any variation in the reference voltage affects the sensitivities of the converter and the external circuitry by equal and opposite amounts, so the effects of the variation cancel.

While providing a constant voltage output, pin 2 also provides scale-factor, or sensitivity control for the converter. Current supplied to an external circuit by this terminal comes from the supply (V<sub>S</sub>) through the current mirror and the transistor. The op amp drives this transistor to hold pin 2 at a constant voltage equal to the internal reference, which is nominally 1.9V.

The current mirror provides a current to the switch that's essentially identical to that in pin 2. This means that a resistor to ground or a signal from a current source will set the current that is switched to pin 1. In most circuits, a capacitor goes from pin 1 to ground, and the switched current from this pin recharges the capacitor during the pulse from the one-shot.

The one-shot circuit is somewhat like the well known 555 timer's circuit. In the quiescent state, the reset transistor is on and holds pin 5 near ground. When pin 7 becomes more positive than pin 6 (or pin 6 falls below pin 7), the input comparator sets the flip-flop in the one-shot.

The flip-flop turns on the current limited output transistor (pin 3) and switches the current coming from the current mirror to pin 1. The flip-flop also turns off the reset transistor, and the timing capacitor  $C_T$  starts to charge toward  $V_S$ . This charge is exponential, and  $C_T$ 's voltage reaches 2/3 of  $V_S$  in about 1.1  $R_T C_T$  time constants. (The quantity 1.1 is -In 0.333...) When pin 5 reaches this voltage, the oneshot's comparator resets the flip-flop which turns off the current to pin 1, discharges  $C_T$ , and turns off the output transistor.

If the voltages at pins 6 and 7 still call for setting the flip-flop after pin 5 has reached 2/3 V<sub>S</sub>, internal logic not shown in this simplified diagram overrides the reset signal from the one-shot's own comparator, and the flip-flop stays set. In this instance,  $C_T$  continues charging past 2/3 Vs.



# **ROOT LOOP COMPUTES**

The circuit in *Figure 7* is an implicit loop (see "Computing Square Roots Implicitly") that uses IC1 as a voltage-to-frequency converter and divider, and IC2 as a frequency-tovoltage converter. The F/V converter, IC2, and the current pump that includes A1 and the transistor return the output of IC1 to its denominator input. A relatively elaborate feedback circuit like this is needed to convert IC1's frequency output back to a current for its denominator input.

Looking at the circuit in more detail, IC1 puts out a frequency proportional to V<sub>IN</sub> divided by the feedback voltage, V<sub>X</sub>. The current  $I_1$  is generated by a current pump that has  $V_X$ as its input (Figure 5a). To develop the feedback IC2 converts the pulse output from IC1 into standard width precision current pulses that charge capacitor C1. This capacitor integrates them into the voltage  $V_X$ , thus closing the loop.

Op amp A2, serving as a comparator, ensures that the circuit will always start and continue running. If V<sub>IN</sub> suddenly jumps to a higher voltage, one pulse from the one-shot in IC1 may not be enough to recharge  $C_1$  to a voltage higher than the input. In such a case, the IC's internal logic keeps its internal current switch turned on, and the voltage on C ramps up until it exceeds the input. During this time, however, IC1's output hasn't changed state. (Such a temporary hang-up isn't unique to this circuit, and equivalent things happen to other V/Fs besides the LM131/LM331.) What is worse here, though, is that the lack of pulses to IC2 means that  $V_X$  and  $I_1$  decay. The recharging current,  $I_2$ , is the same as I<sub>1</sub>, so it not only becomes progressively harder for the voltage on C<sub>L</sub> to catch up with the input, it may even fail to catch up entirely if  $(|_2 \times R_1|)$  is less than the input voltage.

As a sign of this condition, when the converter hangs up, the one-shot's timing node, pin 5, continues to charge well beyond its normal peak of 2/3 Vs. As soon as the comparator A2 detects this rise, it pulls up voltage  $V_X$ , current  $I_1$ increases, and the loop catches its breath again.

After all these nonlinear computations, this last circuit is about as linear as it can be. It's a precision, ultralinear V/F converter based on an LM331A (Figure 8) that has several detail refinements over previous V/F converter circuits. Choosing the proper components and trimming the tempco give less than 0.02% error and 0.003% nonlinearity for a  $±20^{\circ}$ C range around room temperature.

This circuit has an active integrator, which includes the op amp and the integrating feedback capacitor, CF. The integrator converts the input voltage, which is negative, into a positive-going ramp. When the ramp reaches the converter IC's comparator threshold, the one-shot fires and switches a pulse of current to the integrator's summing junction. This current makes the integrator's output ramp down quickly. When the one-shot times out, the cycle repeats.

There are several reasons this converter circuit gives high performance:

- A feedback limiter prevents the op amp from driving pin 7 of the LM331A negative. The limiter circuit arrangement bypasses the leakage through CR5 to ground via R5, so it won't reach the summing junction. Bypassing leakage this way is especially important at high temperatures.
- The offset trimming pot is connected to the stable 1.9V reference at pin 2 instead of to a power supply bus that might be unstable and noisy.



TL/H/8742-11

FIGURE 7. Two converter ICs generate an output frequency proportional to the square root of the input voltage. The circuit is an implicit loop in which IC1 serves as a divider and V/F converter. This IC's output goes back to its denominator input through F/V converter IC2 to make the circuit output equal the input's square root.



# FIGURE 8. An ultraprecision V/F converter, capable of better than 0.02% error and 0.003% nonlinearity for a ±20°C range about room temperature, augments the basic converter with an external integrator.

- A small fraction (180  $\mu$ V, full-scale) of the input voltage goes via R4 to the R<sub>S</sub> network, which improves the nonlinearity from 0.004% to 0.002%.
- . Resistors R2 and R3 are the same value, so that resistors such as Allen-Bradley type CC metal-film types can provide excellent tempco tracking at low cost. (This tracking is very good when equal values come from the same batch.) Resistor R1 should be a low tempco metalfilm or wirewound type, with a maximum tempco of  $± 10$  ppm/ $°C$  or  $± 25$  ppm/ $°C$ .

In addition. C<sub>T</sub> should be a polystyrene or Teflon type. Polystyrene is rated to 80°C, while Teflon goes to 150°C. Both types can be obtained with a tempco of  $-110 \pm 30$  ppm/°C. Choosing this tempco for  $C_T$  makes the tempco, due to  $C_T$ , of the full-scale output frequency 110 ppm/°C.

Using tight tolerance components results in a total tempco between 0 ppm/°C and 220 ppm/°C, so the tempco will never be negative. The voltage at CR1 and R<sub>X</sub> has a tempco of  $-6$  mV/°C, which can be used to compensate the tempco of the rest of the circuit. Trimming  $R_X$  compensates for the tempco of the V/F IC, the capacitor, and all the resistors.

A good starting value for selecting R<sub>X</sub> is 430 k $\Omega$ , which will give the 135 µA flowing out of pin 2 a slope of 110 ppm/°C. If the output frequency increases with temperature, a little more conductance should be added in parallel with R<sub>X</sub>.

When doing a second round of trimming, though, note that a resistor of, say, 4.3 M $\Omega$ , has about the same effect on tempco when shunted across a 220 k $\Omega$  resistor that it does when shunted across one of 430 k $\Omega$ , namely, -11 ppm/°C. This technique can give tempcos below  $\pm 20$  ppm/°C or even  $± 10$  ppm/ $°C$ .

Some precautions help this procedure converge:

- 1. Use a good capacitor for  $C_T$ . The cheapest polystyrene capacitors will shift in value by 0.05% or more per temperature cycle. The actual temperature sensitivity would be indistinguishable from the hysteresis, and the circuit would never be stable.
- 2. After soldering, bake and/or temperature-cycle the circuit (at a temperature not exceeding 75°C if CT is polystyrene) for a few hours, to stabilize all components and to relieve the strains from soldering.
- 3. Don't rush the trimming. Recheck the room temperature value, before and after the high temperature data are taken, to ensure that hysteresis per cycle is reasonably low.
- 4. Don't expect a perfect tempco at  $-25^{\circ}$ C if the circuit is trimmed for ±5 ppm/°C between 25°C and 60°C. If it's been trimmed for zero tempco while warm, none of its components will be linear to much better than 5 ppm/°C or 10 ppm/°C when it's cold.

