$\pm 5V$ to $\pm 18V$

±1% unadjusted

National Semiconductor

AF100 Universal Active Filter

General Description

Connection Diagrams

The AF100 state variable active filter is a general second order lumped RC network. Only four external resistors program the AF100 for specific second order functions. Lowpass, highpass, and bandpass functions are available simultaneously at separate outputs. Notch and allpass functions are available by summing the outputs in the uncommitted output summing amplifier. Higher order systems are realized by cascading AF100 active filters with appropriate programming resistors.

Any of the classical filter configurations, such as Butterworth, Bessel, Cauer, and Chebyshev can be formed.

Features

- Military or commercial specifications
- Independent Q, frequency, gain adjustments
- Low sensitivity to external component variation
- Separate lowpass, highpass, bandpass outputs
- Inputs may be differential, inverting, or non-inverting
- Allpass and notch outputs may be formed using uncommitted amplifier
- Operates to 10 kHz
- Q range to 500
- Power supply range
- Frequency accuracy
- Q frequency product ≤50,000



Absolute Maximum Ratings

If Military/Aerospace specified devices are required, please contact the National Semiconductor Sales Office/Distributors for availability and specifications.

Office/Distributors for availability an	a specifications.	AF 100-10G, AF 100-20G,	
Supply Voltage	±18V	AF100-1CN, AF100-2CN	-25°C to +85°C
Power Dissipation	900 mW/Package	AF100-1G, AF100-2G	-55°C to +125°C
·	(500 mW/Amp)	Storage Temperature	
Differential Input Voltage	±36V	AF100-1G, AF100-2G	-65°C to +125°C
Output Short Circuit Duration (Note 1)	Infinite	AF100-100, AF100-200, AF100-101, AF100-201	
Lead Temperature (Soldering, 10 sec.)	300°C	AF100-1CN, AF100-2CN	-25°C to +100°C

Operating Temperature AF100-1CJ, AF100-2CJ,

Electrical Characteristics (Complete Active Filter) (Note 2)

Parameter	Conditions	Min	Тур	Max	Units
Frequency Range	$f_{C} \times Q \leq 50,000$			10k	Hz
Q Range	$f_{C} \times Q \le 50,000$			500	Hz/Hz
f ₀ Accuracy AF100-1, AF100-1C AF100-2, AF100-2C	$f_{C} \times Q \le 10,000, T_{A} = 25^{\circ}C$ $f_{C} \times Q \le 10,000, T_{A} = 25^{\circ}C$			±2.5 ±1.0	%
fo Temperature Coefficient			± 50	± 150	ppm/°C
Q Accuracy	$f_{C} \times Q \le 10,000 T_{A} = 25^{\circ}C$			±7.5	%
Q Temperature Coefficient			±300	±750	ppm/°C
Power Supply Current	$V_{\rm S} = \pm 15 V$		2.5	4.5	mA

Electrical Characteristics (Internal Op Amp) (Note 3)

Parameter	Conditions	Min	Тур	Max	Units
Input Offset Voltage	$R_S \le 10 k\Omega$		1.0	6.0	m∨
Input Offset Current			4	50	nA
Input Bias Current			30	200	nA
Input Resistance			2.5		MΩ
Large Signal Voltage Gain	$R_L \ge 2k$ $V_{OUT} = \pm 10V$	25	160		V/mV
Output Voltage Swing	$R_{L} = 10 k\Omega$ $R_{L} = 2 k\Omega$	± 12 ± 10	±14 ±13		v
Input Voltage Range		±12			v
Common Mode Rejection Ratio	$R_S \le 10 k\Omega$	70	90		dB
Supply Voltage Rejection Ratio	R _S ≤ 10 kΩ	77	96		dB
Output Short Circuit Current			25		mA
Slew Rate (Unity Gain)			0.6		V/µs
Small Signal Bandwidth			1		MHz
Phase Margin			60		Degrees

Note 1: Any of the amplifiers can be shorted to ground indefinitely, however more than one should not be simultaneously shorted as the maximum junction temperature will be exceeded.

Note 2: Specifications apply for $V_S = \pm 15V$, over $-25^{\circ}C$ to $+85^{\circ}C$ for the AF100-1C and AF100-2C and over $-55^{\circ}C$ to $+125^{\circ}C$ for the AF100-1 and AF100-2, unless otherwise specified.

Note 3: Specifications apply for V_S = $\pm 15V$, T_A = 25°C.



