

## Second Order Filters

**Second Order Filters** which are also referred to as VCVS filters, because the op-amp is used as a Voltage Controlled Voltage Source amplifier, are another important type of active filter design because along with the active first order RC filters we looked at previously, higher order filter circuits can be designed using them.

In this **Filters** section tutorials we have looked at both passive and active filter designs and have seen that first order filters can be easily converted into second order filters simply by using an additional RC network within the input or feedback path. Then we can define second order filters as simply being: “two 1st-order filters cascaded together with amplification”.

Most designs of second order filters are generally named after their inventor with the most common filter types being: *Butterworth*, *Chebyshev*, *Bessel* and *Sallen-Key*. All these types of filter designs are available as either: low pass filter, high pass filter, band pass filter and band stop (notch) filter configurations, and being second order filters, all have a 40-dB-per-decade roll-off.

The Sallen-Key filter design is one of the most widely known and popular 2nd order filter designs, requiring only a single operational amplifier for the gain control and four passive RC components to accomplish the tuning.

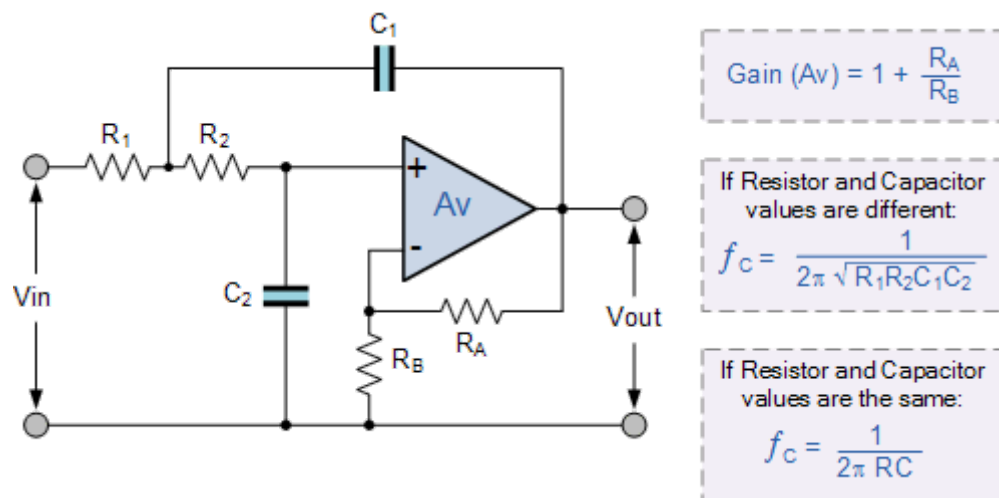
Most active filters consist of only op-amps, resistors, and capacitors with the cut-off point being achieved by the use of feedback eliminating the need for inductors as used in passive 1st-order filter circuits.

Second order (two-pole) active filters whether low pass or high pass, are important in Electronics because we can use them to design much higher order filters with very steep roll-off's and by cascading together first and second order filters, analogue filters with an  $n^{\text{th}}$  order value, either odd or even can be constructed up to any value, within reason.

## Second Order Low Pass Filter

Second order low pass filters are easy to design and are used extensively in many applications. The basic configuration for a Sallen-Key second order (two-pole) low pass filter is given as:

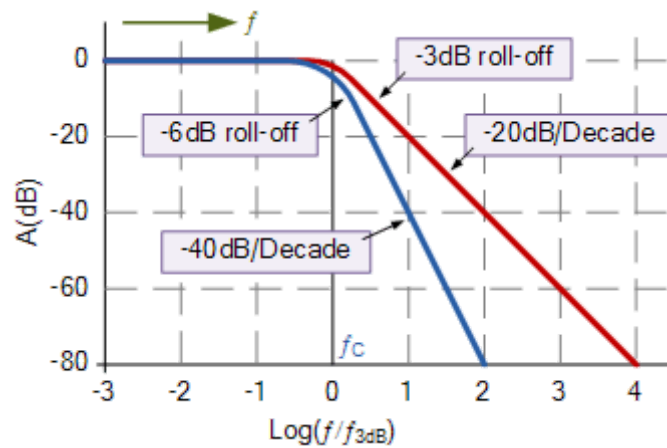
## Second Order Low Pass Filter



This second order low pass filter circuit has two RC networks,  $R_1 - C_1$  and  $R_2 - C_2$  which give the filter its frequency response properties. The filter design is based around a non-inverting op-amp configuration so the filter's gain,  $A$ , will always be greater than 1. Also the op-amp has a high input impedance which means that it can be easily cascaded with other active filter circuits to give more complex filter designs.