The values shown in this circuit are generally optimum for  $±12V$  to  $±16V$  regulated supplies but any stable supplies between  $\pm$  4V and  $\pm$  22V would be usable, after changing a few component values.

# **The Monolithic Operational Amplifier: A Tutorial Study**

# Invited Paper-

# IEEE Journal of Solid-State Circuits, Vol. SC-9, No. 6

Abstract-A study is made of the integrated circuit operational amplifier (IC op amp) to explain details of its behavior in a simplified and understandable manner. Included are analyses of thermal feedback effects on gain, basic relationships for bandwidth and slew rate, and a discussion of polesplitting frequency compensation. Sources of second-order bandlimiting in the amplifier are also identified and some approaches to speed and bandwidth improvement are developed. Brief sections are included on new JFET-bipolar circuitry and die area reduction techniques using transconductance reduction.

# **1.0 INTRODUCTION**

The integrated circuit operational amplifier (IC op amp) is the most widely used of all linear circuits in production today. Over one hundred million of the devices will be sold in 1974 alone, and production costs are falling low enough so that op amps find applications in virtually every analog area. Despite this wide usage, however, many of the basic performance characteristics of the op amp are poorly understood.

National Semiconductor Appendix A



It is the intent of this study to develop an understanding for op amp behavior in as direct and intuitive a manner as possible. This is done by using a variety of simplified circuit models which can be analyzed in some cases by inspection, or in others by writing just a few equations. These simplified models are generally developed from the single representative op amp configuration shown in Figures 1 and 2.

The rationale for starting with the particular circuit of Figure 1 is based on the following: this circuit contains, in simplified form, all of the important elements of the most commonly used integrated op amps. It consists essentially of two yoltage gain stages, an input differential amp and a common emitter second stage, followed by a class-AB output emitter follower which provides low impedance drive to the load. The two interstages are frequency compensated by a single small "pole-splitting" capacitor (see below) which is usually included on the op amp chip. In most respects this circuit is directly equivalent to the general purpose LM101 [1],  $\mu$ A 741 [2], and the newer dual and quad op amps [3], so the results of our study relate directly to these devices. Even for









more exotic designs, such as wide-band amps using feedforward [4], [5], or the new FET input circuits [6], the basic analysis approaches still apply, and performance details can be accurately predicted. It has also been found that a good understanding of the limitations of the circuit in Figure 1 provides a reasonable starting point from which higher performance amplifiers can be developed.

The study begins in Section 2, with an analysis of dc and low frequency gain. It is shown that the gain is typically limited by thermal feedback rather than electrical characteristics. A highly simplified thermal analysis is made, resulting in a gain equation containing only the maximum output current of the op amp and a thermal feedback constant.

The next three sections apply first-order models to the calculation of small-signal high frequency and large-signal slewing characteristics. Results obtained include an accurate equation for gain-bandwidth product, a general expression for slew rate, some important relationships between slew rate and bandwidth, and a solution for voltage follower behavior in a slewing mode. Due to the simplicity of the results in these sections, they are very useful to designers in the development of new amplifier circuits.

Section 6 applies more accurate models to the calculation of important second-order effects. An effort is made in this section to isolate all of the major contributors to bandlimiting in the modern amp.

In the final section, some techniques for reduction of op amp die size are considered. Transconductance reduction and layout techniques are discussed which lead to fabrication of an extremely compact op amp cell. An example vielding 8000 possible op amps per 3-in. wafer is given.

# **2.0 GAIN AT DC AND LOW FREQUENCIES**

# A. The Electronic Gain

The electronic voltage gain will first be calculated at dc using the circuit of Figure 1. This calculation becomes straightforward if we employ the simplified transistor model shown in Figure 3(a). The resulting gain from Figure 3(b) is

$$
A_V(0) = \frac{v_{\text{out}}}{v_{\text{in}}} \cong \frac{g_{\text{m1}} \beta_5 \beta_6 \beta_7 R_L}{1 + r_{\text{I2}} / r_{01}} \tag{1}
$$

where

$$
a_2 \approx \beta_5(r_{e5} + \beta_6r_{e6})
$$

$$
r_{01}' \approx r_{04}/r_{02}.
$$

It has been assumed that

$$
\beta_7 R_L < r_{06} / r_{09}, g_{m1} = g_{m2}, \beta_7 = \beta_8
$$

The numerical subscripts relate parameters to transistor Q numbers (i.e.,  $r_{e5}$  is  $r_e$  of  $Q_5$ ,  $\beta_6$  is  $\beta_0$  pf  $Q_6$ , etc.). It has also been assumed that the current mirror transistors Q<sub>3</sub> and Q<sub>4</sub> have  $\alpha$ 's of unity, and the usually small loading of R<sub>B</sub> has been ignored. Despite the several assumptions made in obtaining this simple form for (1), its accuracy is quite adequate for our needs.

An examination of (1) confirms the way in which the amplifier operates: the input pair and current mirror convert the input voltage to a current  $g_{m1}V_{in}$  which drives the base of the second stage. Transistors Q<sub>5</sub>, Q<sub>6</sub>, and Q<sub>7</sub> simply multiply this current by  $\beta^3$  and supply it to the load R<sub>1</sub>. The finite output resistance of the first stage causes some loss when compared with second stage input resistance, as indicated by the term  $1/(1 + r_{i2}/r_{01})$ . A numerical example will help our perspective: for the LM101A,  $I_1 \cong 10 \mu A$ ,  $I_2 \cong 300 \mu A$ ,  $\beta_5 = \beta_6 \approx 150$ , and  $\beta_7 \approx 50$ . From (1) and dc voltage gain with  $R_L = 2 k\Omega$  is

$$
\lambda_V(0) \cong 625,000 \tag{2}
$$

The number predicted by (2) agrees well with that measured on a discrete breadboard of the LM101A, but is much higher than that observed on the integrated circuit. The reason for this is explained in the next section.

## **B. Thermal Feedback Effects on Gain**

The typical IC op amp is capable of delivering powers of 50-100 mW to a load. In the process of delivering this power, the output stage of the amp internally dissipates similar power levels, which causes the temperature of the IC chip to rise in proportion to the output dissipated power. The silicon chip and the package to which it is bonded are good thermal conductors, so the whole chip tends to rise to the same temperature as the output stage. Despite this, small



FIGURE 3. (a) Approximate  $\pi$  model for CE transistor at dc. Feedback element  $r_\mu \cong \beta_4 r_0$  is ignored since this greatly simplifies hand calculations. The error caused is usually less than 10 percent because  $\beta_4$ , the intrinsic  $\beta$  under the emitter, is quite large. Base resistance ry is also ignored for simplicity. (b) Circuit illustrating calculation of electronic gain for op amp of *Figure 1.* Consideration is given only to the fully loaded condition (R<sub>L</sub>  $\cong$  2 kΩ) where  $\beta_7$  is falling (to about 50) due to high current density. Under this condition, the output resistance of Q6 and Q9 are nondominant.

temperature gradients from a few tenths to a few degrees centigrade develop across the chip with the output section being hotter than the rest. As illustrated in Figure 4, these temperature gradients appear across the input components of the op amp and induce an input voltage which is proportional to the output dissipated power.

To a first order, it can be assumed that the temperature difference  $(T_2 - T_1)$  across a pair of matched and closely spaced components is given simply by

$$
(\mathsf{T}_2 - \mathsf{T}_1) \cong \pm \mathsf{K}_\mathsf{T} \mathsf{P}_\mathsf{d} \quad \text{°C} \tag{3}
$$

where

P<sub>d</sub> power dissipated in the output circuit,

 $K<sub>T</sub>$  a constant with dimensions of °C/W.

