How to Design High Order Filters with Stopband Notches Using the LTC1562 Operational Filter (Part 2) by Nello Sevastopou by Nello Sevastopoulos

This is the second in a series of articles describing applications of the LTC1562 connected as a lowpass, highpass or bandpass filter with added stopband notches to increase selectivity. Part 1 (Linear Technology VIII:2, May 1998, pp. 28-31) described one method of coupling the four Operational Filter[™] building blocks of the LTC1562 to design an 8th order lowpass filter with two stopband notches. Part 2 expands the technique of Part 1 to design an 8th order bandpass filter with two stopband notches.

Throughout this series of articles, notches will be generated by first summing the input signal with a 180 degree out-of-phase signal appearing at the output(s) of the LTC1562 Operational Filter and second, by adjusting the summation gains to yield a zero sum.

Part 1 showed one proprietary method of creating notches in the stopband of a lowpass filter. The essence of this method is briefly revisited in Figure 1, where two of four Operational Filter sections are coupled to form a 4th order lowpass filter with one stopband notch. The notch is obtained by summing the input signal, V_{IN} , with the output, V1A, into the inverting node of the next section of the IC. The two signals, V_{IN} and V1A, will tend to cancel each other at a frequency where they are 180 degrees out of phase. The cancellation will be complete if the amplitudes of V_{IN} and V_{IA} yield equal (and opposite) currents at the summing junction of the op amp of Figure 1. that is if:

$$R_{IN2} = R_{FF2} \cdot (R_{Q1}/R_{IN1})$$
 (1)

In Figure 1, the lead capacitor C_{IN1} raises the frequency where a 180 degree phase shift occurs above the center frequency of the 2nd order section (f₀). The resulting notch frequency is then higher than the cutoff frequency of the 4th order filter.

Figure 1 can be easily modified to make the frequency of the notch lower than the center frequency of the 2nd

order section from which it is derived. This is useful in bandpass filters where an unwanted frequency lower than the center frequency of the filter must be rejected. This is shown in Figure 2, where the input signal is summed with output V2A instead of output V1A. The frequency of the resulting notch is:

$$f_{N2} = f_{O1} \cdot \sqrt{1 - \frac{R1}{R_{O1}} \cdot \frac{C}{C_{IN1}} \cdot \frac{R21}{R_{IN1}}}$$
 (2)

(R1 = 10k; C = 159.15pF)

and the gain conditions dictating Equation 1 now translate to:

$$R_{IN2} = R_{FF2} \cdot \left(\frac{R_{Q1}}{R_1} \cdot \frac{C_{IN1}}{C}\right)$$
(3)

The circuit of Figure 2 can be used to build a 4th order bandpass filter with one notch below its center frequency. Such a filter can simultaneously detect a tone and reject an unwanted frequency located in the vicinity of the passband.



Figure 1. Two out of four Operational Filter sections are coupled to form a 4th order lowpass filter with one stopband notch.



Figure 2. Figure 1's circuit modified to make the frequency of the notch lower than the center frequency of the 2nd order section from which it is derived.



Figure 3. Theoretical amplitude response of 8th order, 100kHz bandpass filter

The notch techniques of Figures 1 and 2 will be referred as "feedforward." This is necessary to separate these techniques from others to be shown later, in Part 3 of this series of articles.

The feedforward notch technique of Figure 2 can be advantageously combined with Figure 1 to realize sharp bandpass filters with two stopband notches: one notch below and one above the center frequency. Filters of this type can be very selective, although they are quite cumbersome to design. A step-by-step design procedure is illustrated below.

A Practical Example

An 8th order 100kHz bandpass filter is realized, through FilterCAD[™] for Windows[®] (available at no charge from Linear Technology—see the "Design Tools" page in this issue), by cascading four 2nd order sections of equal Q. The -3dB band-edges are arithmetrically symmetric with respect to the filter's 100kHz center frequency and signals below 80kHz and above 125kHz are attenuated by 60dB or more. Figure 3 shows the theoretical amplitude response and Table 1 shows the desired filter parameters, namely, the center frequencies, Qs and notch frequencies. The filter of Figure 3/Table 1 can be realized by decomposing the 8th order realization into two independent 4th order filter sections and then cascading these two 4th order sections, which is an easier task than designing an 8th order elliptic bandpass filter all at once. FilterCAD, in custom mode,

Table 1. Parameters of the four sections of an 8th order, 100kHz bandpass filter					
f ₀	Q	f _N	Q _N	Туре	
99.9687e3	10.0000			BP	
96.9964e3	10.0000	129.2814e3		LPN	
103.0322e3	10.0000	77.3023e3		HPN	
100.0000e3	10.0000			BP	

should be used to perform this operation. Figure 4 and Table 2 show the filter decomposition and the cascading sequence; note the left and right notches. Figure 5 uses the LTC1562 Operational Filter to realize the filter of Figure 3 as decomposed in Figure 4. The design is split into two 4th order sections. The algorithm to calculate the external passive components is outlined below.