The normalised frequency response of the second order low pass filter is fixed by the RC network and is generally identical to that of the first order type. The main difference between a 1st and 2nd order low pass filter is that the stop band roll-off will be twice the 1st order filters at 40dB/decade (12dB/octave) as the operating frequency increases above the cut-off frequency  $f_c$ , point as shown.

## Normalised Low Pass Frequency Response



The frequency response bode plot above, is basically the same as that for a 1st-order filter. The difference this time is the steepness of the roll-off which is  $-40\text{dB/decade}$  in the stop band. However, second order filters can exhibit a variety of responses depending upon the circuits voltage magnification factor,  $Q$  at the the cut-off frequency point.

In active second order filters, the damping factor,  $\zeta$  (zeta), which is the inverse of  $Q$  is normally used. Both  $Q$  and  $\zeta$  are independently determined by the gain of the amplifier,  $A$  so as  $Q$  decreases the damping factor increases. In simple terms, a low pass filter will always be low pass in its nature but can exhibit a resonant peak in the vicinity of the cut-off frequency, that is the gain can increases rapidly due to resonance effects of the amplifiers gain.

Then  $Q$ , the quality factor, represents the “peakiness” of this resonance peak, that is its height and narrowness around the cut-off frequency point,  $f_c$ . But a filters gain also determines the amount of its feedback and therefore has a significant effect on the frequency response of the filter.

Generally to maintain stability, an active filters gain must not be more than 3 and is best expressed as:

## The Quality Factor, “Q”

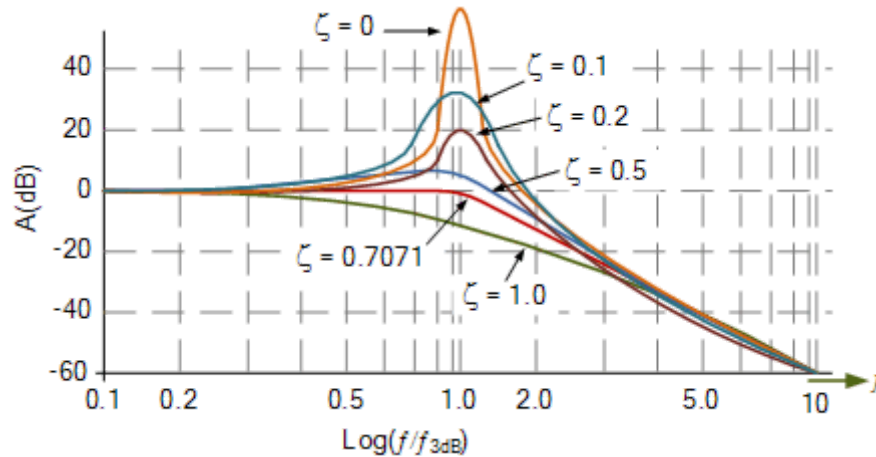
$$A = 3 - (2 \times \xi)$$

$$\text{Where: } \xi = \frac{3 - A}{2} = \frac{1}{2Q}$$

$$\therefore A = 3 - \frac{1}{Q}$$

Then we can see that the filter's gain,  $A$  for a non-inverting amplifier configuration must lie somewhere between 1 and 3 (the damping factor,  $\zeta$  between zero and 2). Therefore, higher values of  $Q$ , or lower values of  $\zeta$  gives a greater peak to the response and a faster initial roll-off rate as shown.

