

# Lecture 6

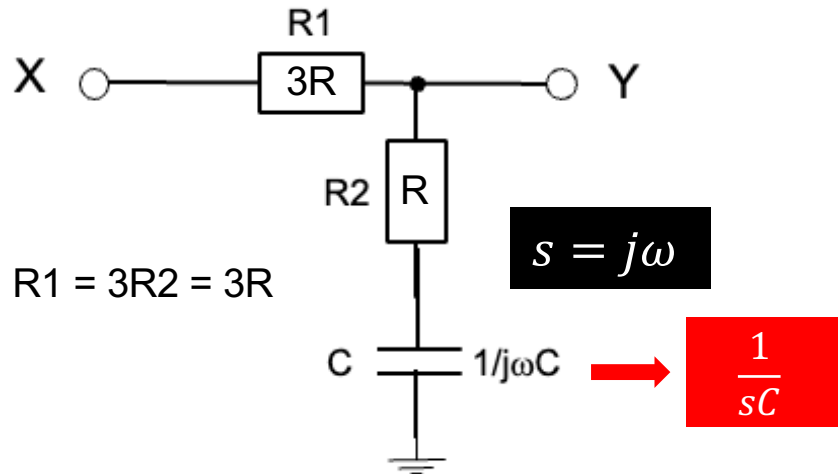
## Active Filters

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# Transfer Function of 1<sup>st</sup> order LP Filter

From Year 1 ADC Part 1 Lecture 11, slide 3.



- ❖ More general if use **complex frequency  $s$**  to represent the quantity  $j\omega$ .
- ❖ Covered in Signals and Systems module this term, and Control Systems next term.
- ❖ Express impedance of capacitor as  $\frac{1}{sC}$  instead of  $\frac{1}{j\omega C}$ .
- ❖ Capture both steady state (ac) and transient behaviour.

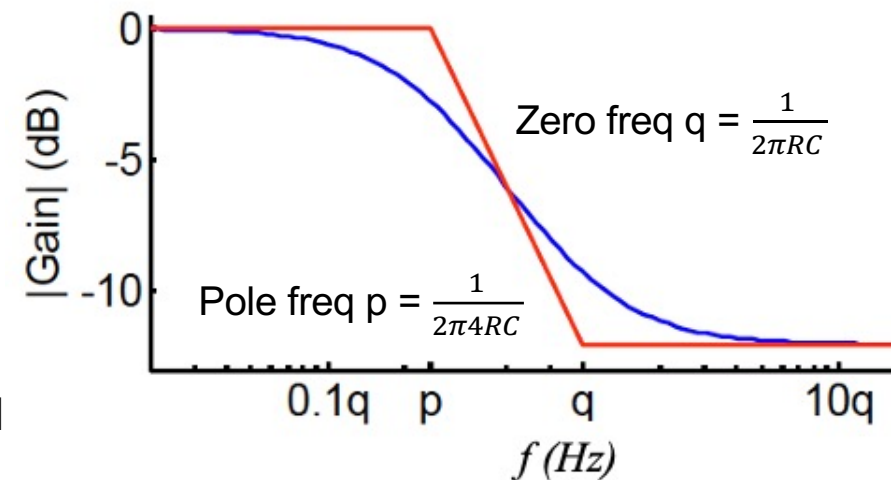
- ❖ Transfer function defined as:

$$H(s) = \frac{Y(s)}{X(s)} = \frac{R + 1/sC}{4R + 1/sC} = \frac{1 + sRC}{1 + 4sRC}$$

- ❖ Frequency response is calculated as

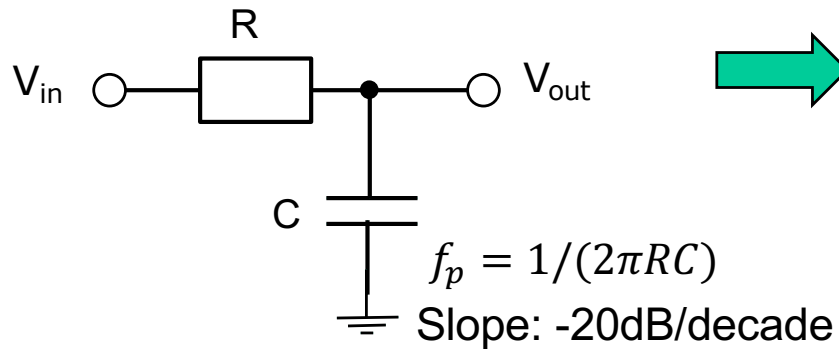
$$H(s)|_{s=j\omega} = \frac{Y(j\omega)}{X(j\omega)} = \frac{1 + j\omega RC}{1 + 4j\omega RC}$$

