

# Wireless World Circard

## Series 1: Basic active filters

This set of cards—the best selling one of all—deals mainly with second-order active filters i.e. those with active devices for gain having a second-order transfer function. The most popular nowadays are probably the Sallen & Key types using unity-gain amplifiers (cards 4 and 5). Though widely used, they suffer from the disadvantage that two resistors need to be made variable and ganged for variable frequency applications. The positive feedback Wien-bridge notch filter too needs to have two components made variable to alter its centre frequency. Three cards deal with filters that are tunable by just one potentiometer (cards 3, 8, 10), and card 9 includes a bibliographical reference to a not so well-known twin-T circuit having single-component variation of the notch frequency.

One tunable filter technique that uses digital integrated circuits is given in the *n*-path filter card. Here resonant frequency is varied by changing commutating frequency, leaving bandwidth constant.

Many of the circuits given use integrated-circuits, mostly the 741 op-amp. (Card 6 is an exception, using the TAA960 i.c. with four inverters.) But for many low-frequency, low-Q applications, the op-amps can be replaced by a single transistor or Darlington pair. In the Sallen & Key types, the discrete device would need to be used as an emitter follower.

- Wien-bridge bandpass filter 1
- Wien-bridge all-pass network 2
- Voltage-controlled filters 3
- Low-pass Sallen & Key filter 4
- High-pass Sallen & Key filter 5
- Low, high and band-pass triple amplifier 6
- Op-amp triple with phase compensation 7
- Multi-feedback filter 8
- Adjustable-Q twin-T notch filter 9
- Easily-tuned notch filter 10
- Compound filters 11
- N-path filter 12

# Basic Active Filters

Like the dotting parent reproached for a birthday present of an encyclopaedia with the response "It tells me more about things than I want to know", any writer on this topic has to exercise care.

The filters to be described belong to that broad class of circuits using active devices together with passive components, some of which are reactive, to produce a frequency-dependent transfer function. The possibilities range from single  $CR$  time-constants in the feedback path of an amplifier to complex multi-gyator circuits. The former have applications in audio frequency work, while the latter are still the subject of continuing research. It is the in-between varieties that we shall consider, in particular second-order filters of various kinds.

These have numerous applications used separately but may in addition be combined to produce more complex filter functions. By second-order is meant that the transfer function is represented by an equation containing  $\omega^2$  as well as  $\omega$ . This is because of the presence of at least two reactive components — capacitors or inductors. The latter are inconvenient and expensive below the radio-frequency range, and active-filters dispense with them by synthesizing the desired function using capacitors.

The common filter functions are

**Low-pass:** signals below a given cut-off frequency are passed at, or amplified to, a fixed level, falling by 3 dB at the cut-off frequency with a slope that reaches 12 dB per octave for a second-order filter — Fig. 1.

**High-pass:** the pass-band lies above the cut-off frequency, with 3 dB attenuation at the cut-off frequency, and with attenuation increasing at 12 dB/octave below this frequency — Fig. 2.

**Band-pass:** signals at both low and high frequencies are progressively attenuated with a maximum response at the centre frequency; the sharpness of the response is measured by the  $Q$  of the circuit with  $Q = f_0/2\Delta f$ , where  $f_0$  is the centre frequency and  $2\Delta f$  is the bandwidth for an attenuation of 3 dB relative to that at the centre frequency — Fig. 3.

**Band-stop or notch:** low and high frequencies are passed with a maximum attenuation (ideally zero transmission) at the centre or notch frequency — Fig. 4.

**All-pass:** the magnitude of the transfer-function is constant but with a frequency-dependent phase-shift — Fig. 5.

The low- and high-pass filters are sometimes modified so that the output falls to a low but constant value at some other frequency — Figs. 6 and 7. The band-pass and

band-stop filters may require extended frequency ranges within which the pass and stop functions are maintained. One possibility is the use of complex high-order filters. Alternatively a combination of one or more second-hand filters tuned to different frequencies can be used. A perfect null response is possible only at precise specified frequencies — Figs. 8 and 9.

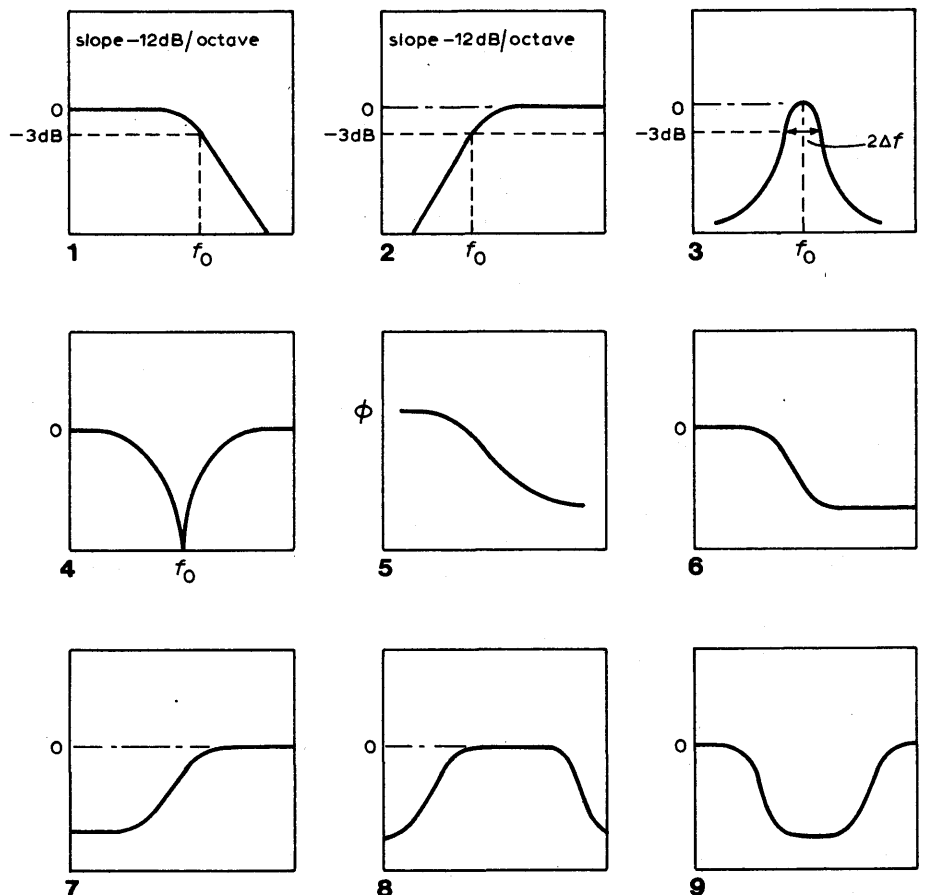
In determining which active circuit configuration most suits a particular filter function, the required  $Q$  is a major factor. High  $Q$  is often a requirement for band-pass and notch filters. It can be attained by two fundamentally different methods.

The first, requiring a single amplifier in the simplest cases, arranges for  $Q$  to depend on the difference between two nearly equal terms in the transfer function. The circuits using this technique are sometimes recognizable as a bridge circuit with the bridge almost at balance, combined negative and positive feedback with the latter almost equalling the former, or a negative-impedance converter in which an imped-

ance in one part of a circuit appears elsewhere with a negative sign.

Any given circuit may be recognizable under more than one of the above headings which might be considered as different manifestations of the same underlying process. While high  $Q$  is possible, it becomes over sensitive to component and environmental changes, as it depends on the near-cancellation of comparable terms. The fractional change in  $Q$  caused by a change in some component is broadly proportional to the nominal  $Q$  aimed for, e.g. a circuit with  $Q = 20$  as the design value might have a practical value of  $Q = 25$  for a 1 or 2% change in a particular resistor.