In order to obtain a practical realization that closely approximates the theoretical one, the Q of each 2nd order section will be lowered by 15%. (Please consult the LTC1562 final data sheet.)

In order to follow the long and tedious algorithm below, consider the intuitive outline: We need to calculate the following set of passive components for the first 4th order section: R_{IN1} , C_{IN1} , R21, R_{Q1} , and R_{IN2} , R_{FF2} , R22 and R_{Q2} . The resistors R21, R_{Q1} ,

R22 and R_{Q2} are easily calculated via the expression for the center frequency, f_{Oi}, and Q_i for the 2nd order section "i." The expression for the notch, equation (2), involves the product of $R_{IN1} \bullet C_{IN1}$, so neither component can be calculated separately. Instead, R_{IN1} is calculated by considering the maximum gain (which occurs around the center frequency f_{01} at either node V1A or V2A. This controls premature internal clipping. Once R_{IN1} is set, C_{IN1} is easily calculated via equation (2) for the lower band notch. Similarly, equation (3) defines the ratio of R_{IN2} to R_{FF2} , so neither of these components can be calculated independently of the other. R_{FF2} is calculated by considering the gain factor ("GAIN") of the 4th order filter section at the V1B output (Figure 1/ Table 2)). Once R_{FF2} is set, R_{IN2} is calculated via equation (3).





Table 2. Filter decomposition and cascading sequence				
f ₀₁ = 96.9964k Q1 = 10	f ₀₃ = 100k Q3 = 10			
$f_{02} = 99.9687k$ Q2 = 10 $f_{N2} = 77.3k$	$f_{O4} = 103.0322k$ Q4 = 10 $f_{N2} = 129.2814k$			
$H(s) = GAIN \cdot N(s)/D(s)$	$H(s) = GAIN \cdot N(s)/D(s)$			
GAIN = 0.2823	GAIN = 0.1788			
$N(s) = A1s(s^2 + 235 \cdot 9072 \cdot 10^9)$	$N(s) = A1s(s^2 + 659 \cdot 83 \cdot 10^9)$			
$A1 = 62.8122 \cdot 10^3$	$A1 = 62.8319 \cdot 10^3$			

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The same design method is later repeated to derive the passive components for the second 4th order section:

I. Calculate the passive components of the of the first 4th order section

 $(f_{O1} = 96.9964 \text{kHz}, Q = 8.5, f_{O2} = 99.9687 \text{kHz}, Q = 8.5, f_{n2} = 77.3 \text{kHz})$

- Calculate the center frequencysetting resistor, R21: (For details, please refer to the LTC1562 data sheet.) R21 = (100kHz/f₀₁)² • 10k = 10.629k (choose the closest 1% value, R21 = 10.7k (1%))
- 2. Calculate the Q-setting resistor, R_{Q1} : (For details, please refer to the

LTC1562 data sheet) $R_{Q1} = Q1 \sqrt{R21 \cdot 10k} = 87.925k$ (choose the closest 1% value, $R_{Q1} = 86.6k$ (1%))

3. Calculate the input resistor R_{IN1} from the following expression(s): 3a. if $f_{O1} \leq 100$ kHz (for LTC1562)

$$R_{IN1} = Q1 \cdot R21 \cdot \sqrt{1 + \frac{1}{Q1^2 \cdot \left(1 - \frac{f_{N2}^2}{f_{O1}^2}\right)^2}}$$
(4)

R_{IN1} = 95.56k

Although not applicable for this example, thoroughness dictates mentioning the case below:

3b. if $f_{01} \ge 100$ kHz (for LTC1562)

$$R_{IN1} = R_{Q1} \cdot \sqrt{\frac{1 + \frac{1}{Q1^2 \cdot \left(1 - \frac{f_{N2}^2}{f_{O1}^2}\right)^2}}$$
 (5)

Make sure, in either case 3a or 3b, that R_{IN1} is greater than R21, that is, the DC gain at pin 3 in Figure 5 is less than unity; if not set $R_{IN1} = R21$ and proceed to step 4a.

The expression for R_{IN1} sets the gain at f_{O1} equal to unity at the node of maximum swing (V1A or V2A). Note that, for high Qs, the gain at f_{O1} is the maximum gain. If you know the spectrum of the signals that will be applied to the filter input and if internal gains higher than unity will be allowed, the value of R_{IN1} can be reduced to improve the input signal-to-noise ratio.