The plus/minus sign is needed because the direction of the thermal gradient is unknown. In fact, the sign may reverse polarity during the output swing as the dominant source of heat shifts from one transistor to another. If the dominant input components consist of the differential transistor pair of Figure 4, the thermally induced input voltage V<sub>int</sub> can be calculated as

$$
V_{int} \cong \pm K_T P_d (2 \times 10^{-3})
$$
  
\n
$$
\cong \pm \gamma_T P_d
$$
 (4)

where  $\gamma_T = K_T(2 \times 10^{-3})$  V/W, since the transistor emitter-base drops change about -2 mV/°C.

For a thermally well designed IC op amp, in which the power output devices are made to approximate either a point or a line source and the input components are placed on the resulting isothermal lines (see below and Figure 8), typical values measured for K<sub>T</sub> are

$$
K_{\text{T}} \approx 0.3^{\circ}\text{C/W} \tag{5}
$$

in a TO-5 package.

١

The dissipated power in the class-AB output stage Pd is written by inspection of Figure 4:

$$
P_{d} = \frac{V_{0}V_{s} - V_{0}^{2}}{R_{L}} \tag{6}
$$

where

$$
\begin{aligned} \mathsf{V}_\mathrm{s} &= +\mathsf{V}_{\mathrm{CC}} & \text{when } \mathsf{V}_0 &> 0 \\ \mathsf{V}_\mathrm{s} &= -\mathsf{V}_{\mathrm{ee}} & \text{when } \mathsf{V}_0 < 0. \end{aligned}
$$

A plot of (6) is Figure 5 resembles the well-known class-AB dissipation characteristics, with zero dissipation occurring



FIGURE 4. Simple model illustrating thermal feedback in an IC op amp having a single dominant source of self-heat, the output stage. The constant  $\gamma_T \approx 0.6$  mV/W and P<sub>d</sub> is power dissipated in the output. For simplicity, we ignore input drift due to uniform heating of the package. This effect can be significant if the input stage drift is not low, see [7].





for  $V_0 = 0$ ,  $+V_{cc}$ ,  $-V_{ee}$ . Dissipation peaks occur for  $V_0 =$ + $V_{cc}/2$  and  $-V_{ee}/2$ . Note also from (4) that the thermally induced input voltage V<sub>int</sub> has this same double-humped shape since it is just equal to a constant times P<sub>d</sub> at dc.

Now examine Figures 6(a) and (b) which are curves of open-loop  $V_0$  versus  $V_{in}$  for the IC op amp. Note first that the overall curve can be visualized to be made up of two components: a) a normal straight line electrical gain curve of the sort expected from (1) and b) a double-humped curve similar to that of Figure 5. Further, note that the gain characteristic has either positive or negative slope depending on the value of output voltage. This means that the thermal feedback causes the open-loop gain of the feedback amplifier to change phase by 180°, apparently causing negative feedback to become positive feedback. If this is really true, the question arises: which input should be used as the inverting one for feedback? Further, is there any way to close

the amplifier and be sure it will not find an unstable operating point and latch to one of the power supplies?

The answers to these questions can be found by studying a simple model of the op amp under closed-loop conditions, including the effects of thermal coupling. As shown in Figure 7, the thermal coupling can be visualized as just an additional feedback path which acts in parallel with the normal electrical feedback. Noting that the electrical form of the thermal feedback factor is [see (4) and (6)]

$$
\beta_{\mathsf{T}} = \frac{\partial \mathsf{V}_{\mathsf{int}}}{\partial \mathsf{V}_{\mathsf{0}}} = \pm \frac{\gamma_{\mathsf{T}}}{\mathsf{R}_{\mathsf{L}}} (\mathsf{V}_{\mathsf{S}} - 2\mathsf{V}_{\mathsf{0}}). \tag{7}
$$

The closed-loop gain, including thermal feedback is

$$
A_V(0) = \frac{\mu}{1 + \mu(\beta_e \pm \beta_{\mathsf{T}})}
$$
(8)





where  $\mu$  is the open-loop gain in the absence of thermal feedback  $[(1)]$  and  $\beta_e$  is the applied electrical feedback as in Figure 7. Inspection of (8) confirms that as long as there is sufficient electrical feedback to swamp the thermal feedback (i.e.,  $\beta_e > \beta_T$ ), the amplifier will behave as a normal closed-loop device with characteristics determined principally by the electrical feedback (i.e.,  $A_V(0) \approx 1/\beta_e$ ). On the other hand, if  $\beta_e$  is small or nonexistant, the thermal term in (8) may dominate, giving an apparent open-loop gain characteristic determined by the thermal feedback factor  $\beta_{\text{T}}$ . Letting  $\beta_e = 0$  and combining (7) and (8), A<sub>V</sub>(0) becomes

$$
A_V(0) = \frac{\mu}{1 \pm \frac{\mu \gamma T}{D} (V_s - 2V_0)}
$$
(9)

Recalling from (6) that  $V_0$  ranges between 0 and  $V_S$ , we note that the incremental thermal feedback is greatest when  $V_0 = 0$  or  $V_s$ , and it is at these points that the thermally limited gain is smallest. To use the amplifier in a predictable manner, one must always apply enough electrical feedback to reduce the gain below this minimum thermal gain. Thus, a maximum usable gain can be defined as that approximately equal to the value of (9) with  $V_0 = 0$  or  $V_s$  which is

 $A_V(0)|_{max} \approx \frac{R_L}{\gamma_T V_S}$ 

Ωľ

$$
A_V(0)|_{\text{max}} \simeq \frac{1}{\gamma \tau!_{\text{max}}}.\tag{11}
$$

 $(10)$ 

It was assumed in (10) and (11) that thermal feedback dominates over the open-loop electrical gain,  $\mu$ . Finally, in (11) a maximum current was defined  $I_{max} = V_S/R_L$  as the maximum current which would flow if the amplifier output could swing all the way to the supplies.

Equation (11) is a strikingly simple and quite general result which can be used to predict the expected maximum usable gain for an amplifier if we know only the maximum output current and the thermal feedback constant  $\gamma_T$ .

Recall that typically  $K_T \cong 0.3^{\circ}$ C/W and  $\gamma_T = (2 \times 10^{-3})$  $K_T \approx 0.6$  mV/W. Consider, as an example, the standard IC op amp operating with power supplies of  $V_S = \pm 15V$  and a minimum load of 2 k $\Omega$ , which gives  $I_{\text{max}} = 15V/2 k\Omega = 7.5$ mA. Then, from (11), the maximum thermally limited gain is about:

$$
A_V(0)|_{\text{max}} \cong 1/(0.6 \times 10^{-3})(7.5 \times 10^{-3})
$$
  
\n
$$
\cong 220,000.
$$
 (12)

Comparing (2) and (12), it is apparent that the thermal characteristics dominate over the electrical ones if the minimum load resistor is used. For loads of 6 k $\Omega$  or more, the electrical characteristics should begin to dominate if thermal feedback from sources other than the output stage is negligible. It should be noted also that, in some high speed, high drain op amps, thermal feedback from the second stage dominates when there is no load.

As a second example, consider the so-called "power op amp" or high gain audio amp which suffers from the same thermal limitations just discussed. For a device which can deliver 1W into a 16 $\Omega$  load, the peak output current and voltage are 350 mA and 5.7V. Typically, a supply voltage of about 16V is needed to allow for the swing loss in the IC output stage. I<sub>max</sub> is then 8V/16 $\Omega$  or 0.5A. If the device is in a TO-5 package  $\gamma_T$  is approximately 0.6 mV/W, so from (11) the maximum usable dc gain is

$$
A_V(0)|_{\text{max}} \simeq \frac{1}{(0.6 \times 10^{-3})(0.5)} \simeq 3300. \tag{13}
$$

This is quite low compared with electrical gains of, say, 100,000 which are easily obtainable. The situation can be improved considerably by using a large die to separate the nower devices from the inputs and carefully placing the inputs on constant temperature (isothermal) lines as illustrated in *Figure 8*. If one also uses a power package with a



TL/H/8745-11

# FIGURE 8. One type layout in which a quad of input transistors is cross connected to reduce effect of nonuniform thermal gradients. The output transistors use distributed stripe geometrics to generate predictable isothermal lines.

heavy copper base,  $\gamma$ <sup>-</sup>'s as low as 50  $\mu$ V/W have been observed. For example, a well-designed 5W amplifier driving an  $8\Omega$  load and using a 24V supply, would have a maximum gain of 13,000 in such a power package.