- ❖ Easier to perform algebra manipulation than using  $j\omega$ .
- ❖ Provides better intuitions on system behaviour.
- ❖ This simple filter is first-order low-pass with 1 pole and 1 zero.
- ❖ The break frequency occurs when real and imaginary parts are equal in numerator (zero freq) and denominator (pole freq).

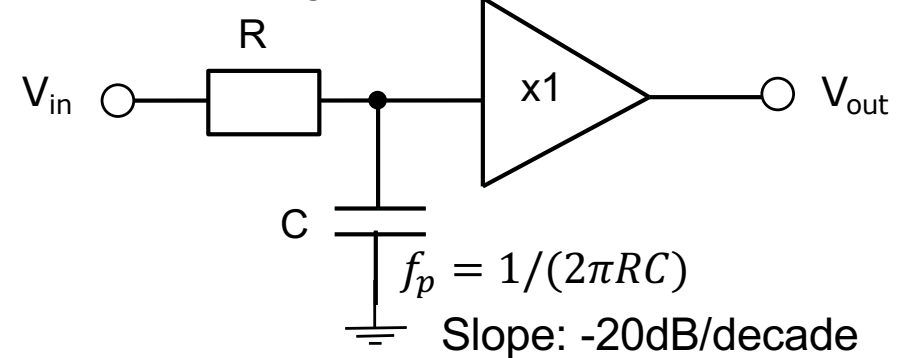


# 1<sup>st</sup> order Active Filter

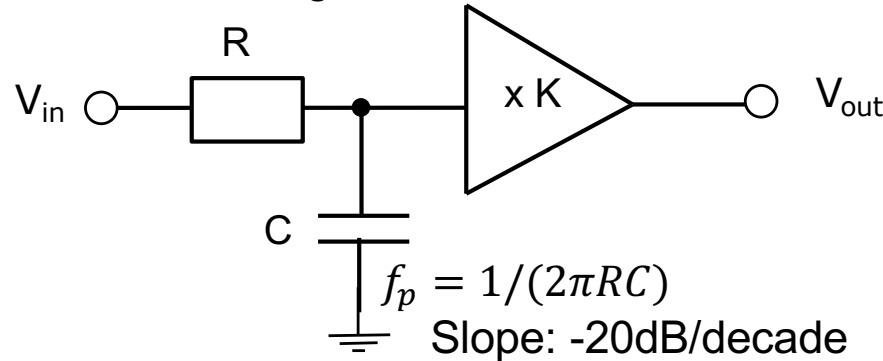
1<sup>st</sup> order passive filter  
with 1 pole