The corresponding practical values for nominal  $Q$  values of 10 and so might be 11 and 75 respectively. The centre frequency is also dependent on component values but active circuits are comparable to passive networks in this respect. The exception is at high frequencies when the amplifier limitations may be all too apparent. This is shown by the increasing departures at the observed



In these graphs illustrating low-pass, high-pass, band-pass, band-stop and all-pass active filters, transfer function magnitude is plotted as the ordinate with frequency as the abscissa, except in 5 where the ordinate is phase angle.

centre-frequency from the nominal value, coinciding often with a fall in the achievable value of  $Q$ .

The second high- $Q$  approach uses multiple active devices. One type, well-known in analogue computing, uses two integrators in a closed loop such that the open-loop gain of the system at the centre frequency, and hence  $Q$ , is controlled by a single resistor. The same pattern of passive components reappears in filters based on the gyrator, a device which has an impedance at one port related to that presented to a second port in such a way that a capacitance is "gyrated" into an inductance, a parallel tuned circuit into a series tuned circuit, etc. Gyrotors may be constructed from various configurations of amplifiers, with the usual limitations on frequency response, for example.

In neither approach is the  $Q$  excessively sensitive to component variations. Broadly the sensitivity is comparable to that obtaining with passive circuits, i.e. a 1% change

in any component will not normally change the  $Q$  by more than 1% even at high values of  $Q$ . With multiple amplifier filters, there is more than one output available, and some circuits have band-pass, low-pass and high-pass functions available simultaneously. The falling cost of operational amplifiers makes such solutions attractive, particularly for larger systems where the flexibility is a considerable advantage. A problem can be the phase-shift in multiple amplifiers, which can result in oscillation when high  $Q$  is attempted at frequencies near the upper frequency limit of the amplifier.

For simple low-pass and high-pass applications the single-amplifier methods are suitable, and as an example scratch and rumble filters may use twin-gang potentiometers for widely variable cut-off frequencies. Using special i.c. amplifiers with very high input resistance the cut-off frequency can be readily extended below 1 Hz if required, while the upper limit approaches 1 MHz.

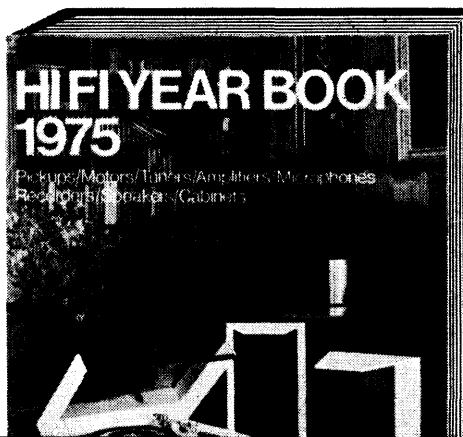
A completely different approach which is certain to become dominant in many fields is that of the  $n$ -path filter. There will presumably be as many variants as there are at present with the analogue approach, but the basic technique appearing in some recently published circuits is as follows.

In place of a single filter a number, say  $n$ , of identical units are switched into the circuit in succession. These may be, for example, low-pass filters, passive or active, or simply the capacitors of a set of  $CR$  circuits using a common  $R$ . Switching frequency is set to  $n$  times the fundamental frequency to which the filter is to be tuned.

The filter has responses at harmonics of the fundamental frequency and may have a very high  $Q$  at each, but simple low-cost filters of the type described earlier can be used to attenuate these responses.

A vital point is that the filter centre-frequency is controlled by the clock-rate driving the switches, and therefore readily variable over a wide range, as is bandwidth.

# THE WHO, WHAT AND WHERE OF HI-FI

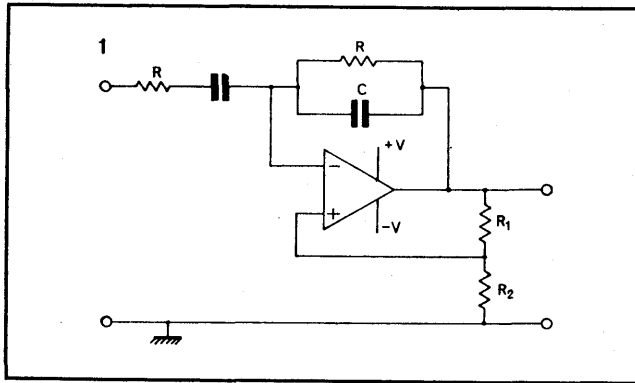


In the search for quality it helps to know where to look. Hi-Fi Year Book tells you everything you need to know, with separate illustrated sections for every major category, giving you prices and specifications of over 2,000 products. And it's got directories of dealers and manufacturers — plus a host of articles on the latest hi-fi developments and their application. So if you want information like you want hi-fi, order your copy today because it sells out pretty quickly! £2.00 inclusive from the publishers, or from newsagents and bookshops.

## HI-FI YEAR BOOK 1975

IPC Electrical-Electronic Press Ltd.,  
Dorset House, Stamford Street, London SE1 9LU.

## Wien-bridge bandpass filter



## Typical performance at 1kHz

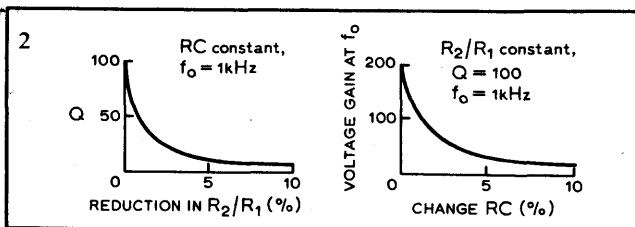
$$f_0 = \frac{1}{2\pi CR} \text{ Hz}$$

$$Q = \frac{1}{2 - (R_2/R_1)}$$

$$\text{Voltage gain, } \frac{V_{out}}{V_{in}} \approx -3Q$$

To find  $Q$  measure voltage gain and divide by  $-3$

IC : 741; supplies  $\pm 15V$   
 $C = 1nF \pm 1\%$   
 $R = 150k\Omega \pm 5\%$   
 $(R_2 + R_1) = 10k\Omega$   
 Source resistance:  $60\Omega$   
 $V_{out}(\text{max.}) = 9.2V \text{ r.m.s.}$   
 Max. load current:  $12mA$   
 $Q$  is constant for supplies between  $\pm 3V$  and  $\pm 18V$  if clipping is avoided.  
 $Z_{in} \propto 1/Q$ ;  $Z_{out} \propto Q$ .