FIGURE 9. First-order model of op amp used to calculate small signal high frequency gain. At frequencies of interest the input impedance of the second stage becomes low compared to first stage output impedance due to C<sub>c</sub> feedback. Because of this, first stage output impedance can be assumed infinite, with no loss in accuracy.





As a final comment, it should be pointed out that the most commonly observed effect of thermal feedback in high gain circuits is low frequency distortion due to the nonlinear transfer characteristic. Differential thermal coupling typically falls off at an initial rate of 6 dB/octave starting around 100-200 Hz, so higher frequencies are uneffected.

## 3.0 SMALL-SIGNAL FREQUENCY RESPONSE

At higher frequencies where thermal effects can be ignored, the behavior of the op amp is dependent on purely electronic phenomena. Most of the important small and large signal performance characteristics of the classical IC op amp can be accurately predicted from very simple first-order models for the amplifier in *Figure 1* (8). The small-signal model that is used assumes that the input differential amplifier and current mirror can be replaced by a frequency independent voltage controlled current source, see Figure 9. The second stage consisting essentially of transistors Q<sub>5</sub> and Q<sub>6</sub>, and the current source load, is modeled as an ideal frequency independent amplifier block with a feedback or "integrating capacitor" identical to the compensation capacitor, c<sub>c</sub>. The

output stage is assumed to have unity voltage gain and is ignored in our calculations. From Figure 9, the high frequency gain is calculated by inspection:

$$
A_V(\omega) = \left| \frac{V_0}{V_i}(s) \right| = \left| \frac{g_{m1}}{sC_c} \right| = \frac{g_{m1}}{\omega C_c}
$$
 (14)

where dc and low frequency behavior have not been included since this was evaluated in the last section. Figure 10 is a plot of the gain magnitude as predicted by (14). From this figure it is a simple matter to calculate the open-loop unity gain frequency  $\omega_{11}$ , which is also the gain-bandwidth product for the op amp under closed-loop conditions:

$$
\omega_{\mathsf{u}} = \frac{\mathsf{g}_{\mathsf{m1}}}{\mathsf{C}_{\mathsf{c}}} \,. \tag{15}
$$

In a practical amplifier,  $\omega_u$  is set to a low enough frequency (by choosing a large C<sub>c</sub>) so that negligible excess phase over the 90° due to C<sub>c</sub> has built up. There are numerous contributors to excess phase including low  $f_1$  p-n-p's, stray capacitances, nondominant second stage poles, etc.

These are discussed in more detail in a later section, but for now suffice it to say that, in the simple IC op amp,  $\omega_0/2\pi$  is limited to about 1 MHz. As a simple test of (15), the LM101 or the  $\mu$ A741 has a first stage bias current  $I_1$  of 10  $\mu$ A per side, and a compensation capacitor for unity gain operation, C<sub>c</sub>, of 30 pF. These amplifiers each have a first stage g<sub>m</sub> which is half that of the simple differential amplifer in Figure 1 so  $g_{m1} = qi_1/2kT$ . Equation (15) then predicts a unity gain corner of

$$
f_{\rm u} = \frac{\omega_{\rm u}}{2\pi} = \frac{g_{\rm m1}}{2\pi C_{\rm c}} = \frac{(0.192 \times 10^{-3})}{2\pi (30 \times 10^{-12})} = 1.02 \text{ MHz (16)}
$$

which agrees closely with the measured values



TI /H/8745-14



# 4.0 SLEW RATE AND SOME SPECIAL LIMITS

#### A. A General Limit on Slew Rate

If an op amp is overdriven by a large-signal pulse or square wave having a fast enough rise time, the output does not follow the input immediately. Instead, it ramps or "slews" at some limiting rate determined by internal currents and capacitances, as illustrated in Figure 11. The magnitude of input voltage required to make the amplifier reach its maximum slew rate varies, depending on the type of input stage used. For an op amp with a simple input differential amp, an input of about 60 mV will cause the output to slew at 90 percent of its maximum rate, while a  $\mu$ A741, which has half the input g<sub>m</sub>, requires 120 mV. High speed amplifiers such as the LM118 or a FET-input circuit require much areater overdrive, with 1-3V being common. The reasons for these overdrive requirements will become clear below.

An adequate model to calculate slew limits for the representative op amp in the inverting mode is shown in Figure 12, where the only important assumption made is that  $1<sub>2</sub> \ge 21<sub>1</sub>$  in *Figure 1*. This condition always holds in a welldesigned op amp. (If one lets  $I_2$  be less than  $2I_1$ , the slew is limited by l<sub>2</sub> rather than l<sub>1</sub>, which results in lower speed than is otherwise possible.) Figure 12 requires some modification for noninverting operation, and we will study this later.

The limiting slew rate is now calculated from Fig. 12. Letting the input voltage be large enough to fully switch the input differential amp, we see that all of the first stage tail current 21<sub>1</sub> is simply diverted into the integrator consisting of A and C<sub>c</sub>. The resulting slew rate is then:

$$
slew rate = \frac{dv_0}{dt}\bigg|_{max} = \frac{i_c(t)}{C_c}.
$$
 (17)

Noting that  $i_c(t)$  is a constant  $2i_1$ , this becomes

$$
\left. \frac{d\mathsf{v}_0}{dt} \right|_{\mathsf{max}} = \frac{2l_1}{C_c}.
$$
 (18)

As a check of this result, recall that the  $\mu$ A741 has  $I_1$  = 10  $\mu$ A and C<sub>1</sub> = 30 pF, so we calculate:

$$
\left. \frac{dv_0}{dt} \right|_{\text{max}} = \frac{2 \times 10^{-5}}{30 \times 10^{-12}} = 0.67 \frac{V}{\mu s} \tag{19}
$$

which agrees with measured values.



TI /H/8745-15

FIGURE 12. Model used to calculate slew rate for the amp of Figure 1 in the inverting mode. For simplicity, all transistor  $a$ 's are assumed equal to unity, although results are essentially independent of  $a$ . An identical slew rate can be calculated for a negative-going output, obtained if the applied input polarity is reversed.

The large and small signal behavior of the op amp can be usefully related by combining (15) for  $\omega_{\rm H}$  with (18). The slew rate becomes

$$
\left.\frac{dv_0}{dt}\right|_{\text{max}} = \frac{2\omega_u I_1}{g_{\text{m1}}}.
$$
 (20)

Equation (20) is a general and very useful relationship. It shows that, for a given unity-gain frequency,  $\omega_{\rm U}$ , the slew rate is determined entirely by just the ratio of first stage operating current to first stage transconductance, I<sub>1</sub>/g<sub>m1</sub>. Recall that  $\omega_{11}$  is set at the point where excess phase beains to build up, and this point is determined largely by technology rather than circuit limitations. Thus, the only effective means available to the circuit designer for increasing op amp slew rate is to *decrease* the ratio of first stage transconductance to operating current,  $g_{m1}/_1$ .

# B. Slew Limiting for Simple Bipolar Input Stage

The significance of (20) is best seen by considering the specific case of a simple differential bipolar input as in Figure 1. For this circuit, the first stage transconductance (for  $\alpha_1$  =  $1)$  is<sup>1</sup>

$$
g_{m1} = qI_1/kT \tag{21}
$$

so that

$$
\frac{9m_1}{11} = q/kT.
$$
 (22)

Using this in (20), the maximum bipolar slew rate is

$$
\left. \frac{dv_0}{dt} \right|_{\text{max}} = 2\omega_u \frac{kT}{q}.
$$
 (23)

This provides us with the general (and somewhat dismal) conclusion that slew rate in an op amp with a simple bipolar input stage is dependent only upon the unity gain corner and fundamental constants. Slew rate can be increased only by incerasing the unity gain corner, which we have noted is generally difficult to do. As a demonstration of the severity of this limit, imagine an op amp using highly advanced technology and clever design, which might have a stable unity gain frequency of 100 MHz. Equation (23) predicts that the slew rate for this advanced device is only

$$
\left. \frac{dv_0}{dt} \right|_{max} = 33 \frac{V}{\mu s}
$$
 (24)

which is good, but hardly impressive when compared with the difficulty of building a 100 MHz op amp.<sup>2</sup> But, there are some ways to get around this limit as we shall see shortly.