FIGURE 1. AF100 Schematic

CIRCUIT DESCRIPTION AND OPERATION

A schematic of the AF100 is shown in *Figure 1*. Amplifier A1 is a summing amplifier with inputs from integrator A2 to the non-inverting input and integrator A3 to the inverting input. Amplifier A4 is an uncommitted amplifier.

By adding external resistors the circuit can be used to generate the second order system

$$T(s) = \frac{a_3s^2 + a_2s + a_1}{s^2 + b_2s + b_1}$$

The denominator coefficients determine the complex pole pair location and the quality of the poles where

 $\omega_0 = \sqrt{b_1} =$ the radian center frequency ω_0

 $Q = \frac{\omega_0}{b_2} = \text{ the quality of the complex pole pair}$ If the output is taken from the output of A1, numerator coef-

ficients a_1 and a_2 equal zero, and the transfer function becomes:

$$T(s) = \frac{a_3 s^2}{s^2 + \frac{\omega_0}{O} s + \omega_0^2}$$
 (highpass)

If the output is taken from the output of A2, numerator coefficients a_1 and a_3 equal zero and the transfer function becomes:

*External components

$$T(s) = \frac{a_2 s}{s^2 + \frac{\omega_0}{Q} s + \omega_0^2}$$
 (bandpass)

If the output is taken from the output of A3, numerator coefficients a_3 and a_2 equal zero and the transfer function becomes:

$$\Gamma(s) = \frac{a_1}{s^2 + \frac{\omega_0}{\Omega}s + \omega_0^2}$$
 (lowpass)

Using proper input and output connections the circuit can also be used to generate the transfer functions for a notch and allpass filter.

In the transfer function for a notch function a_2 becomes zero, a_1 equals 1, and a_3 equals $\omega_Z{}^2.$ The transfer function becomes:

$$\Gamma(s) = \frac{s^2 + \omega_Z^2}{s^2 + \frac{\omega_0}{\Omega}s + \omega_0^2}$$
(notch)

In the allpass transfer function $a_1 = 1$, $a_2 = -\omega_0/Q$ and $a_3 = \omega_0^2$. The transfer function becomes:

$$\Gamma(s) = \frac{s^2 - \frac{\omega_0}{Q}s + \omega_0^2}{s^2 + \frac{\omega_0}{Q}s + \omega_0^2}$$
(allpass)

COMMON CONFIGURATIONS

The specific transfer functions for some of the most useful circuit configurations using the AF100 are illustrated in *Figures 2* through *B*. Also included are the gain equations for each transfer function in the frequency band of interest, the Q equation, center frequency equation and the Q determining resistor equation.



FIGURE 2. Non-Inverting Input (Q > Q_{MIN}, see Q Tuning Section)



a) Non-inverting input (Figure 2) transfer equations are:













$$HQ = \frac{Q}{\sqrt{0.1\frac{\omega_2}{\omega_1}}} \left(1.1 + \frac{10}{H_{\text{H}}} \right)$$

1-9



FIGURE 6. Output Notch Using All Four Amplifiers

e) Output notch (Figure 6) transfer function equations are:

AF100

$$\frac{e_{n}}{e_{IN}} = \frac{\left(s^{2} + \omega_{Z}^{2}\right)\left[\frac{1.1}{1 + \frac{R_{IN}}{10^{5}} + \frac{R_{IN}}{RQ}}\right]\frac{R_{g}}{R_{h}}}{s^{2} + s\omega_{1}\left[\frac{1.1}{1 + \frac{10^{5}}{RQ} + \frac{10^{5}}{RQ}}\right] + 0.1 \omega_{1}\omega_{2}}$$
$$\omega_{1} = \frac{10^{9}}{R_{F1}} \qquad \omega_{2} = \frac{10^{9}}{R_{F2}} \qquad \omega_{0} = \sqrt{0.1 \omega_{1}\omega_{2}}$$
$$\omega_{Z} = \omega_{0}\sqrt{\frac{10 R_{h}}{R_{\ell}}}$$
$$\frac{e_{n}}{e_{IN}}\Big|_{s \to 0} = \frac{11}{\left(1 + \frac{R_{IN}}{10^{5}} + \frac{R_{IN}}{RQ}\right)}\frac{R_{g}}{R_{\ell}}$$
$$\frac{e_{n}}{e_{IN}}\Big|_{s \to \infty} = \frac{1.1}{\left(1 + \frac{R_{IN}}{10^{5}} + \frac{R_{IN}}{RQ}\right)}\frac{R_{g}}{R_{h}}$$