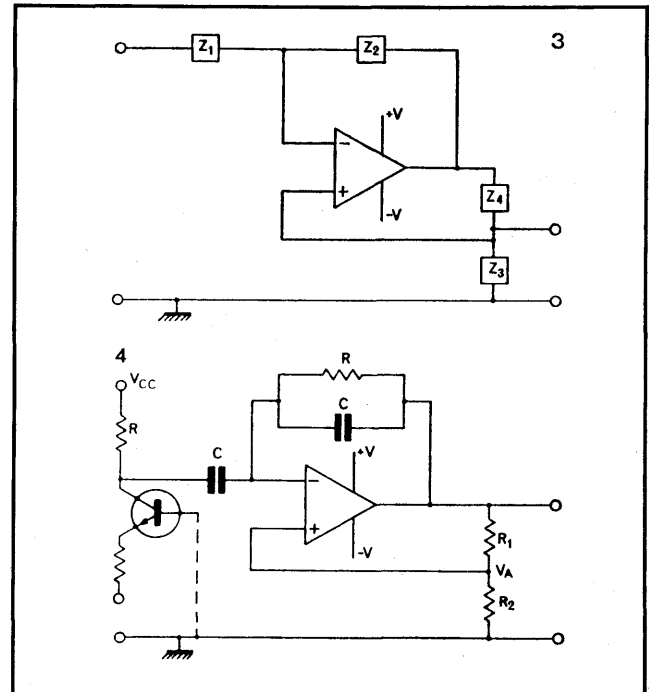
## Circuit description

The Wien network—Fig. 1—gives a maximum response at  $f_0 = 1/2\pi RC$  Hz. Positive feedback through  $R_1$ ,  $R_2$  sharpens the response, e.g. with  $R_1$  and  $R_2$  replaced by a potentiometer, continuous control of circuit  $Q$  is obtained without change in centre frequency. Changing both  $R$ s or both  $C$ s, maintaining equality, varies  $f_0$  without changing  $Q$ . Sensitivity to component changes is proportional to  $Q$  for large  $Q$ . At high frequencies, amplifier phase shift causes  $Q$  and  $f_0$  to depart from nominal values. Input impedance falls as  $Q$  increases. Satisfactory for moderate  $Q$  values (5–25) at frequencies in the audio range. High  $Q$  is obtainable if only short-term stability is required.

## Component changes

● Using  $\pm 1\%$  capacitors and  $\pm 5\%$  resistors,  $f_0$  will typically be within  $\pm 5\%$  of theoretical value up to approximately  $15kHz$  and within  $\pm 10\%$  up to about  $22kHz$ .

- Large  $R$ -values for low  $f_0$  produces an output d.c. level up to about  $1V$  which can be reduced with offset null adjustment.
- Source resistance should be  $\ll R$  for predictable  $f_0$  and  $< R/15Q$  for predictable  $Q$  within  $-5\%$ .
- Circuit will oscillate when  $R_2/R_1 \geq 2$ .
- 741 op-amp may be replaced by a 748 or 301 using a  $30-pF$  compensation capacitor.
- Reducing compensation capacitor of 748 or 301 to  $3.3pF$  typically makes  $f_0$  predictable to within  $\pm 5\%$  of theoretical value up to about  $22kHz$ .



## Circuit modifications

- Fig. 3 shows the general form of the circuit, one version of which has been discussed. Other configurations are possible by interchanging  $Z_1$  and  $Z_4$  and/or  $Z_2$  and  $Z_3$ . Note that the output may not then be taken from the op-amp output.
- Since  $Z_{in} \propto 1/Q$  for high  $Q$  a buffer amplifier may be added at the input. Buffer at the output is only required for alternative versions (except at very high  $Q$ ).
- $R$  at the input of the circuit may be taken to ground and from a current source e.g. from collector of a common-base stage—Fig. 4—or a cascode stage.

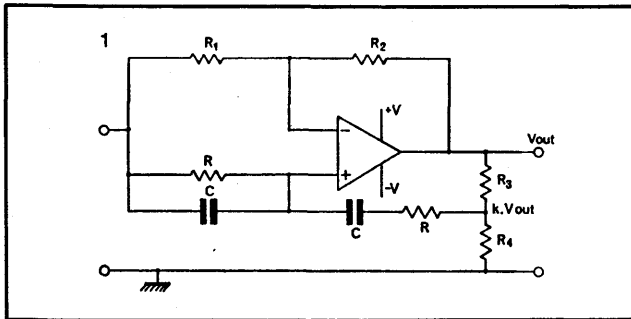
## Further reading

1. Williams, P., Bandpass filters using Wien's bridge, *Electronics Letters*, 1970, vol. 6, p.186.
2. Kesner, D. An op. amp. RC bandpass filter, Motorola application note AN-452.

## Cross references

Series 1, cards 3, 6, 7, 8, 11 & 12.

## Wien-bridge all-pass network



## Typical performance at 1kHz

180° phase shift occurs at  $f_0$ , where

$$f_0 = \frac{1}{2\pi RC} \text{ Hz}$$

$$\lambda = R_1/R_2$$

For unity voltage gain

$$k = \frac{5\lambda - 1}{\lambda + 1}$$

IC : 741; supplies  $\pm 15V$

$C = 1nF \pm 1\%$

$R = 150k\Omega \pm 5\%$

$(R_3 + R_4) = 1k\Omega$

$(R_1 + R_2) = 10k\Omega$

Source resistance: 60 $\Omega$

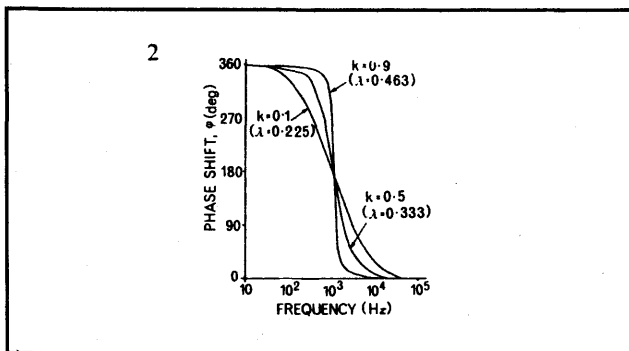
With unity voltage gain

$V_{out(max.)} = 26V$  pk-pk.

No significant change in

performance for supplies

in the range  $\pm 3$  to  $\pm 18V$ .



## Circuit description

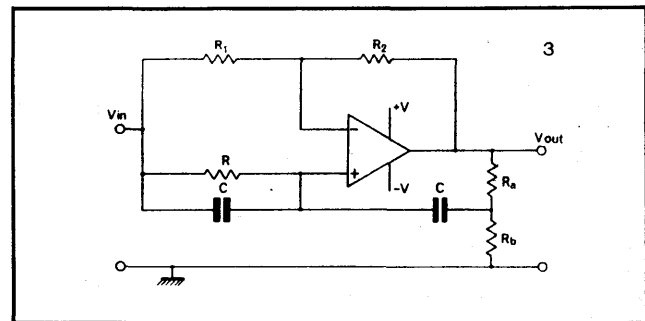
When  $\lambda$  has the correct value for a given value of  $k$  the network has unity voltage gain but the phase difference ( $\phi$ ) between output and input is frequency dependent. At  $f_0 = 1/2\pi RC$  Hz,  $\phi = 180^\circ$ . Increasing  $k$  and adjusting  $\lambda$  to the appropriate value increases the magnitude of  $d\phi/df$  in the region of  $f_0$ . Suitable for use at audio frequencies.

## Component changes

● 741 i.c. may be replaced with a 748 or a 301 using a 30-pF compensation capacitor without change in performance e.g.  $d\phi/df$  characteristic predictable up to  $f_{0(max)}$  of about 10kHz, with unity voltage gain.

● Reducing compensation capacitor to 3.3pF extends range of predictable performance up to  $f_{0(max)}$  of about 20kHz.

● With  $R = 150k\Omega$  and  $(R_1 + R_2) = 10k\Omega$ ,  $(R_3 + R_4)$  may be increased to approximately 22k $\Omega$  without significant change in performance.



## Circuit modifications

● As in the bandpass filter (card 1) the arms of the bridge may be interchanged to produce alternative configurations having differing  $d\phi/df$  effects.

● For fixed  $d\phi/df$  and  $f_0$ , the arrangement in Fig. 3 may be used where  $R_a$  and  $R_b$  in parallel equals  $R$  and their ratio determines the value of  $K$ .

● For variable  $d\phi/df$  characteristic, without having to incorporate output potential divider resistance in  $CR$  calculations, insert a voltage follower between the potential divider and the series network.

● A buffer amplifier may be used at the input, though the input impedance is not likely to be as low as in the high- $Q$  bandpass filter-card 1.