1<sup>st</sup> order active filter  
gain of 1



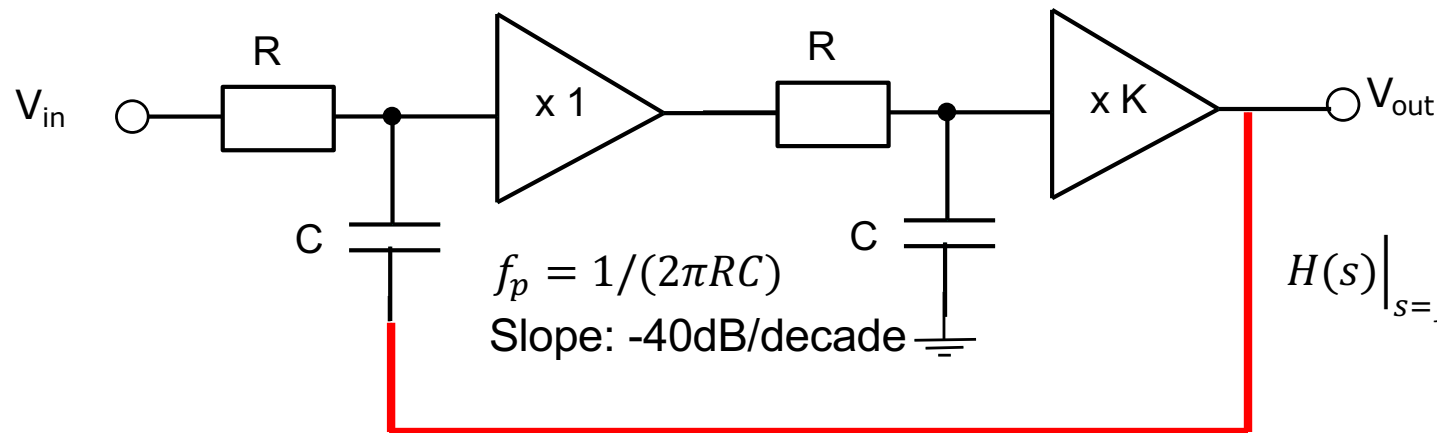
1<sup>st</sup> order active filter  
gain of K



$$H(s)|_{s=j\omega} = \frac{V_{out}(j\omega)}{V_{in}(j\omega)} = \frac{K}{1+j\omega RC}$$

# 2<sup>nd</sup> order Active Lowpass Filter

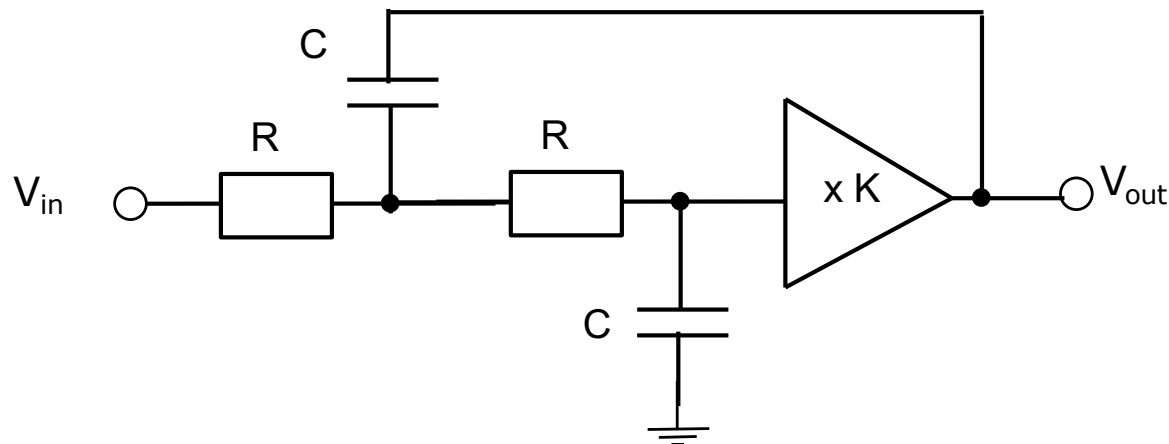
2<sup>nd</sup> order active filter with  
2 poles and dc gain of K



$f_p = 1/(2\pi RC)$   
Slope: -40dB/decade

$$H(s)\Big|_{s=j\omega} = \frac{V_{out}(j\omega)}{V_{in}(j\omega)} = \frac{K}{(1 + j\omega RC)^2}$$

2<sup>nd</sup> order Sallen-Key Filter (1955)