f) Input notch (Figure 7) transfer function equations are:

$$\begin{split} \frac{e_{n}}{e_{IN}} &= \frac{\frac{C_{Z}}{10^{-9}} \left[s_{2} + \omega_{Z}^{2} \right]}{s^{2} + s \, \omega_{1} \left[\frac{1.1 \, RQ}{10^{5} + RQ} \right] + \, \omega_{0}^{2}} \\ \omega_{1} &= \frac{10^{9}}{R_{F1}} \qquad \omega_{2} = \frac{10^{9}}{R_{F2}} \\ \omega_{Z} &= \omega_{0} \sqrt{\frac{RF2 \times 10^{-9}}{R_{Z} \, C_{Z}}} \qquad \omega_{0} = \sqrt{0.1 \, \omega_{1} \omega_{2}} \\ \frac{e_{n}}{e_{IN}} \bigg|_{\omega \to 0} &= -\frac{R_{F2}}{R_{Z}} \\ \frac{e_{n}}{e_{IN}} \bigg|_{\omega \to \infty} &= -\frac{C_{Z}}{10^{-9}} \end{split}$$

g) Allpass (Figure 8) transfer function equations are:

$$\frac{e_{0}}{e_{IN}} = -\left[\frac{s^{2} - s \omega_{1} \left[\frac{1.1}{2 + \frac{R_{IN}}{RQ}}\right] + \omega_{0}^{2}}{s^{2} + s \omega_{1} \left[\frac{1.1}{2 + \frac{R_{IN}}{RQ}}\right] + \omega_{0}^{2}}\right]$$
$$= \frac{2 + \frac{10^{5}}{RQ}}{1.1} \sqrt{0.1 \frac{\omega_{2}}{\omega_{1}}}$$
$$1 = \frac{10^{9}}{R_{F1}} \qquad \omega_{2} = \frac{10^{9}}{R_{F2}}$$

$$\omega_0 = \sqrt{0.1 \omega_1 \omega_2}$$

Time delay at $\omega = \frac{2Q}{\omega_0}$ seconds

Q

ω



FREQUENCY TUNING

To tune the AF100 two resistors are required for frequencies between 200 Hz and 10 kHz. For lower frequencies "T" tuning or addition of external capacitors is required. Using external capacitors allows the user to go as low in frequency as he desires. "T" tuning and external capacitors can be used together.

Two resistor tuning for 200 Hz to 10 kHz



FIGURE 9. Resistive Tuning

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AF100

"T" resistive tuning for f_O < 200 Hz





FIGURE 10. T Tuning

TL/K/10111-14





GRAPH A. Resistive Tuning 1000 r_f-resistance (kΩ) 100 10 1 100 100k 1k 10k fc-FREQUENCY (Hz) TL/K/10111-13



Q TUNING

To tune the Q of an AF100 requires one resistor from pins 1 or 2 to ground. The value of the Q tuning resistor depends on the input connection and input resistance as well as the value of the Q. The Q of the unit is inversely proportional to resistance to ground at pin 1 and directly proportional to resistance to ground from pin 2.



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For any Q in inverting mode:



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FIGURE 14. Q Tuning Inverting Input



TL/K/10111-23

NOTCH TUNING

Two methods to generate notches are the RC input and lowpass/highpass summing. The RC input method requires adding a capacitor and resistor connected to the two integrator inputs. The capacitor connects to "Int 1" and the resistor connects to "Int 2". The output summing requires two resistors connected to the lowpass and highpass output.

For input RC notch tuning:





For output notch tuning:



FIGURE 16. Output Notch

RLP

TL/K/10111-26



Applications Information (Continued) TUNING TIPS

In applications where 2% to 3% accuracy is not sufficient to provide the required filter response, the AF100 stages can be tuned by adding trim pots or trim resistors in series or parallel with one of the frequency determining resistors and the Q determining resistor.

When tuning a filter section, no matter what output configuration is to be used in the circuit, measurements are made between the input and the bandpass (pin 13) output.

Before any tuning is attempted the lowpass (pin 7) output should be checked to see that the output is not clipping. At the center frequency of the section the lowpass output is 10 dB higher than the bandpass output and 20 dB higher than the highpass. This should be kept in mind because if clipping occurs the results obtained when tuning will be incorrect.