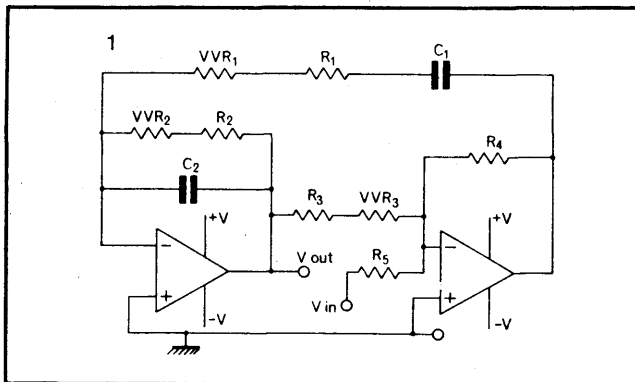
## Further reading

1. Williams, P. Allpass networks using Wien's bridge *Electronics Letters*, vol. 6, 1970 p.183.
2. As 1: Alternative allpass networks using Wien's bridge, p.188.
3. Mitra, S. K., Analysis and synthesis of linear active networks, Wiley, 1969 pp.479-81.

## Cross references

Series 1, card 1.

## Voltage-controlled filters



## Typical performance at 1kHz

$$f_0 = \frac{1}{2\pi RC} \text{ Hz}$$

where  $R = (R_1 + VVR_1) = (R_2 + VVR_2)$

ICs 741; supplies  $\pm 6V$

VVRs:  $1/4 \times CD4016AE$

$C_1 = C_2 = C = 0.1 \mu\text{F} \pm 2\%$

$R_1 = R_2 = 1k\Omega \pm 5\%$

$R_3 = 1.2k\Omega \pm 5\%$

$R_4 = 3.3k\Omega \pm 5\%$

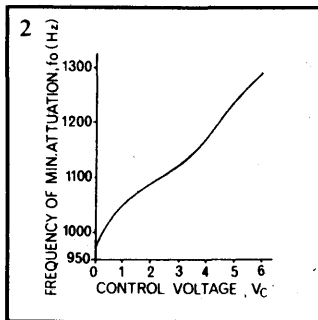
$R_5 = 10k\Omega \pm 5\%$

$V_{DD} = +3V, V_{SS} = -3V$

$V_C = 0$  to  $+6V$

VVRs variable within the range approximately 680 to 280 $\Omega$ .

$V_{C3} = 4.32V$  to produce  $Q \approx 10$  when  $V_{C1} = V_{C2} = 0V$ .



## Circuit description

This bandpass circuit is related to that of card 1. Separation of the frequency-sensitive and resistive networks allows each section to have one terminal as a virtual earth. This simplifies the injection of signals (defined input resistance e.g. through  $R_5$ ) and allows devices such as m.o.s. transistors to be used as frequency and/or  $Q$  control elements, with bias voltages referred to ground. The control elements used were complementary m.o.s. transmission gates with resistances controlled by the applied bias voltages.  $VVR_1$  and  $VVR_2$  control  $f_0$  and  $VVR_3$  controls  $Q$  by varying the ratio  $R_4/(R_3 + VVR_3)$ .

## Component changes

● Supplies may be varied within the range  $\pm 3$  to  $\pm 18V$  and are conveniently chosen to be compatible with supplies required for the VVR elements. (N.B. CD4016AE max  $V_{DD} - V_{SS} = 15V$ )

● The greater the ratio  $VVR_n/R_n$ ,  $n = 1$  or  $2$ , the wider the range of  $f_0$  variation for given values of  $C$  within the audio band.

● For a given value of  $R_4$ ,  $Q$  variation is greatest when  $R_3 \rightarrow 0$ .

## Circuit modifications

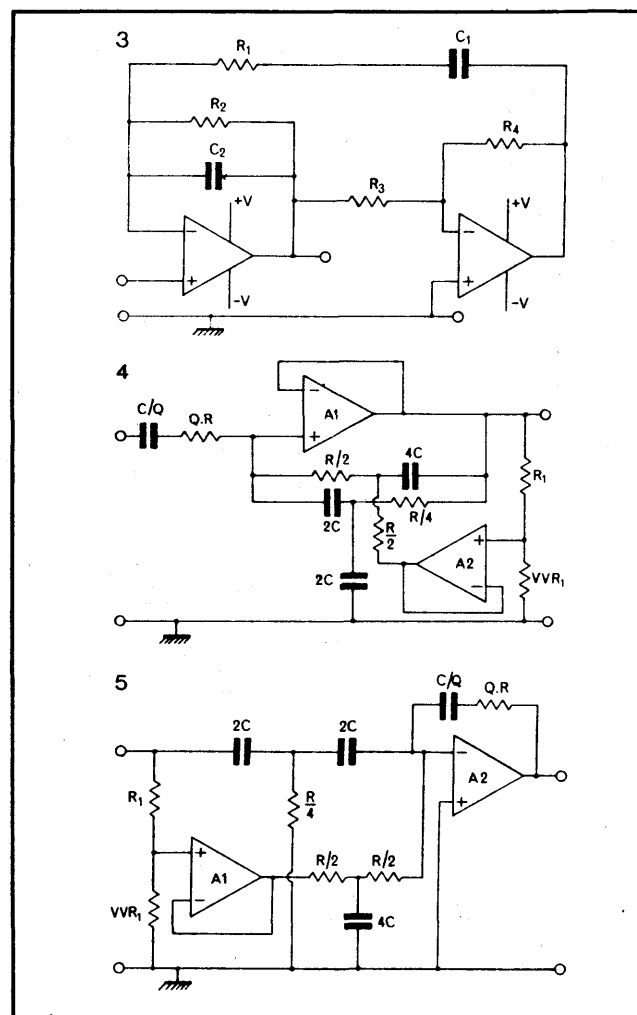
● By scaling  $R_1 = R_2/2$ ,  $C_1 = 2C_2$  and setting  $R_3 \rightarrow R_4$  the bandpass filter will provide outputs from the op-amps that are in antiphase and of almost equal amplitude.

● Any device with a fairly linear resistance, variable by an external parameter may replace the c.m.o.s. gates in this and other filters, (easiest where  $R_s$  have a common point at or near ground potential, permitting this point to be connected to sources of f.e.t.s, for example). Possibilities include CdS

photocells, temperature-sensitive resistances like copper, platinum, and thermistors where signal swings do not change resistance, i.e. relatively high-power types.

● As in all virtual-earth amplifiers, the signal may be injected in voltage form into an otherwise-grounded, non-inverting terminal, as in Fig. 3, providing a high input resistance may alter filter characteristic.

● Figs. 4 and 5 show the voltage control principle applied to a parallel-T bandpass and notch filter respectively. In each circuit  $f_0$  is controlled by the VVR.



$$\text{In Fig. 4: } f_0 = \frac{1}{2\pi RC} \cdot \sqrt{\frac{R_1}{R_1 + VVR_1}} \text{ Hz.}$$

$$\text{In Fig. 5: } f_0 = \frac{1}{2\pi RC} \cdot \sqrt{\frac{VVR_1}{R_1 + VVR_1}} \text{ Hz.}$$

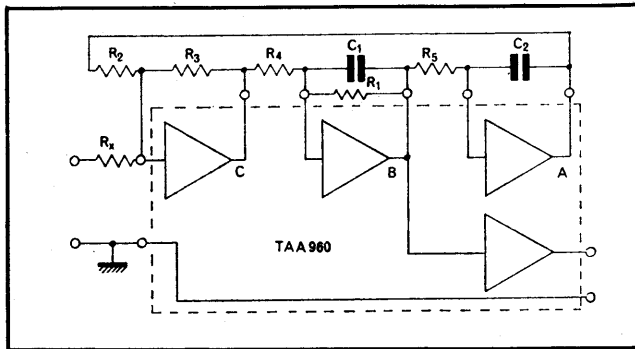
## Further reading

1. Good, E. F., Selective amplifiers with parallel-T feedback, *Electronic Engineering*, 1963 vol. 35, p.330.
2. Douce, J. L. and Edwards, K. H., Simple null filter with variable notch frequency, *Electronic Engineering*, 1964 vol. 36, pp.478/9.
3. Penn, T. C., Simple tuneable RC null networks, *Electronic Engineering*, 1964, vol. 36, p.849.
4. RCA Databook SSD-203, 1972, p.67.