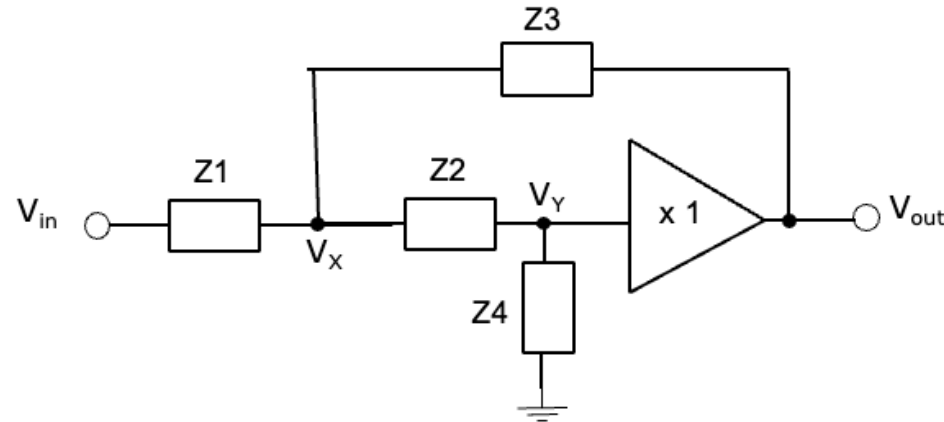


$f_p = 1/(2\pi RC)$

DC gain: K

Slope: -40dB/decade

# Sallen-Key Filter Topology



- ❖ Invented by R.P. Sallen and E.L. Key in 1955 using valves as active devices (!)
- ❖ Z1 to Z4 are arbitrary impedance from resistors, capacitors or inductors.
- ❖ Assume amplifier gain is 1 (can be generalised to K),  $V_Y = V_{out}$ .
- ❖ Apply KCL to  $V_x$  yields:

$$\frac{V_{in} - V_x}{Z_1} + \frac{V_{out} - V_x}{Z_3} + \frac{V_{out} - V_x}{Z_4} = 0$$

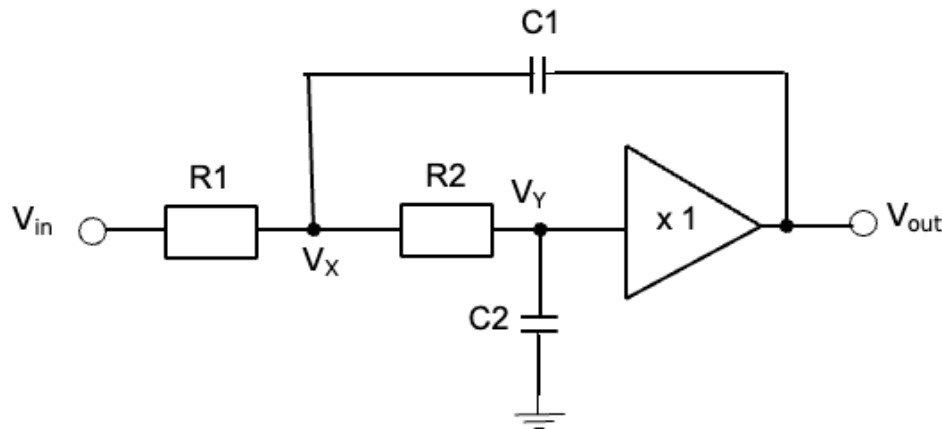
- ❖ Apply KCL at  $V_y$  yields:

$$V_x = V_{out} + \frac{Z_2}{Z_4} V_{out} = V_{out} \left( 1 + \frac{Z_2}{Z_4} \right)$$

- ❖ Combining the two gives a general transfer function equation:

$$\frac{V_{out}}{V_{in}} = \frac{Z_3 Z_4}{Z_1 Z_2 + Z_3 (Z_1 + Z_2) + Z_3 Z_4}$$

# 2<sup>nd</sup> order Sallen-Key Lowpass Filter



$$\frac{V_{out}}{V_{in}} = \frac{Z_3 Z_4}{Z_1 Z_2 + Z_3 (Z_1 + Z_2) + Z_3 Z_4}$$

$$Z_1 = R_1, \quad Z_2 = R_2$$

$$Z_3 = 1/sC_1, \quad Z_4 = 1/sC_2$$

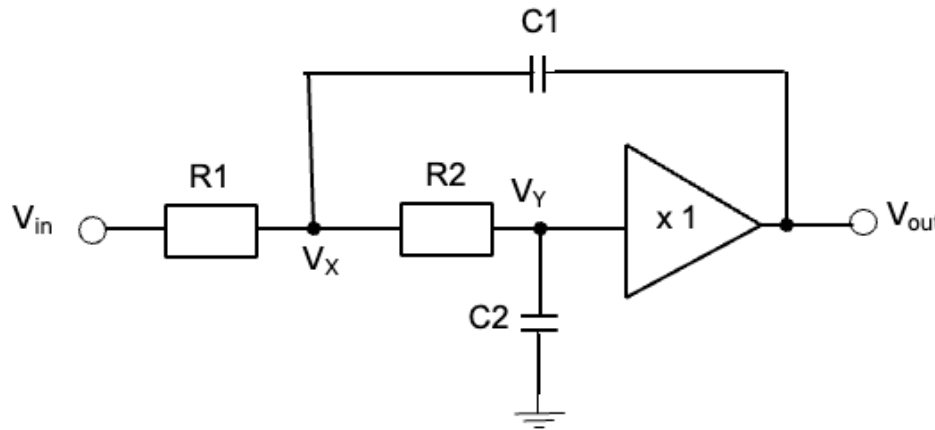
- ❖ Using the transfer function equation  $H(s)$  from previous slide:

$$H(s) = \frac{V_{out}(s)}{V_{in}(s)} = \frac{\frac{1}{s^2 C_1 C_2}}{R_1 R_2 + \frac{1}{s C_1} (R_1 + R_2) + \frac{1}{s^2 C_1 C_2}}$$

- ❖ Rearrange and put this in a standard form for a 2<sup>nd</sup> order lowpass system:

$$\frac{V_{out}(s)}{V_{in}(s)} = \frac{1}{1 + \mathbf{C_2(R_1 + R_2)} s + \mathbf{C_1 C_2 R_1 R_2} s^2}$$

# Significance of $\omega_0$ and $Q$ (1)



$$Z_1 = R_1, \quad Z_2 = R_2$$

$$Z_3 = 1/sC_1, \quad Z_4 = 1/sC_2$$

- ❖ Put this into the standard form of a 2<sup>nd</sup> order lowpass filter is:

$$\frac{V_{out}}{V_{in}} = \frac{1}{1 + \frac{1}{\omega_0 Q}s + \frac{1}{\omega_0^2}s^2} = \frac{1}{1 + \textcolor{red}{C_2(R_1+R_2)}s + \textcolor{green}{C_1C_2R_1R_2}s^2}$$

$Q$  is the quality factor  
 $\zeta$  is the damping ratio  
 $Q = \frac{1}{2\zeta}$

$$\frac{2\zeta}{\omega_0} = \frac{1}{\omega_0 Q}$$

$$\frac{1}{\omega_0^2}$$

$\omega_0$  is the cutoff frequency (rad/s)

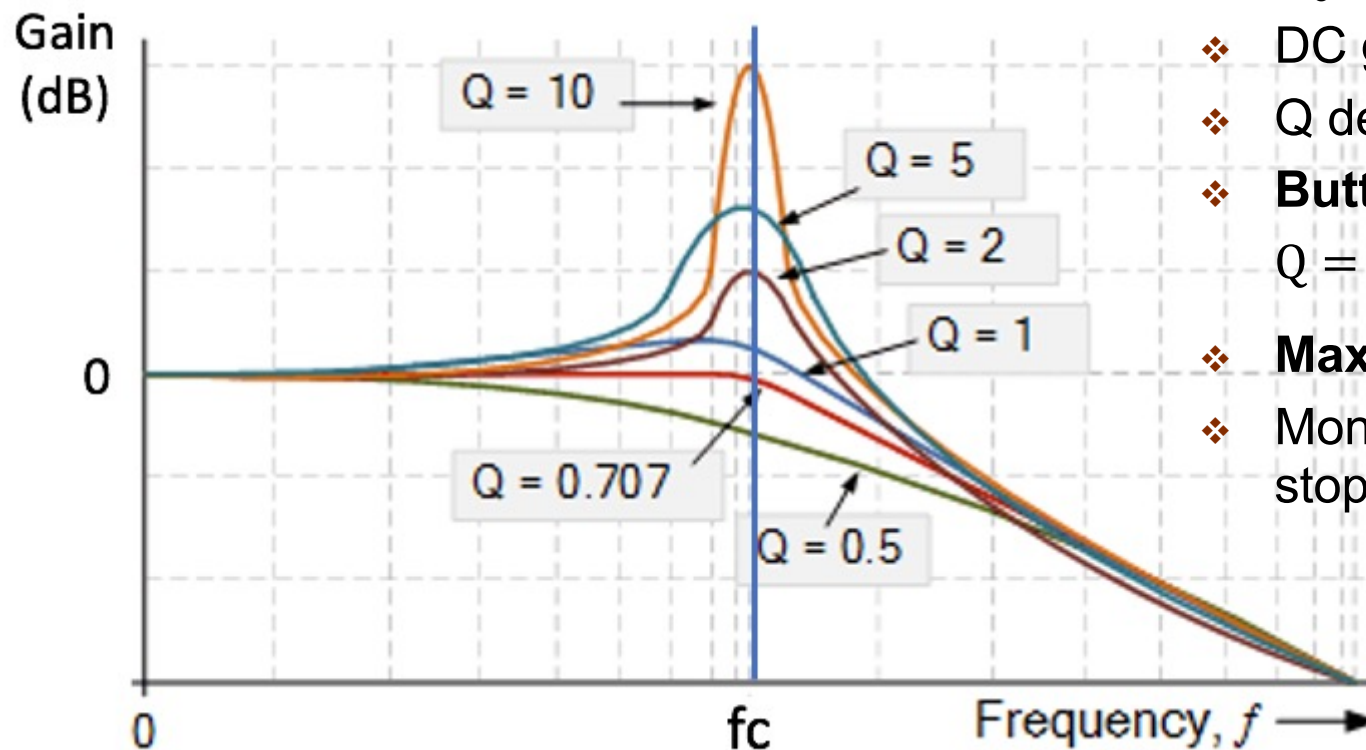
$$f_c = \frac{\omega_0}{2\pi} = \frac{1}{2\pi\sqrt{C_1C_2R_1R_2}} \text{ in Hz}$$