4a. Use the value of R_{IN1} , calculated above, and calculate the value for the input capacitor C_{IN1} from the notch equation (2).

$$C_{IN1} = \frac{R1}{R_{Q1}} \cdot \frac{R21}{R_{IN1}} \cdot C \left[\frac{1}{1 - \left(\frac{f_{N2}^2}{f_{O1}^2} \right)} \right]$$
 (6)

$$(f_{N1} < f_{O1}; C = 159.15 pF)$$

$$C_{IN1} = 5.639 pI$$

Use the commercially available NPO type 0402 surface mount capacitor with the value nearest the ideal value of $C_{\rm IN1}$ calculated above. For instance, for $C_{\rm IN1}$, choose an off-the-shelf 5.6pF capacitor.





Figure 6. Measured amplitude response of Figure 5's filter

- 4b. Recalculate the value of R_{IN1} after C_{IN1} is chosen. $R_{IN1} = (C_{IN1(ideal)} R_{IN1(ideal)})/$ $C_{IN1(NPO,0402)} = 96.22k$ Choose the closest 1% value: $R_{IN1} = 95.3k (1\%)$
- 5. Calculate the frequency- and Qsetting resistors R22, R_{Q2} , as done in steps 1 and 2, above. Choose the closest 1% standard resistor values. R22 = 10k (1%):

$$R_{Q1} = 84.5k(1\%)$$

6. Calculate the feedforward resistor, R_{FF2} : $1/(R_{FF2} C) = Gain \cdot A1;$ C = 159.15 pF

The values for parameter (Gain • A1) are provided by FilterCAD; they relate to the coefficients of the numerator of the transfer function (V1B/ V_{IN} in Figure 1); a passband AC gain of unity is assumed (see Table 2). Please note that, for a lowpass case, as in Part 1 of this article series, the value of (Gain • A1) is the DC gain of the filter and its value can be easily set without software assistance.

Equating the numerator of the filter transfer function with the values provided by FilterCAD:

$$\frac{V_{1B}}{V_{IN}} = \frac{s(s^2 + \omega_{N2}^2)}{(R_{FF2} \cdot C) \cdot D(s)} = \frac{GAIN (A1s)(s^2 + A2)}{D(s)}$$
(7)
GAIN = 0.2823

A1 =
$$62.8122 \cdot 10^{3}$$

A2 = $(2\pi f_{N2})^{2} = 235.9 \cdot 10^{9}$
R_{FF2} = $1/((Gain A1) C) = 354.35k;$
C = $159.15pF$
R_{FF2} = $357k(1\%)$

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Figure 7. Gain linearity of Figure 5's filter, measured at the 100kHz theoretical center frequency

- 7. Solve for R_{IN2} by using Equation (3), which dictates the gain condition for the occurrence of the notch:
- $\begin{array}{l} R_{IN2} = (R_{FF2} \ R_{Q1} \ C_{IN1}) / (R1 \ C) = \\ 108.785 k; \ (R1,C) = (10k, \ 159.15 pF) \\ R_{IN2} = 110 k \ (1\%) \end{array}$
- II. Calculate the passive components of the second 4th order section $(f_{O3} = 100$ kHz, Q3 = 8.5, $f_{O4} =$ 103.0322kHz, Q4 = 8.5, $f_{n4} =$ 129.2814kHz) Except for the bandpass gain calculations, the algorithm will be the same as the lowpass design of Part 1 of this article.
- 1. R23 = $(100 \text{ kHz}/f_{03})^2 \cdot 10 \text{ k} = 10 \text{ k} (1\%)$
- 2. $R_{Q3} = Q3 \sqrt{R23 \cdot 10k} = 85k$, $R_{Q3} = 84.5k (1\%)$
- 3. Calculate the input resistor R_{IN3} from the following expression(s): 3a. if $f_{O3} \le 100$ kHz (for LTC1562)

$$R_{IN3} = Q3 \cdot R23 \cdot \sqrt{1 + \left(1 - \frac{f_{O3}^2}{f_{N4}^2}\right)^2 \cdot Q3^2}$$
(8)

 $R_{IN3} = 302.41k$

3b. if
$$f_{O3} \ge 100$$
kHz (for LTC1562)

$$R_{IN3} = R_{Q3} \cdot \sqrt{1 + \left(1 - \frac{f_{O3}^2}{f_{N4}^2}\right)^2 \cdot Q3^2}$$
 (9)

For $f_{\rm O3}$ = 100kHz, as in the example above, either expression can be used. Note that the expression for $R_{\rm IN3}$ in 3b, above, is the same as expression for $R_{\rm IN1}$ shown in Part 1 of this article.