## Cross references

Series 1, cards 1, 6-10.

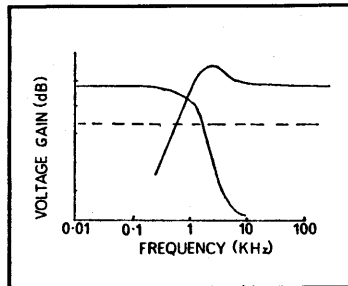
### Low, high & band-pass triple amplifier



#### Typical performance

$$f_0 = \frac{1}{2\pi \sqrt{C_1 C_2 R_4 R_5}}$$

Drive: 400mV pk-pk.  
 Supply: +6V.  
 $R_2, R_3, R_4, R_5 = 10k\Omega \pm 5\%$   
 $C_1, C_2 = 10nF \pm 1\%$   
 $R_X = 470k\Omega \pm 5\%$   
 $R_1 = \infty$   
 $f_0 = 1,583Hz$   $Q = 49.5$



#### Circuit description

Three inverting amplifiers form the filter. A fourth amplifier buffers the output. There are two integrators, one damped by resistor  $R_1$ , and a unity-gain inverting amplifier. A different filter function is obtained at the output of each amplifier; low-pass at A, bandpass at B and modified high-pass at C. Where all four amplifiers are included in a single i.c. as above, a restricted performance is offered by each amplifier individually, e.g. lower voltage gain, and the output resistance of a value that requires buffering if load resistances are not to change the filter characteristic. With  $R_1$  absent, the  $Q$  of the band-pass filter is approximately 50. As  $R_1$  is decreased, the response at A and C approach that usually required for second order, low-pass and high-pass responses respectively. Single-supply operation and direct coupling place a lower

limit on  $R_x$  of approximately  $10k\Omega$ , with  $R_x = 50R_2$  for correct biasing. In addition, all outputs have a d.c. content.

#### Component changes

Supply voltage + 6V:  $f_0 = 1583Hz$ , gain = 1.15,  $Q = 49.5$ .  
 Supply voltage + 5V:  $f_0 = 1562Hz$ , gain = 0.95,  $Q = 40$ .  
 For best  $Q$  values, capacitors should be matched within 1%.  
 For low-pass and modified high-pass filters  $R_1 = 6.8k\Omega$ ,  
 $Q = R_1/R_4 = 0.68$   
 Drive signal 2V pk-pk.  
 Output approx. 50mV pk-pk.

#### Circuit modifications

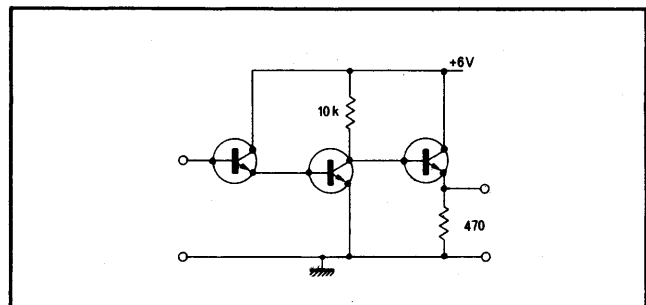
- As the three amplifiers are identical, the buffered amplifier may be used as the output for the low-pass or modified high-pass filter.
- For higher gain without disturbing d.c. conditions, input and output would have to be a.c.-coupled, then the input resistor could be reduced.
- The simple structure of the amplifiers can be duplicated by discrete circuits such as the amplifier with emitter-follower output shown.

#### Further reading

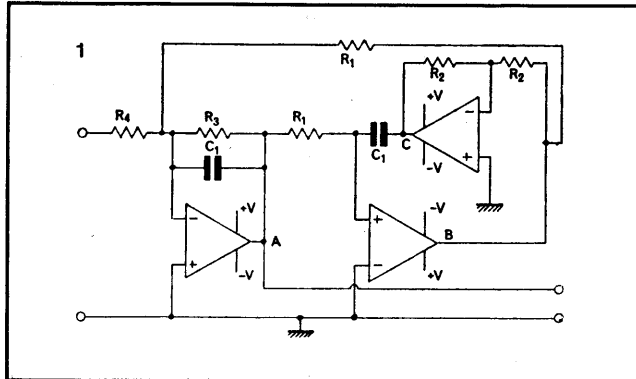
Mullard Ltd.: Triple amplifier for active filters, TAA 960.

#### Cross references

Series 1, cards 1, 3, 7, 8, 11 & 12.



## Op-amp triple with phase compensation



## Typical data

$$R_1 = 3.3\text{k}\Omega \pm 5\%$$

$$C_1 = 0.1\mu\text{F} \pm 10\%$$

$$R_3 = 330\text{k}\Omega \pm 5\%$$

$$f_0 = \frac{1}{2\pi R_1 C_1} = 482\text{Hz}$$

$$Q_0 = R_3/R_1 = 100$$

$$f_0(\text{obs}) = 470\text{Hz}$$

$$Q(\text{obs}) = 105$$

$$V_{\text{out}}/V_{\text{in}} \approx Q \text{ at } f=f_0$$

## Circuit description

The same passive components as in card 6 may be used with separate op-amps having much higher gain and more suitable input and output characteristics. With the same amplifier configuration high  $Q$  is possible with excellent stability. At high frequencies the observed  $Q$  rises because of amplifier phase shifts. The alternative configuration shown provides some degree of cancellation of phase shifts allowing high- $Q$  designs up to 100kHz with low-cost amplifiers. Precautions against over driving under these conditions are necessary as slew-rate limiting changes the amplifier response leading to the possibility of sustained oscillation.

## Component changes

$$R_1 = 470\Omega \text{ to } 1\text{M}\Omega$$

$$R_3 = 470\Omega \text{ to } \infty$$

$$C_1 = 330\text{pF} \text{ to } 10\mu\text{F} \text{ (non-polarized)}$$

Amplifiers: N5741V, SN72741P, MC1741, etc.

Supply  $\pm 5$  to  $\pm 18\text{V}$  ( $\pm 3\text{V}$  in some cases).

Alternative amplifiers: any op-amp compensated for integrator operation.

$$f_0 = < 1\text{Hz} \text{ to } > 100\text{kHz}$$

$$Q < 1 \text{ to } > 100$$

## Circuit modifications

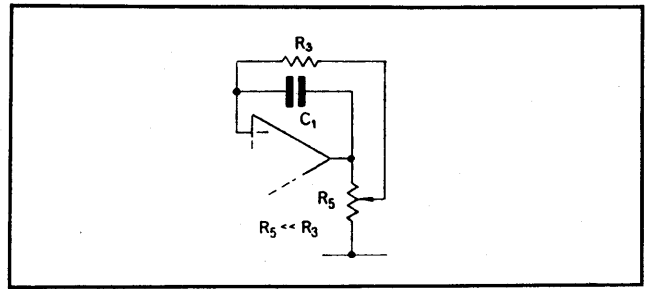
- Variable damping at high  $Q$  inconvenient because of large value for  $R_3$ . Replace by fixed resistor fed from tapping across amplifier as shown e.g.  $R_5 = 10\text{k}\Omega$  pot.,  $R_3 = 100\text{k}\Omega$ .