## Second Order Filter Amplitude Response



The amplitude response of the second order low pass filter varies for different values of damping factor,  $\zeta$ . When  $\zeta = 1.0$  or more (2 is the maximum) the filter becomes what is called “overdamped” with the frequency response showing a long flat curve. When  $\zeta = 0$ , the filter's output peaks sharply at the cut-off point resembling a sharp point at which the filter is said to be “underdamped”.

Then somewhere in between,  $\zeta = 0$  and  $\zeta = 2.0$ , there must be a point where the frequency response is of the correct value, and there is. This is when the filter is “critically damped” and occurs when  $\zeta = 0.7071$ .

One final note, when the amount of feedback reaches 4 or more, the filter begins to oscillate by itself at the cut-off frequency point due to the resonance effects, changing the filter into an oscillator. This effect is called self oscillation. Then for a low pass second order filter, both  $Q$  and  $\zeta$  play a critical role.

We can see from the normalised frequency response curves above for a 1st order filter (red line) that the pass band gain stays flat and level (called maximally flat) until the frequency response reaches the cut-off frequency point when:  $f = f_r$  and the gain of the filter reduces past its corner frequency at  $1/\sqrt{2}$ , or 0.7071 of its maximum value. This point is generally referred to as the filter's -3dB point and for a first order low pass filter the damping factor will be equal to one, ( $\zeta = 1$ ).

However, this -3dB cut-off point will be at a different frequency position for second order filters because of the steeper -40dB/decade roll-off rate (blue line). In other words, the corner frequency,  $f_r$  changes position as the order of the filter increases. Then to bring the second order filters -3dB point back to the same position as the 1st order filter's, we need to add a small amount of gain to the filter.

So for a Butterworth second order low pass filter design the amount of gain would be: **1.586**, for a Bessel second order filter design: **1.268**, and for a Chebyshev low pass design: **1.234**.

### Second Order Filter Example No1

A **Second Order Low Pass Filter** is to be design around a non-inverting op-amp with equal resistor and capacitor values in its cut-off frequency determining circuit. If the filters characteristics are given as:  $Q = 5$ , and  $f_c = 159\text{Hz}$ , design a suitable low pass filter and draw its frequency response.

Characteristics given:  $R_1 = R_2$ ,  $C_1 = C_2$ ,  $Q = 5$  and  $f_c = 159\text{Hz}$

From the circuit above we know that for equal resistances and capacitances, the cut-off frequency point,  $f_c$  is given as:

$$f_c = \frac{1}{2\pi RC}$$

Choosing a suitable value of say,  $10\text{k}\Omega$ 's for the resistors, the resulting capacitor value is calculated as:

$$f_c = \frac{1}{2\pi RC} \quad \therefore C = \frac{1}{2\pi R f_c}$$

$$C = \frac{1}{2\pi R f_c} = \frac{1}{2\pi \times 10\text{k}\Omega \times 159\text{Hz}}$$

$$\therefore C = 100\text{nF} \text{ or } 0.1\mu\text{F}$$

Then for a cut-off corner frequency of **159Hz**,  $R = 10\text{k}\Omega$  and  $C = 0.1\mu\text{F}$ .

with a value of  $Q = 5$ , the filters gain,  $A$  is calculated as:

$$Q = 5, \text{ and } A = 3 - \frac{1}{Q}$$

$$\therefore A = 3 - \frac{1}{5} = 3 - 0.2 = 2.8$$

We know from above that the gain of a non-inverting op-amp is given as:

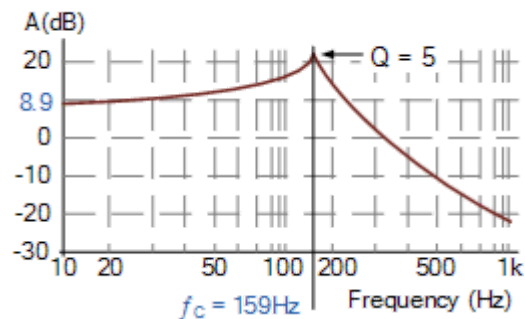
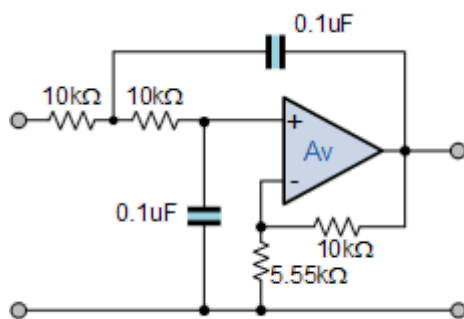
$$A = \frac{V_{out}}{V_{in}} = \left( 1 + \frac{R_A}{R_B} \right) = 2.8$$