#### **C. Power Bandwidth**

Our intuition regarding slew rate will be enhanced somewhat if we relate it to a term called "power bandwidth". Power bandwidth is defined as the maximum frequency at which full output swing (usually 10V peak) can be obtained without distortion. For a sinusoidal output voltage  $v_0(t)$  =  $V_{\text{o}}$ sin $\omega$ t, the rate of change of output, or slew rate, required to reproduce the output is

$$
\frac{dv_0}{dt} = \omega V_p \cos \omega t.
$$
 (25)

This has a maximum when  $\cos \omega t = 1$  giving d'

$$
\left.\frac{v_0}{\text{rt}}\right|_{\text{max}} = \omega V_{\text{p}},\tag{26}
$$

so the highest frequency that can be reproduced without slew limiting,  $\omega_{\text{max}}$  (power bandwidth) is

$$
\omega_{\text{max}} = \frac{1}{V_0} \frac{dv_0}{dt}\bigg|_{\text{max}}.\tag{27}
$$

Thus, power bandwidth and slew rate are directly related by the inverse of the peak of the sine wave  $V_p$ . Figure 13 shows the severe distortion of the output sine wave which results if one attempts to amplify a sine wave which results if one attempts to amplify a sine wave of frequency  $\omega$  >  $\omega_{\text{max}}$ 

1Note that (21) applies only to the simple differential input stage of Figure 12. For compound input stages as in the LM101 or  $\mu$ A741,  $g_{m1}$  is half that in (21), and the slew rate in (23) is doubled.

 $2$  We assume in all of these calculations that  $C_c$  is made large enough so that the amplifier has less than 180° phase lag at  $\omega_{\text{U}}$ , thus making the amplifier stable for unity closed-loop gain. For higher gains one can of course reduce C<sub>c</sub> (if the IC allows external compensation) and increase the slew rate according to (18).



TL/H/8745-16

FIGURE 13. Slew limiting effects on output sinewave that occur if frequency is greater than power bandwidth,  $\omega_{\text{max}}$ . The onset of slew limiting occurs very suddenly as  $\omega$  reaches  $\omega_{\text{max}}$ . No distortion occurs below  $\omega_{\text{max}}$ , while almost complete triangularization occurs at frequencies just slightly above  $\omega_{\text{max}}$ .

Some numbers illustrate typical op amp limits. For a µA741 or LM101 having a maximum slew rate of  $0.67V/\mu s$ , (27) gives a maximum frequency for an undistorted 10V peak output of

$$
f_{\text{max}} = \frac{\omega_{\text{max}}}{2\pi} = 10.7 \text{ kHz}, \qquad (28)
$$

which is a quite modest frequency considering the much higher frequency small signal capabilities of these devices. Even the highly advanced 100 MHz amplifier considered above has a 10V power bandwidth of only 0.5 MHz, so it is apparent that a need exists for finding ways to improve slew rate.



Ti /H/8745-17

FIGURE 14. Resistive degeneration used to provide slew rate enhancement according to (29).

# D. Techniques for Increasing Slew Rate

1) Resistive Enhancement of the Bipolar Stage: Equation (20) indicates that slew rate can be improved if we reduce first stage g<sub>m1</sub>/l<sub>1</sub>. One of the most effective ways of doing this is shown in *Figure 14*, where simple resistive emitter degeneration has been added to the input differential amplifier (8). With this change, the  $g_{m1}/I_1$  drops to

$$
\frac{g_{m1}}{I_1} = \frac{38.5}{1 + T_E I_1 / 26 \text{ mV}} \tag{29}
$$

at 25°C

The quantity  $g_{m1}/l_1$  is seen to decrease rapidly with added  $R_E$  as soon as the voltage drop across  $R_E$  exceeds 26 mV. The LM118 is a good example of a bipolar amplifier which uses emitter degeneration to enhance slew rate [4]. This device uses emitter resistors to produce  $R_E I_1 = 500$  mV, and has a unity gain corner of 16 MHz. Equations (20) and (29) then predict a maximum inverting slew rate of

$$
\left.\frac{\text{d}v_0}{\text{d}t}\right|_{\text{max}} = 2\omega_U \frac{I_1}{g_{\text{m1}}} = \omega_U = 100 \frac{V}{\mu s} \tag{30}
$$

which is a twenty-fold improvement over a similar amplifier without emitter resistors.

A penalty is paid in using resistive slew enhancement, however. The two added emitter resistors must match extremely well or they add voltage offset and drift to the input. In the LM118, for example, the added emitter R's have values of 2.0 k $\Omega$  each and these contribute an input offset of 1 mV for each  $40.02$  percent) of mismatch. The thermal noise of the resistors also unavoidably degrades noise performance.

2) Slew Rate in the FET Input Op Amp: The FET (JFET or MOSFET) has a considerably lower transconductance than a bipolar device operating at the same current. While this is normally considered a drawback of the FET, we note that this "poor" behavior is in fact highly desirable in applications to fast amplifiers. To illustrate, the drain current for a JFET in the "current saturation" region can be approximated by

$$
I_D \cong I_{DSS} (V_{GS}/V_T - 1)^2 \tag{31}
$$

where

 $I_{\text{DSS}}$  the drain current for  $V_{\text{GS}} = 0$ ,

- V<sub>GS</sub> the gate source voltage having positive polarity for forward gate-diode bias.
- $V_T$ the threshold voltage having negative polarity for JFET's.

The small-signal transconductance is obtained from (31) as  $g_m = \partial I_D / \partial V_G$ . Dividing by  $I_D$  and simplifying, the ratio g<sub>m</sub>/I<sub>D</sub> for a JFET is

$$
\frac{g_M}{I_D} \approx \frac{2}{(V_{GS} - V_T)} = \frac{2}{-V_T} \left[ \frac{I_{DSS}}{I_D} \right]^{1/2}.
$$
 (32)

Maximum amplifier slew rate occurs for minimum  $g_m/I_D$ and, from (32), this occurs when  $I_D$  or  $V_{GS}$  is maximum. Normally it is impractical to forward bias the gate junction so a practical minimum occurs for (32) when  $V_{GS} \approx 0V$  and  $I_D$  $\cong$  I<sub>DSS</sub>. Then

$$
\left.\frac{g_m}{I_D}\right|_{\text{min}} \simeq -2\frac{2}{V_T}.\tag{33}
$$

Comparing (33) with the analogous bipolar expression. (22). we find from (20) that the JFET slew rate is greater than bipolar by the factor

$$
\frac{\text{JFET slew}}{\text{bipolar slew}} \approx \frac{-V_T 2 \omega_{\text{uf}}}{2kT/q\omega_{\text{ub}}}
$$
(34)

where  $\omega_{\text{uf}}$  and  $\omega_{\text{ub}}$  are unity-gain bandwidths for JFET and bipolar amps, respectively. Typical JFET thresholds are around 2V ( $V_T = -2V$ ), so for equal bandwidths (34) tells us that a JFET-input op amp is about forty times faster than a simple bipolar input. Further, if JFET's are properly substituted for the slow p-n-p's in a monolithic design, bandwidth improvements by at least a factor of ten are obtainable. JFET-input op amps, therefore, offer slew rate improvements by better than two orders of magnitude when compared with the conventional IC op amp. (Similar improvements are possible with MOSFET-input amplifiers.) This characteristic, coupled with picoamp input currents and reasonable offset and drift, make the JFET-input op amp a very desirable alternative to conventional bipolar designs.