- Summing outputs at three amplifiers with conventional virtual earth summer gives more general transfer function.

- Low-pass output available at C (normally used with  $Q = 0.7$  for simple low-pass filters as in audio applications—higher  $Q$  gives peak in response just below cut-off frequency).

- Modified high-pass output at B—comments as above.

- Where configuration as in card 6 is used phase lead may be introduced by a small capacitor across input resistor of inverter—alternative method of neutralizing phase-shift.



## Further reading

Kerwin, W. J., Huelsman, L. P., and Newcomb, R. W. State-variable synthesis for insensitive integrated circuit transfer functions, *IEEE J. Solid-State Circuits*, vol. SC-2, 1967 pp.87-92.

Vogel, P. W. Method for phase correction in active RC circuits using two integrators, *Electronic Letters* 1971, pp.273-5.

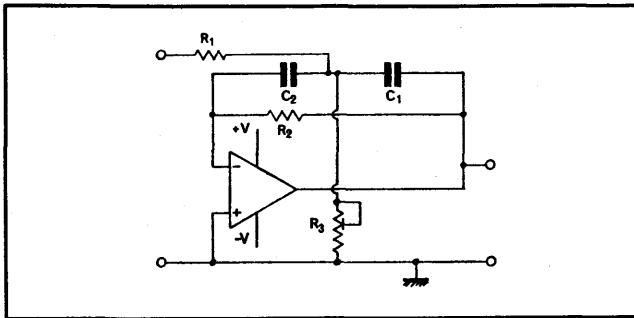
Girling, F. E. J. and Good, E. F., Active filters-7 The two-integrator loop, *Wireless World*, vol. 77, 1970 pp.76-80.

## Cross references

Series 1, cards 1, 3, 6, 8, 11 & 12.



### Multi-feedback filter



#### Typical performance

For  $C_1 = C_2 = C$

$$f_0 = \frac{1}{2\pi C} \sqrt{\frac{R_1 + R_3}{R_1 R_2 R_3}}$$

$$Q = \frac{1}{2} \sqrt{\frac{R_2 (R_1 + R_3)}{R_1 R_3}}$$

$A_0 = R_2/2R_1$   
 $C = 0.01 \mu\text{F} \pm 10\%$   
 $R_1 = R_3 = 1.5\text{k}\Omega \pm 5\%$   
 $R_2 = 470\text{k}\Omega \pm 5\%$   
 Amplifier: 741  
 $f_0(\text{obs}) = 845\text{Hz}$   
 $Q(\text{obs}) = 11.3$

#### Circuit description

The inverting amplifier has multipath feedback which has a minimum at a single frequency. The overall response is then band-pass peaking at that frequency, which may be changed by varying  $C_1$ ,  $C_2$  together, the circuit  $Q$  remaining constant. Changing  $R_3$  also varies the centre frequency keeping the bandwidth constant.

#### Component changes

$R_2$ : 1.5k $\Omega$  to 1M $\Omega$

$R_1$ ,  $R_3$ : 1k $\Omega$  to 100k $\Omega$

$C_1$ ,  $C_2$ : 150pF to 10 $\mu\text{F}$

$Q = 1$  to 50 (At high  $Q$  values, sensitivity to component variations is excessive)

$f_0 < 1\text{Hz}$  to  $> 100\text{kHz}$

#### Circuit modifications

● Varying  $R_3$  changes the centre frequency, leaving bandwidth and centre-frequency gain unchanged.

● Feeding from a current source,  $R_1$  can be omitted, readjusting  $R_3$  to give required characteristic. Alternatively if preceding stage has specified output resistance it can be incorporated into  $R_1$ .

● For low-gain low- $Q$  applications the amplifier may be replaced by a single transistor in the common-emitter mode.  $R_2$  provides base-current. Typical collector load resistances in range 1 to 10k $\Omega$  for passive network given. Alternatives include Darlington-pair amplifiers.

● Alternative passive networks for low and high-pass characteristics given in the reference.

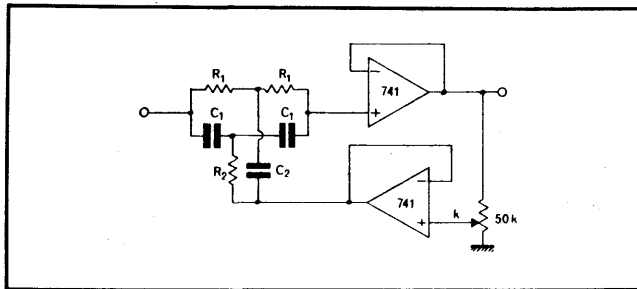
#### Further reading

Graeme, J. G., Tobey, G. E., Huelsman, L. P., Operational amplifiers: design and applications, McGraw-Hill, pp.287-95.

#### Cross references

Series 1, cards 1, 3, 6, 7, 8, 11 & 12.

## Adjustable-Q twin-T notch filter



## Typical performance

$$f_0 = \frac{\sqrt{n}}{2\pi R_1 C_1}$$

where  $n = 2C_1/C_2 =$

$R_1/2R_2 (=1 \text{ usually})$

$f_0 = 1000\text{Hz}$  with

$C_1 = 10\text{nF}$ ,

$C_2 = 20\text{nF}$  (trimmed)

$R_1 = 16.2\text{k}\Omega$

$R_2 = 8.1\text{k}\Omega$  (trimmed)

$f_1 = 680\text{Hz}$   $f_2 = 1650\text{Hz}$

$f_0 = 50\text{Hz}$  with

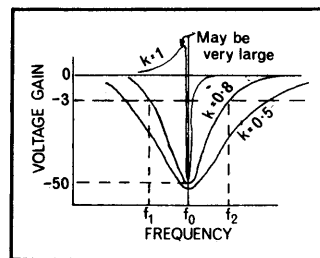
$C_1 = 10\text{nF}$ ,  $C_2 = 20\text{nF}$

(trimmed)

$R_1 = 324\text{k}\Omega$

$R_2 = 162\text{k}\Omega$  (trimmed)

$f_1 = 33\text{Hz}$   $f_2 = 76\text{Hz}$



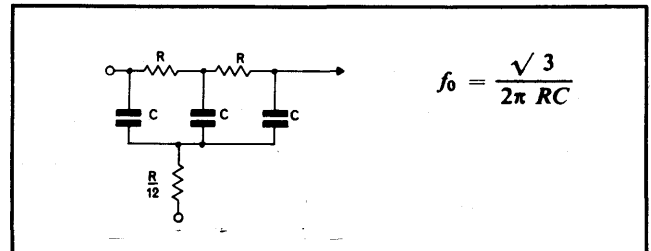
## Circuit description

The passive network has a perfect null at a defined frequency if the components are accurately matched. The rate of approach to that notch is sharpened by positive feedback, the buffer amplifier having a negligible output resistance and hence theoretically not disturbing the depth of the notch. Variation in notch frequency requires simultaneous variation e.g. of all three capacitors or of  $R_2$  and  $C_2$  and the circuit is most suitable for fixed frequency operation. In addition too much positive feedback ( $k \rightarrow 1$ ) may give unsatisfactory results.