Frequency Tuning

By adjusting the resistance between pins 7 and 13 the center frequency of a section can be adjusted. If the input is through pin 1 the phase shift at center frequency will be 180° and if the input is through pin 2 the phase shift at center frequency will be 0°. Adjusting center frequency by phase is the most accurate but tuning for maximum gain is also correct.

"Q" Tuning

The "Q" is tuned by adjusting the resistance between pin 1 or 2 and ground. Low Q tuning resistors will be from pin 2 to ground (Q < 0.6). High Q tuning resistors will be from pin 1 to ground. To tune the Q correctly the signal source must have an output impedance very much lower than the input resistance of the filter since the input resistance affects the Q. The input must be driven through the same resistance the circuit will see to obtain precise adjustment.

The lower 3 dB (45°) frequency, f_L , and the upper 3 dB (45°) frequency, f_H , can be calculated by the following equations:

$$f_{H} = \left(\frac{1}{2Q} + \sqrt{\left(\frac{1}{2Q}\right)^{2} + 1}\right) \times (f_{O})$$

where f_{O} = center frequency

$$\mathbf{f}_{\mathsf{L}} = \left(\sqrt{\left(\frac{1}{2\mathsf{Q}}\right)^2 + 1} - \frac{1}{2\mathsf{Q}}\right) \times (\mathbf{f}_{\mathsf{O}})$$

When adjusting the Q, set the signal source to either f_H or f_L and adjust for 45° phase change or a 3 dB gain change.

Notch Tuning

If a circuit has a jw axis zero pair the notch can be tuned by adjusting the ratio of the summing resistors (lowpass/highpass summing) or the input resistance (input RC).

In either case the signal is connected to the input and the proper resistor is adjusted for a null at the output.

Special Cases

When using the input RC notch the unit cannot be tuned through the normal input so an additional 100k resistor can be added at pin 1 and the unit can be tuned normally. Then the 100k input resistor should be grounded and the notch tuned through the normal RC input.

An alternative way of tuning is to tune using the Q resistor as the input. This requires the Q resistor be lifted from ground and connecting the signal source to the normally grounded end of the Q resistor. This has the problem that when the Q resistor is grounded after tuning, its value is decreased by the output impedance of the source. This technique has the advantage of not requiring an additional resistor.

TUNING PROCEDURE (See Figure 17)

Center Frequency Tuning

Set oscillator to center frequency desired for the filter section, adjust amplitude and check that clipping does not occur at the lowpass output pin 5 (AF100J).

Adjust the resistance between pins 13 and 7 until the phase shift between input and bandpass output is 180°.

Q Tuning

Set oscillator to upper or lower 45° frequency (see tuning tips) and tune the Q resistor until the phase shift is 135° (upper 45° frequency) or 225° (lower 45° frequency).

Zero Tuning

Set the oscillator output to the zero frequency and tune the zero resistor for a null at the ouput of the summing amplifier.

Gain Adjust

Set the oscillator to any desired frequency and the gain can be adjusted by measuring the output of the summing amplifier and adjusting the feedback resistance.





FILTER DESIGN

Since most filter tables are in terms of a normalized lowpass prototype, the filter to be designed is usually reduced to a lowpass prototype. After the lowpass transfer function is found, it is transformed to obtain the transfer function for the actual filter desired. *Graph I* shows the lowpass amplitude response which can be defined by four quantities.

Applications Information (Continued) GRAPH I. Lowpass Prototype Response



 A_{MAX} = the maximum peak to peak ripple in the passband.

- A_{MIN} = the minimum attenuation in the stopband.
- f_C = the passband cuttoff frequency.
- fs = the stopband start frequency.

By defining these four quantities for the lowpass prototype the normalized pole and zero locations and the Q (quality) of the poles can be determined from tables or by computer programs.

To obtain the lowpass prototype for the highpass filter (Graph J) A_{MAX} and A_{MIN} are the same as for the lowpass case but $f_C = 1/f_2$ and $f_S = 1/f_1$.