- ❖ Therefore, the cutoff frequency is:  $f_c = \frac{\omega_0}{2\pi} = \frac{1}{2\pi\sqrt{C_1C_2R_1R_2}} \text{ Hz}$
- ❖ The quality factor  $Q$  is:  $Q = \frac{\sqrt{C_1C_2R_1R_2}}{C_2(R_1+R_2)}$

# Significance of $\omega_0$ and $Q$ (2)

- ❖ Rewrite the transfer function as:

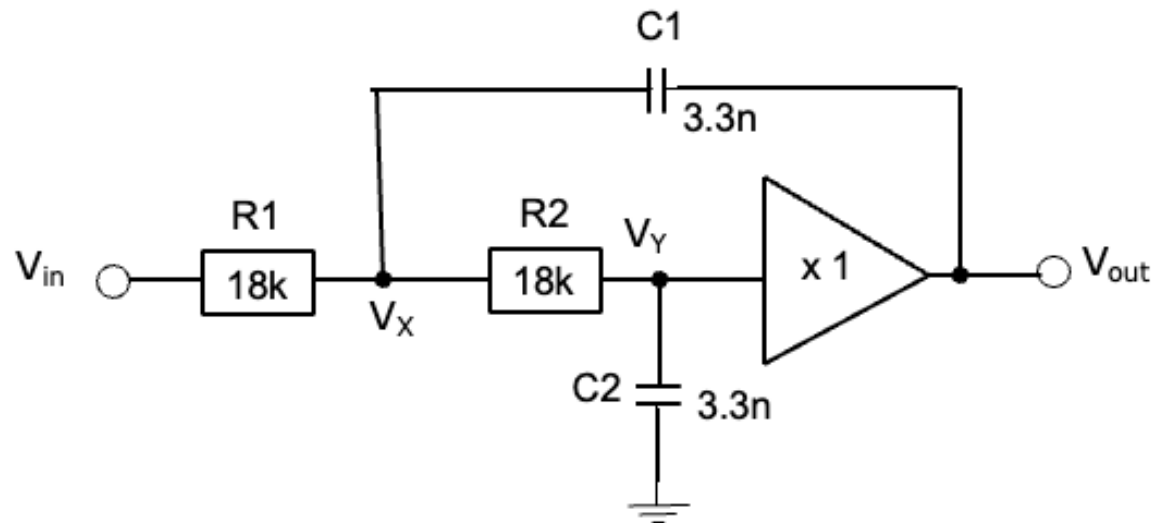
$$H(s) = \frac{V_{out}(s)}{V_{in}(s)} = \frac{1}{1 + \frac{1}{\omega_0 Q}s + \frac{1}{\omega_0^2}s^2} = \frac{\omega_0^2}{s^2 + \frac{\omega_0}{Q}s + \omega_0^2} \quad f_0 = \frac{\omega_0}{2\pi}$$