## Component changes

With low values of  $C_1$  gain at high frequencies may not be

0dB. A 12% deviation from the nominal 0dB was observed with  $C_1 = 270\text{pF}$ ,  $R_1 = 600\text{k}\Omega$ ,  $n = 1$ ,  $f_0 = 1\text{kHz}$ . Very large resistors ( $> 10\text{M}\Omega$ ) should be avoided.



$$f_0 = \frac{\sqrt{3}}{2\pi RC}$$

## Circuit modifications

● Any circuit with a notch e.g. above may be used in place of the twin-T network e.g. ref. 1. Trimming of  $R/12$  may be necessary if  $k$  is varied over a wide range.

● Notch filters having a narrow notch width can be unsatisfactory with signals whose frequency stability is not good. This can be overcome by cascading two notch filters, having slightly different notch frequencies e.g. two filters tuned to 48 and 52Hz respectively will effectively remove mains pick-up<sup>2</sup>. It may be possible to simply cascade the passive networks with due attention to loading of the first by the second, and still only use two op-amps.

● Single component variation of the notch frequency is possible with a twin-T network (card 3 and ref. 3) and is also possible with other networks (card 10). Buffer amplifier may be omitted if potentiometer value is low.

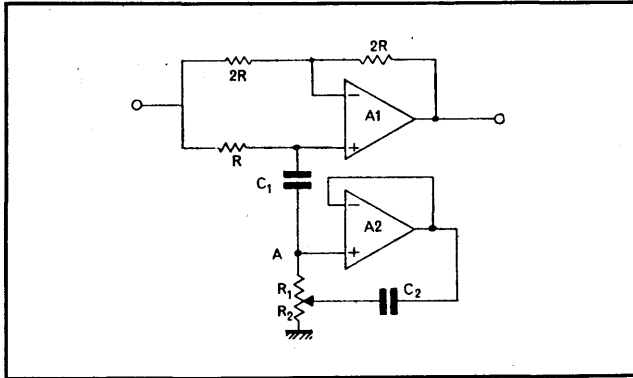
## Further reading

1. Graf. *Electronic Design Data Book*, Van Nostrand Reinhold.
2. N. B. Rowe, *Designing a low frequency active notch filter*, *Electronic Engineering*, April 1972.
3. Douce and Edwards, *Electronic Engineering*, July 1964.
4. National Semiconductor Linear Brief LB-5.

## Cross references

Series 1, cards 3 & 10.

## Easily-tuned notch filter



## Typical performance

$Q$  of resonant branch is

$$\frac{1}{R_1 + R_2} \sqrt{\frac{C_2 R_1 R_2}{C_1}}$$

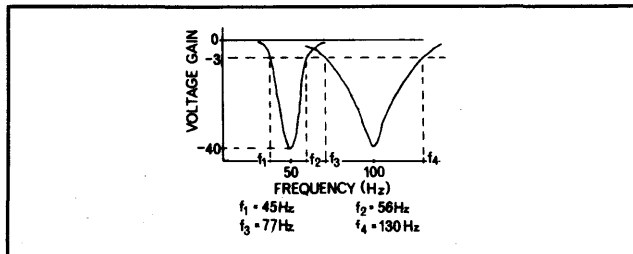
$$f_0 = \frac{1}{2\pi} \sqrt{\frac{C_1 C_2 R_1 R_2}{C_1}}$$

$f_0 = 50\text{Hz}$  with

$R_1, R_2 : 50\text{k}\Omega, C_2 : 1\mu\text{F}$

$C_1 : 4\text{nF}, R : 100\text{k}\Omega$

Results shown are for these values, with some trimming of one of the  $2R$  resistors for good notch depth. 741 op-amps were used. Considerable trimming is necessary at higher frequencies.



## Circuit description

The impedance between A and ground is equivalent to an inductor,  $C_2 R_1 R_2$ , in series with a resistor,  $R_1 + R_2$ . At a specific frequency  $C_1$  and the equivalent inductance series resonate, leaving the equivalent resistance to bring the bridge into balance provided  $(R_1 + R_2) = R$  i.e. an overall notch characteristic. Variation in the equivalent inductance or in  $C_1$  changes the notch frequency with no theoretical change in notch depth. In practice component imperfections, circuit strays etc. may require significant departures from bridge nominal resistances to reach balance, particularly at frequencies above 100Hz. If the circuit  $Q$  is large, A2 will saturate at the notch frequency unless  $V_{in}$  is kept low. At frequencies of the order of 1kHz or more the notch depth and  $Q$  are sensitive to resistance in series with  $C_2$ . A low-loss capacitor and low-contact-resistance potentiometer should then be used (avoid potentiometers with non-metallic contacts).

## Component variations

R:  $50\text{k}\Omega$  to  $1\text{M}\Omega$

$C_2$ :  $10\mu\text{F}$  to  $0.1\mu\text{F}$

$C_1$ :  $40\text{ nF}$  to  $100\text{pF}$ .

## Circuit modification

$R_1$  and  $R_2$  may be kept as fixed resistors if a variable capacitor for  $C_1$  is available.

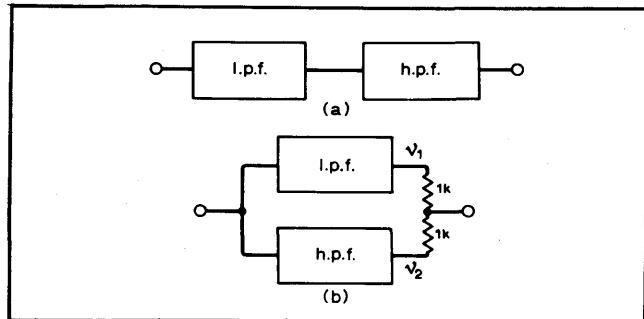
## Further reading

National Semiconductor LM307 Data Sheet.

## Cross references

Series 1, card 9.

## Compound filters



(a) provides the band-pass characteristic shown. Low and high-pass filters used are described in cards 4 and 5 with cut-off frequencies 10kHz and 2.35kHz respectively.  $f_1 = 5\text{kHz}$ ,  $f_2 = 1.9\text{kHz}$  and  $f_3 = 13\text{kHz}$ .

(b) provides the band-stop characteristic shown. In this case the low and high-pass filters had cut-off frequencies 0.76kHz and 12kHz respectively.

### Circuit description

The basic filter functions described may be combined in a variety of ways to produce alternative functions or to improve on existing ones. The two examples shown illustrate ways of

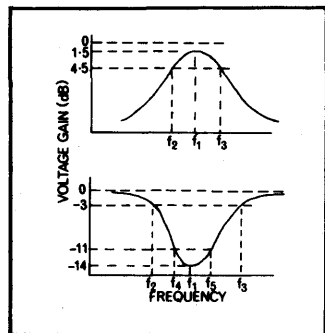
$$v_0 = (v_1 + v_2)/2,$$

$$f_1 = 3\text{kHz}, f_2 = 0.76\text{kHz},$$

$$f_3 = 12\text{kHz},$$

$$f_4 = 2.1\text{kHz and}$$

$$f_5 = 4.5\text{kHz}.$$



producing simple band-pass and band-stop filters from low-pass and high-pass sections. Cascading l.p. and h.p. filters with the l.p. filter cut-off frequency above that of the h.p. filter gives a pass-band lying between these two frequencies, and for second-order filters, an attenuation approaching 12dB/octave outside this range. Multi-order l.p. and h.p. filters may be used to increase this attenuation.

Conversely if the filter responses are added with the l.p. filter cut-off frequency being the lower the response is minimum between the cut-off frequencies with the depth of the attenuation dependent on the separation. Simple resistor summing, or a virtual earth summing amplifier may be used.

### Circuit modifications

One can cascade second-order low-pass and high-pass sections to achieve steeper sides to the stop and passbands. This only produces even-order filters with poor roll-off characteristics. Improved higher-order filters can easily be used (see ref.).