To obtain the lowpass prototype for a bandpass filter (Graph K) A_{MAX} and A_{MIN} are the same as for the lowpass case but

$$f_{\rm C} = 1$$
 $f_{\rm S} = \frac{f_5 - f_1}{f_4 - f_2}$

where $f_3 = \sqrt{f_1 f_5} = \sqrt{f_2 f_4}$ i.e., geometric symmetry $f_5 - f_1 = A_{MIN}$ bandwidth

 $f_4 - f_2 = Ripple bandwidth$

GRAPH K. Bandpass Response



To obtain the lowpass prototype for the notch filter (Graph L) A_{MAX} and A_{MIN} are the same as for the lowpass case and

$$f_{\rm C} = 1$$
 $f_{\rm S} = \frac{f_5 - f_1}{f_4 - f_2}$

where $f_3 = \sqrt{f_1 f_5} = \sqrt{f_2 f_4}$



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Normalized Lowpass Transformed to Un-Normalized Lowpass

The normalized lowpass filter has the passband edge normalized to unity. The un-normalized lowpass filter instead has the passband edge at f_C. The normalized and un-normalized lowpass filters are related by the transformation s = sw_C. This transforms the normalized passband edge s = j to the un-normalized passband edge s = jw_C.

Normalized Lowpass Transformed to Un-Normalized Highpass

The transformation that can be used for lowpass to highpass is $S=\omega_C/s$. Since S is inversely proportional to s, the low frequency and high frequency responses are interchanged. The normalized lowpass $1/(S^2+S/Q+1)$ transforms to the un-normalized highpass

$$\frac{s^2}{s^2 + \frac{\omega_C}{Q}s + \omega_C^2}$$

Normalized Lowpass Transformed to Un-Normalized Bandpass

The transformation that can be used for lowpass to bandpass is S = $(s^2 + \omega_0^2)/BWs$ where ω_0^2 is the center frequency of the desired bandpass filter and BW is the ripple bandwidth.

Normalized Lowpass Transformed to Un-Normalized Bandstop (or Notch)

The bandstop filter has a reciprocal response to a bandpass filter. Therefore a bandstop filter can be obtained by first transforming the lowpass prototype to a highpass and then performing the bandpass transformation.

SELECTION OF TRANSFER FUNCTION

The selection of a function which approximates the shape of the response desired is a complicated process. Except in the simplest cases it requires the use of tables or computer programs. The form of the transfer function desired is in terms of the pole and zero locations. The most common approximations found in tables are Butterworth, Tschebycheff, Elliptic, and Bessel. The decision as to which approximation to use is usually a function of the requirements and

system objectives. Butterworth filters are the simplest but have the disadvantage of requiring high order transfer functions to obtain sharp roll-offs.

The Tschebycheff function is a min/max approximation in the passband. This approximation has the property that it is equiripple which means that the error oscillates between maximums and minimums of equal amplitude in the passband. the Tschebycheff approximation, because of its equiripple nature, has a much steeper transition region than the Butterworth approximation.

The elliptic filter, also known as Cauer or Zolotarev filters, are equiripple in the passband and stopband and have a steeper transition region than the Butterworth or the Tschebycheff.

For a specific lowpass filter three quantities can be used to determine the degree of the transfer function: the maximum passband ripple, the minimum stopband attenuation, and the transition ratio (tr = ω_S/ω_C). Decreasing A_{MAX}, increasing A_{MIN}, or decreasing tr will increase the degree of the transfer function. But for the same requirements the elliptic filter will require the lowest order transfer function. Tables and graphs are available in reference books such as "Reference Data for Radio Engineers", Howard W. Sams & Co., Inc., 5th Edition, 1970 and Erich Christian and Egon Eisenmann, "Filter Design Tables and Graphs", John Wiley and Sons, 1966.

For specific transfer functions and their pole locations such text as Louis Weinberg, "Network Analysis and Synthesis", McGraw Hill Book Company, 1962 and Richard W. Daniels, "Approximation Methods for Electronic Filter Design", McGraw-Hill Book Company, 1974, are available.

DESIGN OF CASCADED MULTISECTION FILTERS

The first step in designing is to define the response required and define the performance specifications:

1. Type of filter:

Lowpass, highpass, bandpass, notch, allpass

- 2. Attenuation and frequency response
- 3. Performance

Center frequency/corner frequency plus tolerance and stability

Insertion loss/gain plus tolerance and stability

Source impedance

Load impedance

Maximum output noise Power consumption Power supply voltage Dynamic range Maximum output level

The second step is to find the pole and zero location for the transfer function which meet the above requirements. This can be done by using tables and graphs or network synthesis. The form of the transfer function which is easiest to convert to a cascaded filter is a product of first and second order terms in these forms:

First Order	Second Order

2ء

$$\frac{K}{s + \omega_{R}} \qquad \frac{K}{s^{2} + \frac{\omega_{0}}{Q}s + \omega_{0}^{2}} \qquad \text{(lowpass)}$$

$$\frac{Ks}{s + \omega_{R}} \qquad \frac{Ks^{2}}{2 + \frac{\omega_{0}}{Q}s + \omega_{0}^{2}} \qquad \text{(highpass)}$$

$$\frac{1}{s^2 + \frac{\omega_0}{Q}s + \omega_0^2}$$
 (highpass)

$$\frac{\text{Ks}}{\frac{\omega_0}{2} \text{s} + \omega_0^2}$$
 (bandpass)

$$\frac{K(s^2 + \omega_Z^2)}{s^2 + \frac{\omega_0}{Q}s + \omega_0^2} \quad \text{(notch)}$$
$$s^2 - \frac{\omega_0}{Q}s + \omega_0^2$$

$$\frac{Q}{s^2 + \frac{\omega_0}{Q}s + \omega_0^2}$$
 (allpass)

Each of the second order functions is realizable by tuning an AF100 stage. By cascading these stages the desired transfer function is realized.

CASCADING SECOND ORDER STAGES

The primary concern in cascading second order stages is to minimize the maximum difference in amplitude from input to output over the frequencies of interest. A computer program is probably required in very complicated cases but some general rules that can be used that will usually give satisfactory results are:





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Applications Information (Continued)

- 1. The highest "Q" pole pair should be paired with the zero pair closest in frequency.
- If highpass and lowpass stages are cascaded the lowpass sections should be the higher frequency and highpass sections the lower frequency.
- 3. In cascaded filters of more than two sections the first section should be the section with "Q" closest to 0.707 and then additional stages should be added in order of least difference between first stage Q and their Q.













Applications Information (Continued)



Input Level 1V rms 0 dBV

FIGURE 25. Test Circuit Block Diagram

COMPUTER AIDED DESIGN EXAMPLE*

This design is an example of a 60 Hz notch filter. The response is to have the following specifications:

Maximum passband ripple 0.1 dB

Minimum rejection 35 dB

0.1 dB bandwidth 15 Hz max

-35 dB bandwidth 1.5 Hz min

- The steps in the design of this filter are:
- 1. Design a lowpass "prototype" for the filter.
- 2. Transformation of the lowpass prototype into a notch filter design.
- 3. Using the pole and zero locations found in step two calculate the value of the resistors required to build the filter.
- 4. Draw a schematic of filter using values obtained in step three.

*Computer programs shown are user interactive. Bold copy is user input, light copy is computer response, and line indications in parenthesis are included for easy identification of data common to several programs.

PROGRAM NO. 1

RUN

THIS PROGRAM DESIGNS BUTTERWORTH CHEBYCHEFF OR ELLIPTIC NORMALIZED LOWPASS FILTERS WHAT TYPE OF FILTER? B-C-E

ELLIPTIC

```
DO YOU KNOW THE ORDER OF THE FILTER? Y/N
? NO
INPUT FC,FS,AMAX,AMIN
? 1, 10, .1, 35
    FC
                       1.000
   FS
                      10.000
    AMAX
                        .100
   AMIN
                      35.000
   Ν
                      2.000
    ATT AT FS
                    -35.671
                                (ATTENUATION IN dB)
IS THIS SATISFACTORY? Y/N
? YES
                Q
    F
      1.823 (Line 1.1)
                                      .775 (Line 1.2)
    z
    14.124 (Line 1.3)
```

PROGRAM NO. 2 (DETERMINES UN-NORMALIZED POLE + ZERO LOCATIONS OF FIRST SECTION) (DATA ENTERED FROM PROGRAM NO. 1) RUN WHAT TYPE FILTER BANDPASS OR NOTCH ? NOTCH ENTER # OF POLE PAIRS? 1

ENTER # OF JW AXIS ZEROS? 1 ENTER # OF REAL POLES? 0 ENTER # OF ZEROS AT ZERO? 0 ENTER # OF COMPLEX ZEROS? 0 ENTER # OF REAL ZEROS? 0 ENTER F&Q OF EACH POLE PAIR ? 1.823. .775 (FROM LINE 1.1 AND LINE 1.2) ENTER VALUES OF JW AXIS ZEROS ? 14.124 (FROM 1.3)