As an example, *Figure 15*, illustrates one design for an op amp employing compatible p-channel JFET's on the same chip with the normal bipolar components. This circuit exhibits a unity gain corner of 10 MHz, a 33  $V/\mu s$  slew rate, an input current of 10 pA and an offset voltage and drift of 3 mV amd 3  $\mu$ V/°C [6]. Bandwidth and slew rate are thus improved over simple IC bipolar by factors of 10 and 100, respectively. At the same time input currents are smaller by about 103, and offset voltages and drifts are comparable to or better than slew enhanced bipolar circuits.



waveform v<sub>ON</sub>(t), which exhibits a step slew "enhancement" on the positive going output, and a slew "degradation" on the negative going output. For a p-n-p input stage, these effects are reversed as shown by v<sub>op</sub>(t).

# 5.0 SECOND-ORDER EFFECTS: VOLTAGE FOLLOWER **SLEW BEHAVIOR**

If the op amp is operated in the noninverting mode and driven by a large fast rising input, the ouput exhibits the characteristic waveform in *Figure 16*. As shown, this waveform does not have the simple symmetrical slew characteristic of the inverter. In one direction, the output has a fast step (slew "enhancement") followed by a "normal" inverter slewing response. In the other direction, it suffers a slew "degradation" or reduced slope when compared with the inverter slewing response.

We will first study slew degradation in the voltage follower connection, since this represents a worst case slewing condition for the op amp. A model which adequately represents the follower under large-signal conditions can be obtained from that in *Figure 12* by simply tying the output to the inverting input, and including a capacitor C<sub>s</sub> to account for the presence of any capacitance at the output of the first stage (tail) current source, see Figure 17. This "input tail" capacitance is important in the voltage follower because the input stage undergoes rapid large-signal excursions in this connection, and the charging currents in C<sub>s</sub> can be quite large. Circuit behavior can be understood by analyzing Figure 17

as follows. The large-signal input step causes  $Q_1$  to turn OFF, leaving  $Q_2$  to operate as an emitter follower with its emitter tracking the variational output voltage,  $v_0(t)$ . It is seen that  $v_0(t)$  is essentially the voltage appearing across both  $C_s$  and  $C_c$  so we can write

$$
\frac{dv_0}{dt} \cong \frac{i_c}{C_c} \cong \frac{i_s}{C_s}.
$$
 (35)

Noting that  $i_c \approx 2i_1 - i_s$  (unity  $\alpha$ 's assumed), (35) can be solved for is:

$$
i_{\rm s} \simeq \frac{2l_1}{1 + C_{\rm c}/C_{\rm s}}\tag{36}
$$

which is seen to be constant with time. The degraded voltage follower slew rate is then obtained by substituting (36) into (35):

$$
\left.\frac{\mathsf{v}_0}{\mathsf{dt}}\right|_{\mathsf{deq}\mathsf{c}} \cong \frac{\mathsf{i}_s}{\mathsf{C}_s} \cong \frac{2\mathsf{i}_1}{\mathsf{C}_c + \mathsf{C}_s}.\tag{37}
$$

Comparing (37) with the slew rate for the inverter, (18), it is seen that the slew rate is reduced by the simple factor  $1/(1 + C_s/C_c)$ . As long as the input tail capacitance C<sub>s</sub> is small compared with the compensation amplifiers where  $C_c$ is small, degradation become quite noticeable, and one is encouraged to develop circuits with small C<sub>s</sub>.

As an example, consider the relatively fast LM118 which has  $C_c \cong 5$  pF,  $C_8 \cong 2$  pF,  $2I_1 = 500$   $\mu$ A. The calcualted inverter slew rate is  $2I_1/C_c \approx 100V/\mu s$ , and the degraded voltage follower slew rate is found to be  $2I_1/(C_c + C_8) \approx$  $70V/\mu s$ . The slew degradation is seen to be about 30 percent, which is very significant. By contrast a  $\mu$ A741 has C<sub>c</sub>  $\approx$  30 pF and C<sub>8</sub>  $\approx$  4 pF which results in a degradation of less than 12 percent.

The slew "enhanced waveform can be similarly predicted from a simplified model. By reversing the polarity of the input and initally assuming a finite slope on the input step, the enhanced follower is analyzed, as shown in *Figure 18*. Noting that Q<sub>1</sub> is assumed to be turned ON by the step input and Q<sub>2</sub> is OFF, the output voltage becomes

$$
v_0(t) \simeq -\frac{1}{C_C} \int_0^t [2I_1 + i_S(t)] dt.
$$
 (38)

The voltage at the emitter of  $Q_1$  is essentially the same as the input voltage, v<sub>i</sub>(t), so the current in the "tail" capacitance C<sub>8</sub> is

$$
i_{s}(t) \simeq C_{8} \frac{dv_{i}}{dt} \simeq \frac{C_{8}V_{ip}}{t_{1}} \qquad 0 < t < t_{1}.
$$
 (39)

Combining  $(38)$  and  $(39)$ ,  $v_0(t)$  is

$$
-v_0(t) \simeq \frac{1}{C_c} \int_0^t 2l_1 dt + \frac{1}{C_c} \int_0^{t_1} \frac{C_8 V_{ip}}{t_1} dt \qquad (40)
$$



TL/H/8745-20

FIGURE 17. Circuit used for calculation of slew "degradation" in the voltage follower. The degradation is caused by the capacitor C<sub>8</sub>, which robs current from the tail, 2l<sub>1</sub>, thereby preventing the full 2l<sub>1</sub> from slewing C<sub>c</sub>.



TL/H/8745-21

FIGURE 18. Circuit used for calculation of slew "enhancement" in the voltage follower. The fast falling input casues a step output followed by a normal slew response as shown.

or

$$
-v_0(t) \approx \frac{C_8}{C_c} V_{ip} + \frac{2l_1 t}{C_c}.
$$
 (41)

Equation (41) tells us that the output has an initial negative step which is the fraction  $C_B/C_C$  of the input voltage. This is followed by a normal slewing response, in which the slew rate is identical to that of the inverter, see (18). This response is illustrated in Figure 18.

# **6. LIMITATIONS ON BANDWIDTH**

In earilier sections, all bandlimiting effects were ignored except that of the compensation capacitor, C<sub>c</sub>. The unity-gain frequency was set at a point sufficiently low so that negligible excess phase (over the 90° from the dominant pole) due to second-order (high frequency) poles had built up. In this section the major second-order poles which contribute to bandlimiting in the op amp are identified.

## A. The Input Stage: p-n-p's, the Mirror Pole, and the Tail Pole

For many years it was popular to identify the lateral p-n-p's (which have  $f_1$ 's  $\approx$  3 MHz) as the single dominant source of bandlimiting in the IC op amp. It is quite true that the p-n-p's do contribute significant excess phase to the amplifier, but it is not true that they are the sole contributor to excess phase [9]. In the input stage, alone, there is at least one other important pole, as illustrated in Figure 19(a). For the simple differential input stage driving a differential-to-single ended converter ("mirror" circuit), it is seen that the inverting signal (which is the feedback signal) follows two paths, one of which passes through the capacitance C<sub>8</sub>, and the other through C<sub>m</sub>. These capacitances combine with the dynamic resistances at their nodes to form poles designated the mirror pole at

$$
p_m \simeq \frac{I_1}{C_m kT/q},\tag{42}
$$

and the tail pole at

$$
p_t \approx \frac{2l_1}{C_8 kT/q} \tag{43}
$$

It can be seen that if one attempts to operate the first stage at too low a current, these poles will bandlimit the amplifier. If, for example, we choose  $I_1 = 1 \mu A$ , and assume  $C_m \cong$ 7 pF (consisting of 4 pF isolation capacitance and 3 pF emitter transition capacitance) and C<sub>8</sub>  $\approx$  4 pF,<sup>3</sup> p<sub>m</sub>/2 $\pi$   $\approx$ 0.9 MHz and  $p_t/2\pi \approx 3$  MHz either of which would seriously degrade the phase margin of a 1 MHz amplifier.