4a. Use the theoretical value for R_{IN3} , calculated above, and calculate the value of the input capacitor C_{IN3} from the notch equation (2) of part 1 of this article; for convenience this is repeated below:

$$C_{IN3} = C \cdot \frac{R_{O3}}{R_{IN3}} \cdot \left(1 - \frac{f_{O3}^2}{f_{N4}^2}\right)$$
 (10)

 $C_{IN3} = 17.86 pF;$

Use a commercially available NPOtype 0402 surface mount capacitor with the value nearest the ideal value of C_{IN3} calculated above. For instance, $C_{IN3} = 18$ pF.

- 4b. Recalculate the value for R_{IN3} calculated in step 3a after C_{IN3} is chosen.
- $R_{IN3} = (C_{IN3(ideal)} R_{IN3(ideal)})/C_{IN3(NPO,0402)} = 300.058k$
- $R_{IN3} = 294k (1\%)$
 - 5. Calculate the frequency- and Q-setting resistors, R24 and R_{Q4} , as done in steps 1 and 2, above. Choose the nearest 1% standard value.

R24 = 9.42k; R24 = 9.53k (1%) R₀₄ = 82.97k; R₀₄ = 82.5k (1%)

6. Calculate the feedforward resistor, RFF4. First equate the numerator of the 4th order filter transfer function with the values provided by FilterCAD (see Table 2):

$$\frac{V_{OUT}}{V_{1B}} = \frac{s}{R_{FF4} \cdot C} \cdot \frac{\omega_{03}^2}{\omega_{04}^2} \cdot \frac{s^2 + \omega_{N4}^2}{D(s)} = \frac{(11)^2}{D(s)}$$

$$\frac{GAIN \cdot A1s \cdot (s^2 + \omega_{N4}^2)}{D(s)}$$

THEN
$$R_{FF4} = \frac{1}{GAIN \cdot A1} \cdot \frac{1}{C} \cdot \frac{\omega_{03}}{\omega_{N4}^2}$$

GAIN = 0.1788
A1 = 62.8319 \cdot 10^3

 $R_{FF4} = 334.64k$, choose $R_{FF4} = 332k$ (1%).

7. Solve for R_{IN4} by using equation (1) of Part 1 of this article, which dictates the gain

condition for the occurrence of a notch. For convenience, this gain condition is repeated below.

$$R_{IN4} = R_{FF4} \cdot \frac{R_{O3}}{R_{IN3}}$$
(12)

 $R_{IN4} = 95.422k; R_{IN4} = 95.3k(1\%)$

Experimental Results

Figure 6 shows the measured amplitude response of the filter of Figure 5. The values of the passive component are as calculated above and as shown in Figure 5. The measured amplitude response closely approximates the ideal response as synthesized by FilterCAD. The peak frequency with standard 1% resistor values and 5% capacitor values is 100.65kHz (0.65% off). The higher frequency notch, although it shows a respectable depth of 70dB, is not as well defined as the notch below the filter's center frequency, yet the -65dB bandwidth is as predicted by FilterCAD. The 10dB lack of the upper band notch depth is due to the finite speed of the internal op amps; they cause the practical 180 degree phase shift frequency and the gain at V1A's output to depart slightly from the theoretical calculations.

For the sake of perfection, the notch depth can be easily restored by tweaking the value of R_{Q3} ; the new R_{Q3} will be 75k. This is shown with dashed lines in Figure 6. This, however, lowers the passband gain by the ratio of the new to the old R_{Q3} value, that is, by about -1.0dB (you cannot fool mother nature). Depending on the application, the 10dB of additional notch depth for 1.5dB of passband gain loss may be a reasonable trade. The passband gain can also be corrected by lowering the values of either pair, (R_{FF2}, R_{IN2}) or (R_{FF4}, R_{IN4}) , by the same amount (1.5dB). In Figure 6, the gain was restored to 0dB by changing the values of R_{IN2} , R_{FF2} to 93.1k and 300.1k respectively.

The total integrated noise was an impressively low $69\mu V_{RMS}$, allowing a signal-to-noise ratio well in excess of 80dB. The input signal-to-noise ratio can be further increased if the pass-

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band gain can be higher than 0dB or if internal nodes are allowed to have gains higher than 0dB. Please contact the LTC Filter Design and Applications Group for further details. The low noise behavior of the filter makes it useful in applications where the input signal has a wide voltage

range. This is true provided the filter magnitude response does not change with varying input signal levels, that is, the filter gain is linear. The gain linearity measured at the 100kHz theoretical center frequency of the filter is shown in Figure 7. The gain is perfectly linear for input amplitudes up to $1.25V_{RMS}$ (3.5V_{P-P}) so an 84dB dynamic range can be claimed. The input signal, however, can reach amplitudes up to $3V_{RMS}$ (8.4V_{P-P}, 92dB SNR) with some reduction in gain linearity.