Hence:  $\frac{R_A}{R_B} = 1.8$

$$\text{If } R_A = 10\text{k}\Omega, \text{ then } R_B = 5.55\text{k}\Omega$$

Therefore the final circuit for the second order low pass filter is given as:

## Second Order Low Pass Filter Circuit



We can see that the peakiness of the frequency response curve is quite sharp at the cut-off frequency due to the high quality factor value,  $Q = 5$ . At this point the gain of the filter is given as:  $Q \times A = 14$ , or about **+23dB**, a big difference from the calculated value of 2.8, (+8.9dB).

But many books, like the one on the right, tell us that the gain of the filter at the normalised cut-off frequency point, etc, etc, should be at the -3dB point. By lowering the value of Q significantly down to a value of **0.7071**, results in a gain of,  $A = 1.586$  and a frequency response which is maximally flat in the passband having an attenuation of -3dB at the cut-off point the same as for a second order butterworth filter response.

So far we have seen that **second order filters** can have their cut-off frequency point set at any desired value but can be varied away from this desired value by the damping factor,  $\zeta$ . Active filter designs enable the order of the filter to range up to any value, within reason, by cascading together filter sections.

In practice when designing  $n^{\text{th}}$ -order low pass filters it is desirable to set the cut-off frequency for the first-order section (if the order of the filter is odd), and set the damping factor and corresponding gain for each of the second order sections so that the correct overall response is obtained. To make the design of low pass filters easier to achieve, values of  $\zeta$ , Q and A are available in tabulated form for active filters as we will see in the



[Texas Instruments](#) Looking for the latest from TI?  
[Butterworth Filter](#) tutorial. Let's look at another example.

## Second Order Filter Example No2

Design a non-inverting second order Low Pass filter which will have the following filter characteristics:  $Q = 1$ , and  $f_c = 79.5\text{Hz}$ .

From above, the corner frequency,  $f_c$  of the filter is given as:

$$f_c = \frac{1}{2\pi RC}$$

Choosing a suitable value of  $1\text{k}\Omega$  for the filters resistors, then the resulting capacitor values are calculated as:

$$f_c = \frac{1}{2\pi RC} \quad \therefore C = \frac{1}{2\pi R f_c}$$

$$C = \frac{1}{2\pi R f_c} = \frac{1}{2\pi \times 1\text{k}\Omega \times 79.5\text{Hz}}$$

$$\therefore C = 2.0\mu\text{F}$$

Therefore, for a corner frequency of 79.5Hz, or 500 rad/s,  $R = 1k\Omega$  and  $C = 2.0\mu F$ .


With a value of  $Q = 1$  given, the filter's gain  $A$  is calculated as follows:

$$Q = \frac{1}{2\xi}, \quad \therefore \xi = \frac{1}{2Q} = \frac{1}{2 \times 1} = 0.5$$

$$\xi = 0.5 = \frac{3-A}{2}, \quad \therefore A = 3 - 2\xi = 2$$

The voltage gain for a non-inverting op-amp circuit was given previously as:

$$A = \frac{V_{out}}{V_{in}} = \left(1 + \frac{R_A}{R_B}\right) = 2$$

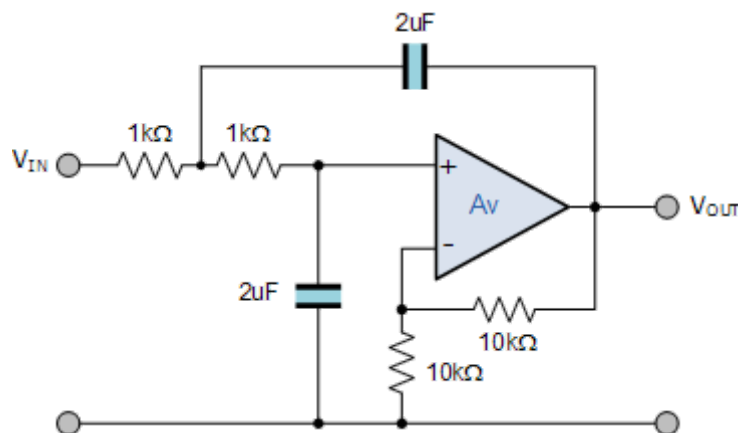
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Hence:  $\frac{R_A}{R_B} = 1$

If  $R_A = 10k\Omega$ , then  $R_B = 10k\Omega$

Therefore the second order low pass filter circuit which has a  $Q$  of 1, and a corner frequency of 79.5Hz is given as:

## Low Pass Filter Circuit

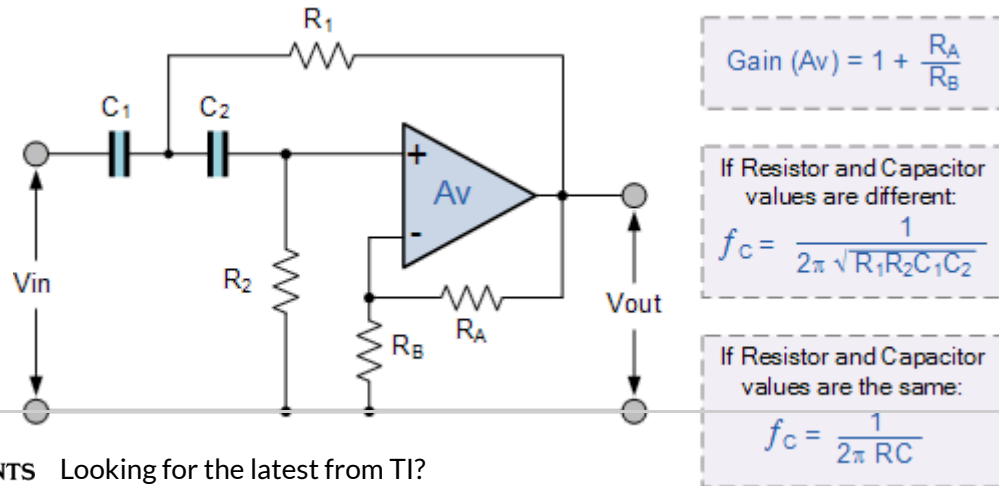


## Second Order High Pass Filter



There is very little difference between the second order low pass filter configuration and the second order high pass filter configuration, the only thing that has changed is the position of the resistors and capacitors as shown.

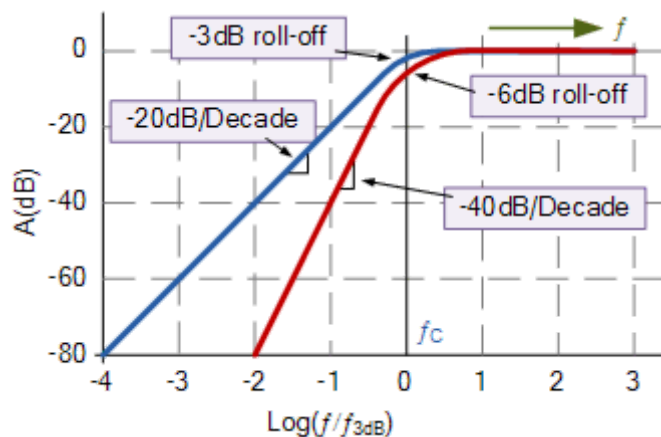
## Second Order High Pass Filter



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Since second order high pass and low pass filters are the same circuits except that the positions of the resistors and capacitors are interchanged, the design and frequency scaling procedures for the high pass filter are exactly the same as those for the previous low pass filter. Then the bode plot for a 2nd order high pass filter is therefore given as:

## Normalised High Pass Frequency Response



As with the previous low pass filter, the steepness of the roll-off in the stop band is  $-40\text{dB/decade}$ .