If a design is chosen in which either the tail pole or the mirror pole is absent (or unimportant), the remaining pole rolls off only half the signal, so the overall response contains a pole-zero pair separated by one octave. Such a pair generally has a small effect on amplifier response unless it occurs near  $\omega_{\rm U}$ , where it can degrade phase margin by as much as 20°.

It is interesting to note that the compound input stage of the classical LM101 and  $\mu$ A741) has a distinct advantage over the simple differential stage, as seen in Figure 19(b). This circuit is noninverting across each half, thus it provides a path in which half the feedback signal bypasses both the mirror and tail poles.

# **B. The Second Stage: Pole Splitting**

The assumption was made in Section 3 that the second stage behaved as an ideal integrator having a single dominant pole response. In practice, one must take care in designing the second stage or second-order poles can cause significant deviation from the expected response. Considerable insight into the basic way in which the second stage operates can be obtained by performing a small-signal analysis on a simplified version of the circuit as shown in Figure 20 [10]. A straightforward two-node analysis of *Figure 20(c)* produces the following expression for vout.

$$
\frac{v_{out}}{i_s} = -g_m R_1 R_2 (1 - sC_p/g_m) \div \n(1 + s[R_1 (C_1 + C_p) + R_2 (C_2 + C_p) + g_m R_1 R_2 C_p] \n+ s^2 R_1 R_2 [C_1 C_2 + C_p (C_1 + C_2)]
$$
\n(44)

<sup>3</sup> C<sub>8</sub> can have a wide range of values depending on circuit configuration. It is largest for n-p-n input differential amps since the current source has a collector-substrate capacitance ( $C_8 \cong 3-4$  pF at its output. For p-n-p input stages it can be as small as 1-2 pF.



FIGURE 19. (a) Circuit showing "mirror" pole due to  $C_m$  and "tail" pole due to  $C_8$ . One component of the signal due to an inverting input must pass through either the mirror or tail poles. (b) Alternate circuit to Figure 19(a) (LM101, µA741) which has less excess phase. Reason is that half the inverting signal path need not pass through the mirror pole or the tail pole.

The denominator of (44) can be approximately factored under conditions that its two poles are widely separated. Fortunately, the poles are, in fact, widely separated under most normal operating conditions. Therefore, one can assume that the denominator of (44) has the form

$$
D(s) = (1 + s/p1)(1 + s/p2)
$$
  
= 1 + s(1/p<sub>1</sub> + 1/p<sub>2</sub>) + s<sup>2</sup>/p<sub>1</sub>p<sub>2</sub>. (45)

With the assumption that  $p_1$  is the dominant pole and  $p_2$  is nondominant, i.e.,  $p_1 \ll p_2$ , (45) becomes

$$
D(s) \approx 1 + s/p_1 + s^2/p_1p_2. \tag{46}
$$

Equating coefficients of s in (44) and (46), the dominant pole  $p_1$  is found directly:

$$
p_1 \cong \frac{1}{R_1(C_1 + C_p) + R_2(C_2 + C_p) + g_m R_1 R_2 C_p} (47)
$$

$$
\frac{1}{g_m R_1 R_2 C_p} \tag{48}
$$

The latter approximation (48), normally introduces little error, because the g<sub>m</sub> term is much larger than the other two. We note at this point that  $p_1$ , which represents the dominant pole of the amplifier, is due simply to the familiar Miller-multiplied feedback capacitance g<sub>m</sub>R<sub>2</sub>C<sub>p</sub> combined with input node resistance,  $R_1$ . The nondominant pole  $p_2$  is found similarly by equating  $s^2$  coefficients in (44) and (46) to get  $p_1p_2$ , and dividing by  $p_1$  from (48). The result is

 $\sim$ 

$$
p_2 \simeq \frac{g_m C_p}{C_1 C_2 + C_p (C_1 + C_2)}.
$$
 (49)

Several interesting things can be seen in examining (48) and (49). First, we note that  $p_1$  is inversely proportional to  $g_m$  (and  $C_p$ ), while  $p_2$  is directly dependent on  $g_m$  (and  $C_p$ ).



TL/H/8745-24

FIGURE 20. Simplification of second stage used for pole-splitting analysis. (a) Complete second stage with input stage and output stage loading represented by R<sub>8</sub>, C<sub>8</sub>, and R<sub>L</sub>, C<sub>L</sub> respectively. (b) Emitter follower ignored to simplify analysis. (c) Hybrid  $\pi$  model substituted for transistor in (b). Source and load impedances are absorbed into model with the total impedances represented by  $R_1, C_1$ , and  $R_2$  and  $C_2$ . Transistor base resistance is ignored and  $C_p$  includes both  $C_c$  and transistor collector-base capacitance.

言<br>(c)



TL/H/8745-25





FIGURE 22. Example of pole-splitting compensation in the  $\mu$ A741 op amp. Values used in (48) and (49) are:  $g_{m2} = 1/87 \Omega$ ,  $C_p = 30$  pF,  $C_1 \cong C_2 = 10$  pF,  $R_1 = 1.7$  M $\Omega$ ,  $R_2 = 100$  k $\Omega$ .

Thus, as either  $C_p$  or transistor gain are increased, the dominant pole decreases and the nondominant pole increases. The poles  $p_1$  and  $p_2$  are being "split-apart" by the increased coupling action in a kind of inverse root locus plot.

This pole-splitting action is shown in Figure 21, where pole migration is plotted for C<sub>p</sub> increasing from 0 to a large value. Figure 22 further illustrates the action by giving specific pole positions for the  $\mu$ A741 op amp. It is seen that the initial poles (for  $C_p = 0$ ) are both in the tens of kHz region and these are predicted to reach 2.5 Hz ( $p_1/2\pi$ ) and 66 MHz  $(p_2/2\pi)$  after compensation is applied. This result is, of course, highly satisfactory since the second stage now has a single dominant pole effective over a wide frequency band.

# C. Failure of Pole Splitting

There are several situations in which the application of polesplitting compensation may not result in a single dominant pole response. One common case occurs in very wide-band op amps where the pole-splitting capacitor is small. In this situation the nondominant pole given by (49) may not become broadbanded sufficiently so that it can be ignored. To

illustrate, suppose we attempt to minimize power dissipation by running the second stage of an LM118 (which has a small-signal bandwidth of 16 MHz) at 0.1 mA. For this op amp C<sub>p</sub> = 5 pF, C<sub>1</sub>  $\cong$  C<sub>2</sub>  $\cong$  10 pF. From (49), the nondominant pole is

$$
\frac{p_2}{2\pi} \approx 16 \text{ MHz} \tag{50}
$$

which lies right at the unity-gain frequency. This pole alone would degrade phase margin by 45°, so it is clear that we need to bias the second stage with a collector current greater than 0.1 mA to obtain adequate g<sub>m</sub>. Insufficient pole-splitting can therefore occur; but the cure is usually a simple increase in second stage g<sub>m</sub>.

A second type of pole-splitting failure can occur, and it is ofen much more difficult to cope with. If, for example, one gets over-zealous in his attempt to broadband the nondominant pole, he soon discovers that other poles exist within the second stage which can cause difficulties. Consider a more exact equivalent circuit for the second stage of Figure 20(a) as shown in Figure 23. If the follower is biased at low currents or if  $c_p$ ,  $Q_2$   $g_m$ , and/or  $r_x$  are high, the circuit can contain at least four important poles rather than the two



 $\pi$  model for the emitter follower (R $\pi_1$ , C $\pi_1$ , g<sub>m1</sub>) and a complete  $\pi$  for Q<sub>2</sub> (rx<sub>2</sub>, R $\pi_2$ , etc.).



FIGURE 24. Root locus for second stage illustrating failure of pole splitting due to high g<sub>m2</sub>, r<sub>x2</sub>, C<sub>p</sub>, and/or low bias current in the emitter follower.

considered in simple pole splitting. Under these conditions, we no longer have a response with just negative real poles as in Figure 21, but observe a root locus of the sort shown in Figure 24. It is seen in this case that the circuit contains a pair of complex, possibly underdamped poles which, of course, can cause peaking or even oscillation. This effect occurs so commonly in the development of wide-band polesplit amplifiers that it has been (not fondly) dubbed "the second stage bump."