One can also cascade the bandstop characteristic with a notch filter to achieve precise nulling of a fixed frequency with reasonable attenuation on either side of the notch. Several notch filters tuned to slightly different frequencies may be used to obtain a bandstop characteristic (card 9). Several band-pass filters (card 1) can be similarly cascaded to give a good bandpass characteristic.

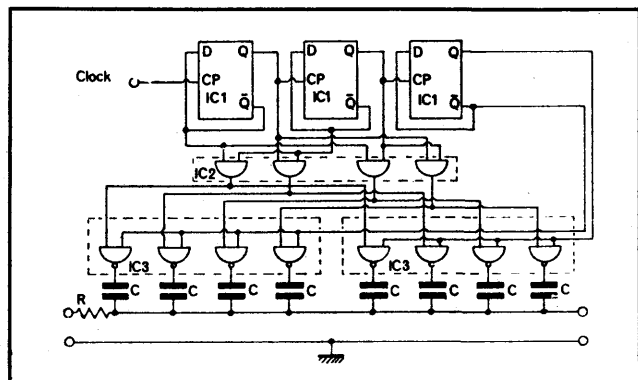
### Component changes

Any low-pass filter or high-pass filter may be used in the above configurations to provide the band-stop and band-pass characteristics.

### Further reading

Graeme, J. G., Tobey, G. E. & Huelsman, L. P. Operational amplifiers. McGraw-Hill p.320.

## N-path filter



## Typical performance

IC<sub>1</sub> 3 SN7474 D-type flip flop

IC<sub>2</sub> 2 off SN7400 connected as two-input AND gates

IC<sub>3</sub> SN7401 quadruple two-input NAND gates with open collector output.

R: 1kΩ ±1%

C: 0.1μF ±10%

f<sub>0</sub> = 14.9kHz Q = 30

Drive signal 600mV pk-pk.

Resonant frequency

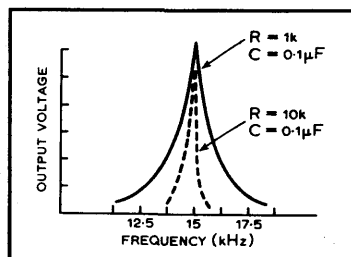
f<sub>0</sub> = clock frequency/N

Bandwidth: 2/NRC Hz,

N = number of low-pass filter sections = 8.

Peak on graph:

460mV pk-pk.



## Circuit description

The input signal is effectively switched between eight low-pass filter sections at a clock rate, eight times the required filter centre frequency. The output waveform comprises discrete levels approximating the input waveform. This stepped format may be removed by a low-Q bandpass filter.

The resonant frequency can be varied simply by changing the clock frequency thus giving a tunable bandpass filter with constant bandwidth.

## Component changes

Increasing R for same C values, increases Q e.g. R = 10kΩ ±10%, Q = 200.

IC<sub>1</sub> 1/3 MC7479; IC<sub>2</sub> MC3001; IC<sub>3</sub> MC7403.

## Circuit modifications

Due to the sampling action of this filter, responses are obtained at multiples of the clock frequency. If a 4-path filter with a fundamental resonant frequency f<sub>0</sub> is cascaded with a 3-path filter with a fundamental resonant frequency 2f<sub>0</sub>, a single bandpass characteristic is obtained, centred on 2f<sub>0</sub> (ref. 2).

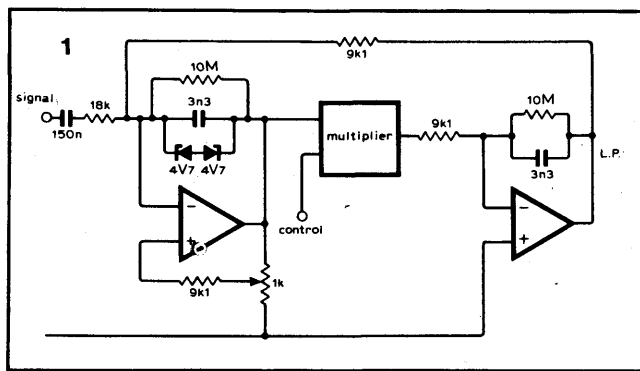
## Further reading

1. Broeker, B. Commutating filter techniques *Design Electronics*, 1971, pp.52-5.
2. Roberts, J. N-path filter techniques *Electron*, June 1972, pp.18-9.
3. Bruton, L. T. and Pederson, R. T. Time-multiplexed active filters, *IEEE Journal of Solid-State Circuits*, vol. SC-7, June 1972, pp.259-65.

### Basic active filters

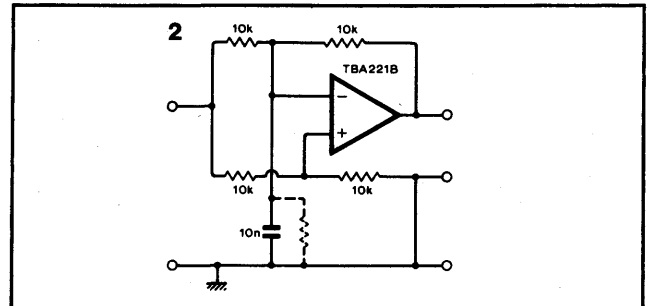
1. Voltage control of active filters is achieved using an analogue multiplier (see Audio circuits, set 5, for details). A loop consisting of a bandpass filter and an integrator has the signal coupled between them by the multiplier acting as a voltage controlled amplifier. The reference article also shows a square-law generator for shaping the frequency/control voltage characteristics. In this configuration the 1-k $\Omega$  potentiometer sets the circuit Q, and the direct control voltage the centre frequency. A low-pass output is also available. The zener diodes prevent overdrive from latching the circuit into a permanently switched state. The article also indicates how the circuit can be modified into an RC oscillator with stabilized amplitude, and a notch filter.

Orr, T. and Thomas, D. W. Electronic sound synthesizer, part 2. *Wireless World*, vol. 79, 1973, pp.429-34.



2. The limited amplitude-frequency response of operational amplifiers restricts the upper frequency at which defined filter characteristics can be obtained; it may result in instability/oscillation in some filters. As many op-amps, such as that shown here (and also 741, 307 etc), have a frequency response defined by a single dominant lag internal to the amplifier, then some second-order filters are possible using just one external capacitor. The reference article shows that the resulting transfer function is that of a band-pass filter with centre-frequency  $\approx 31\text{kHz}$ , centre-frequency gain  $\approx 115$  and  $Q \approx 11.5$ . Adding a resistor across the capacitor controls Q and gain without varying the centre frequency. The filter characteristics are supply and temperature dependent - the former can be turned to advantage in trimming centre frequency if variation in the other parameters is acceptable.

Radhakrishna Rao, K and Srinivasan, S. "Bandpass filter using the operational amplifier pole", *IEEE Journal of Solid-State Circuits*, June 1973, pp.245-6.



3. The design of higher-order filters can be based on cascaded first/second-order sections. Alternative approaches, economical of active and passive elements, use interconnected networks of impedance convertors (n.i.cs, gyrators etc) and passive networks. All the parameter values interact but the approach is very efficient and capable of very low sensitivity to component variation. The example shown uses blocks within dotted lines that act as impedance convertors to simulate lossy inductances, and results in a third-order low-pass elliptic filter with 1.0dB ripple in the passband, 30dB loss in the stopband and with a passband edge frequency of 3.4kHz. Other networks are shown in the article generating higher-order filters.

Wise, D. R. Active simulation of floating lossy inductances, *Proc. IEEE*, vol. 121, 1974 pp.85-7.