In the above two circuits, the value of the op-amp voltage gain, ( $A_v$ ) is set by the amplifiers feedback network. This only sets the gain for frequencies well within the pass band of the filter. We can choose to amplify the output and set this gain value by whatever amount is

suitable for our purpose and define this gain as a constant, K.

2nd order Sallen-Key filters are also referred to as positive feedback filters since the output feeds back into the positive terminal of the op-amp. This type of active filter design is popular because it requires only a single op-amp, thus making it relatively inexpensive.

## 11 Comments

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V V.S.V.Mani

It is always necessary to look back at these topics. I was recently checking a statement by a blogger regarding the product bandwidth\*rise time, and opined that for this product (which is 0.350 for first order), to be 0.5, the system has to become more sluggish and one has to decrease the zeta, when a

second order system is considered. in fact the Buterworth second order uields almost same bandwidth\*rise time product as first order LPF.

Posted on May 17th 2016 | 11:35 am

↩ Reply

V V.S.V.Mani

I should have said “increase” the zeta and not “decrease” the zeta in my earlier comment. regrets.

Posted on May 17th 2016 | 11:37 am

↩ Reply

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G Gberedio Duke

I need help on how to derive the gain for an active second order high pass filter in terms of R1, R2, C1, C2 and the open loop gain A of the OPAMP. Thanks

Posted on October 19th 2015 | 1:29 pm

↩ Reply

R Ravi Tejashree

design state variable filter that can serve a low pass butterwort filter if 3bd cutoff frequency is 1khz.  
2. design a second order butterwort cutoff frequency of 2khz pass band gain 2.

Posted on September 02nd 2015 | 8:26 am

↩ Reply

Q Qwerty James

sorry dude I don't no.....

Posted on November 25th 2015 | 5:16 pm

↩ Reply

D


# Davood Makivand

To whom it may concern;  
May I ask you send me samples of Low/Hi pass filter , band pass filter , .... that they are practical to build .  
Kindly  
Davood Makivand

Posted on March 03rd 2015 | 9:48 am

◀ Reply

# o overshoot

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Posted on February 02nd 2015 | 12:44 pm

◀ Reply

# I lefteris

Hi, I'm a little confused here...Is the damping factor equal to  $1/Q$  or  $1/2Q$ ? Because you say that  $\zeta$  is the reciprocal of  $Q$  and then after a while you set it equal to  $1/2Q$ . Also, in the active PBF tutorial you say that alpha peak is  $a=1/Q$ . Is alpha peak another name for the damping factor? And last but not least in the same article you give  $\xi=2a$  as the damping ratio. What is its relation with the damping factor  $\zeta$ ? Clear things up a little bit if you have the time. Thanks for the good tutorials.

Posted on September 26th 2014 | 7:29 pm

◀ Reply



# Wayne Storr

Second-order circuits are subject to a condition known as damping in which the value of  $\zeta$  determines if the filter is underdamped, overdamped or critically damped. Damping factor  $\zeta$  (zeta) is always equal to  $1/(2Q)$  and is a fractional value within the range of 0 to 1. This formula can also be transposed to give:  $2\zeta = 1/Q$ , or  $Q = 1/(2\zeta)$ , or relating to gain,  $(3-A)/2$ . Alpha ( $a$ ) is the filters damping ratio given as:  $1/Q$  or  $2\zeta$ .  $Q$  is the filters quality factor which describes the selectivity or peakiness of the filter and given as:  $Q = 1/(2\zeta)$ . The  $Q$  and  $\zeta$  are independently determined by the gain of the amplifier, ( $A$ ).

t      tauqeer shaikh

i have to design a filter circuit . but i cant understand how to take the values of Q and zeta to calculate ..or if there is a standard chart for it then forward it on my mail and last one is how to finalize the values of R and C ..and which op\_amp is to use..? reply fast ..and thanks for tutoriols

Posted on January 13th 2015 | 6:41 am

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 s. shammi  
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if a phase relattion is given then it would be more helpful

Posted on April 20th 2014 | 10:00 am

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