There are numerous ways to eliminate the "bump," but no single cure has been found which is effective in all situations. A direct hand analysis of Figure 23 is possible, but the results are difficult to interpret. Computer analysis seems the best approach for this level of complexity, and numerous specific analyses have been made. The following is a list of circuit modifications that have been found effective in reducing the bump in various studies: 1) reduce  $g_{m2}$ ,  $r_{x2}$ ,  $C_{\mu 2}$ , 2) add capacitance or a series RC network from the stage input to ground-this reduces the high frequency local feedback due to  $C_p$ , 3) pad capacitance at the output for similar reasons, 4) increase operating current of the follower, 5) reduce  $C_p$ , 6) use a higher  $f_t$  process.

# D. Troubles in the Output Stage

Of all the circuitry in the modern IC op amp, the class-AB output stage probably remains the most troublesome. None of the stages in use today behave as well as one might desire when stressed under worst case conditions. To illustrate, one of the most commonly used output stages is shown in *Figure 2(b)*. The p-n-p's in this circuit are "substrate" p-n-p's having low current f<sub>t</sub>'s of around 20 MHz. Unfortunately, both  $\beta_0$  and  $f_t$  begin to fall off rapidly at quite low current densities, so as one begins to sink just a few milliamps in the circuit, phase margin troubles can develop. The worst effect occurs when the amplifier is operated with a large capacitive load (>100 pF) while sinking high currents. As shown in *Figure 25*, the load capacitance on the







FIGURE 26. The "BI-FET™" output stage employing JFET's and bipolar n-p-n's to eliminate sensitivity to load capacitance.

output follower causes it to have negative input conductance, while the driver follower can have an inductive output impedance. These elements combine with the capacitance at the interstage to generate the equivalent of a one-port oscillator. In a carefully designed circuit, oscillation is suppressed, but peaking (the "output bump") can occur in most amplifiers under appropriate conditions.

One new type of output circuit which does not use p-n-p's is shown in *Figure 26* [6]. This circuit employs compatible JFET's (or MOSFET's, see similar circuit in [11]) in a FET/ bipolar quasi-complementary output stage, which is insensitive to load capacitance. Unfortunately, this circuit is rather complex and employs extra process steps, so it does not appear to represent the cure for the very low cost op amps.

## 7. The Gain Cell: Linear Large-Scale Integration

As the true limitations of the basic op amp are more fully understood, this knowledge can be applied to the development of more "optimum" amplifiers. There are, of course, many ways in which one might choose to optimize the device. We might, for example, attempt to maximize speed (bandwidth, slew rate, settling time) without sacrificing dc characteristics. The compatible JFET/bipolar amp of Figure 15 represents such an effort. An alternate choice might be to design an amplifier having all of the performance features of the most widely used general purpose op amps (i.e., µA741, LM107, etc.), but having minimum possible die area. Such a pursuit is parallel to the efforts of digital large-scale integration (LSI) designers in their devlelopment of minimum



TI /H/8745-31

FIGURE 27. Basic g<sub>m</sub> reduction obtained by using split collector p-n-p's. C<sub>c</sub> and area are reduced since C<sub>c</sub> = g<sub>m1</sub>/ $\omega_{\rm m}$ .

area memory cells or gates. The object of such efforts, of course, is to develop lower cost devices which allow wide and highly economic usage.

In this section we briefly discuss certain aspects of the linear gain cell, a general purpose, internally compensated op amp having a die area which is significantly smaller than that of equivalent, present day, industry standard amplifiers.

## A. Transconductance Reduction

The single largest area component in the internally compensated op amp is the compensation capacitor (about 30 pF, typically). A major interest in reducing amplifier die area, therefore, centers about finding ways in which this capacitor can be reduced in size. With this in mind, we find it useful to examine (15), which relates compensation capacitor size to two other parameters, unity gain corner frequency  $\omega_{\rm u}$ , and first stage transconductance  $g_{m1}$ . It is immediately apparent

that for a fixed, predetermined unity gain corner (about  $2\pi$  $\times$  1 MHz in our case), there is only one change that can be made to reduce the size of C<sub>c</sub>: the transconductance of the first stage must be reduced. If we restrict our interest to simple bipolar input stages (for low cost), we recall the  $q_{m1}$  $=$  ql<sub>1</sub>/kT. Only by reducing  $I_1$  can g<sub>m1</sub> be reduced, and we earlier found in Section 6-A and Figure 19(a) and (b) that I<sub>1</sub> cannot be reduced much without causing phase margin difficulties due to the mirror pole and the tail pole.

An alternate basic approach to  $g_m$  reduction is illustrated in Figure 27 [12]. there, a multiple collector p-n-p structure, which is easily fabricated in IC form, is used to split the collector current into two components, one component (the larger) of which is simply tied to ground, thereby "throwing away" a major portion of the transistor output current. The result is that the  $g_m$  of the transistor is reduced by the ratio of  $1/(1 + n)$  (see *Figure 27*), and the compensation capaci-



 $(a)$ 



FIGURE 28. Variations on g<sub>m</sub> reduction. (a) Cross-coupled connection eliminates all ac current passing through the mirror, yet maintains dc balance. (b) This approach maintains high current on the diode side of the mirror, thereby broadbanding the mirror pole.

tance can be reduced directly by the same factor. It might appear that the mirror pole would still cause difficulties since the current mirror becomes current starved in Figure 27, but the effect is not as severe as might be expected. The reason is that the inverting signal can now pass through the high current wide-band path, across the differential amp emitters and into the second stage, so at least half the signal current does not become bandlimited. This partial bandlimiting can be further reduced by using one of the circuits in Figure 28(a) or (b).<sup>4</sup> In (a), the p-n-p collectors are cross coupled in such a way that the ac signal is cancelled in the mirror circuit, while dc remains completely balanced. Thus the mirror pole is virtually eliminated. The circuit does have a drawback, however, in that the uncorrelated noise currents coming from the two p-n-p's add rather than subtract at the input to the mirror, thereby degrading noise performance. The circuit in *Figure 28(b)* does not have this defect. but requires care in matching p-n-p collector ratios to n-p-n emitter areas. Otherwise offset and drift will degrade as one attempts to reduce q<sub>m</sub> by large factors.

# **B. A Gain Cell Example**

As one tries to make large reductions in die area for the gain cell, many factors must be considered in addition to novel circuit approaches. Of great importance are special layout/ circuit techniques which combine a maximum number of components into minimum area.

In a good layout, for example, all resistors are combined into islands with transistors. If this is not possible initially, circuit and device changes are made to allow it. The resulting device geometrics within the islands are further modified in shape to allow maximum "packing" of the islands. That is, when the lavout is complete, the islands should have shapes which fit together as in a picture puzzle, with no waste of space. Further area reductions can be had by modifving the isolation process to one having minimum spacing between the isolation diffusion and adiacent p-regions.

As example of a gain cell which employs both circuit and layout optimization is shown in *Figure 29*: This circuit uses the g<sub>m</sub> reduction technique of *Figure 28(a)* which results in a compensation capacitor size of only 5 pF rather than the normal 30 pF. The device achieves a full 1 MHz bandwidth. a 0.67V/us slew rate, a gain greater than 100,000, typical offset voltages less than 1 mV, and other characteristics normally associated with an LM107 or uA741. In quad form each amplifier requires an area of only 23 x 35 mils which is one-fourth the size of today's industry standard uA741 (typically 56 x 56 mils). This allows over 8000 possible gain cells to be fabricated on a single 3-inch wafer. Further, it appears quite feasible to fabricate larger arrays of gain cells, with six or eight on a single chip. Only packaging and applications questions need be resolved before pursuing such a step. <sup>4</sup>The circuit in *Figure 28(a)* is due to R. W. Russell and the variation in *Figure* 

28(b) was developed by D. W. Zobel.



# **ACKNOWLEDGMENT**

Many important contributions were made in the gain cell and FET/bipolar op amp areas by R. W. Russell. The author gratefully acknowledges his very competent efforts.

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