ENTER FREQUENCY SCALING FACTOR

?1

ENTER THE # OF FILTERS TO BE DESIGNED ?1

ENTER THE C.F. AND BW OF EACH FILTER ? 60, 15

OUTPUT OF PROGRAM NO. 2 TRANSFORMED POLE/ZERO LOCATIONS FIRST SECTION POLE LOCATIONS

\cap

CENTER FREQ. 56 93601 (From Line 2.3) 11.31813 (From Line 2.4) 63.228877 (From Line 2.5) 11.31813 (From Line 2.6) JW AXIS ZEROS 59.471339 (From Line 2.1) 60.533361 (From Line 2.2)

PROGRAM NO. 3 (CHECK OF FILTER RESPONSE USING PROGRAM NO. 2 DATA BASE)

RUN

NUMERATOR [ZEROS] $A(I)S \land 2 + R(I)S + Z(I) \land 2$ 0 59.471339 (From Line 2.1) 1 60.533361 0 (From Line 2.2) 1

REAL POLE

60.400

-42.757

COMPLEX POLE PAIRS

	F	Q				
1	5	56.93601 11.31813		(From Lines 2.3 and 2.4)		
2	6	3.228877	11.31813	(From Lii	nes 2.5 and 2.6)	
RUN	N					
FRE	<u>0</u> .	NOR. GAII (DB)	N PHASE	DELAY	NOR. DELAY	
40.0	000	.032	347.69	.002275	5.847169	
45.0	000	.060	342.20	.004107	8.749738	
50.0	000	.100	330.70	.009983	21.268142	
55.0	000	795	290.54	.046620	99.324027	
56.0	000	-2.298	270.61	.063945	136.234562	
57.0	000	-5.813	245.51	.072894	155.299278	
58.0	000	- 12.748	220.19	.065758	140.096912	
58.2	200	- 14.740	215.54	.063369	135.006390	
58.4	100	- 17.032	211.06	.060979	129.914831	
58.6	600	- 19.722	206.76	.058692	125.043324	
58.8	300	-22.983	202.61	.056588	120.561087	
59.0	000	-27.172	198.60	.054724	116.589928	
59.2	200	-33.235	194.72	.053139	113.212012	
59.4	100	-46.300	190.94	.051856	110.478482	
59.6	500	-42.909	7.24	.050888	108.417405	
59.8	300	- 36.897	3.60	.050242	107.040235	
60.0	00	-35.567	360.00	.049916	106.346516	
60.2	200	-36.887	356.41	.049907	106.326777	

352.81 .050206

106.963750

FREQ.	(DB)	PHASE	DELAY	NOR. DELAY
60.600	-47.102	169.17	.050801	108.232021
60.800	-33.650	165.48	.051677	110.096278
61.000	-27.577	161.72	.052809	112.508334
61.200	-23.418	157.87	.054167	115.403169
61.400	-20.198	153.92	.055712	118.694436
61.600	- 17.554	149.85	.057391	122.270086
61.800	- 15.308	145.65	.059136	125.989157
62.000	- 13.362	141.33	.060869	129.681062
63.000	-6.557	118.23	.065975	140.559984
64.000	-2.936	95.30	.059402	126.556312
65.000	-1.215	76.38	.045424	96.774832
66.000	463	62.43	.032614	69.484716
67.000	138	52.44	.023498	50.062947
70.000	.091	35.43	.010452	22.267368
75.000	.085	23.44	.004250	9.054574
80.000	.060	17.80	.002310	4.921727
85.000	.043	14.50	.001460	3.110493
90.000	.032	12.31	.001011	2.154297

NOR GAIN





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FIGURE 27. DC Output Voltage Due to Amplifier VOS

DEFINITION OF TERMS

AMAX Maximum passband peak-to-peak ripple

- A_{MIN} Minimum stopband loss
- fz Frequency of jw axis pair
- fo Frequency of complex pole pair
- Q Quality of pole
- f_C Passband edge
- fs Stopband edge
- AHP Gain from input to highpass output
- ABP Gain from input to bandpass output
- ALP Gain from input to lowpass output
- AAMP Gain from input to output of amplifier
- R_f Pole frequency determining resistance
- R_Z Zero frequency determining resistance
- RQ Pole quality determining resistance
- f_H Frequency above center frequency at which the gain decreases by 3 dB for a bandpass filter
- f_L Frequency below center frequency at which the gain decreases by 3 dB for a bandpass filter
- BW The bandwidth of a bandpass filter
- N Order of the denominator of a transfer function

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