- ❖  $\omega_0$  is the **cut-off frequency** of filter.
- ❖ DC gain of filter is 1 (i.e.  $s=0$ ).
- ❖  $Q$  determine how 'peaky' the filter is.
- ❖ **Butterworth filter:**  $2\zeta = 1.414$ ,  $Q = \frac{1}{2\zeta} = 0.707$ .
- ❖ **Maximally flat** gain in passband
- ❖ Monotonically decreasing gain in stop band.

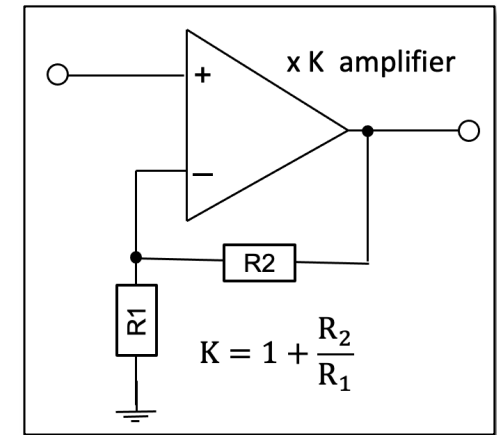
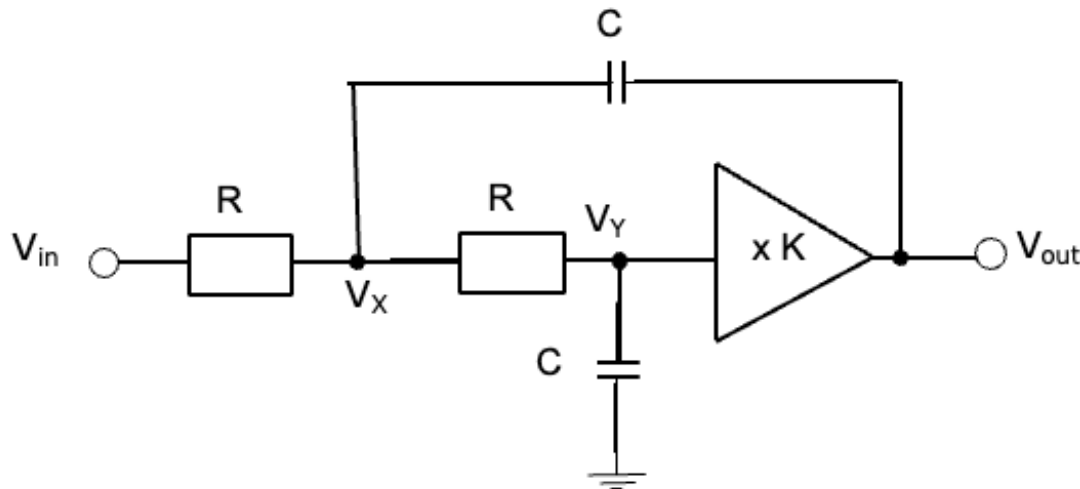


# A simple Sallen-Key filter (from Lab 2)



- ❖ Simplify by making  $R_1 = R_2 = R = 18k\Omega$ , and  $C_1 = C_2 = C = 3.3nF$ .
- ❖ The cutoff frequency is:  $f_c = \frac{1}{2\pi\sqrt{R_1 R_2 C_1 C_2}} = \frac{1}{2\pi RC} = 2.7\text{kHz}$  and
- ❖  $Q = \frac{\sqrt{R_1 R_2 C_1 C_2}}{C_2(R_1 + R_2)} = \frac{RC}{(C \times 2R)} = \frac{1}{2}$ .
- ❖ This is NOT a Butterworth filter because  $Q$  is not  $0.707$  or  $\frac{1}{\sqrt{2}}$ .

# Sallen-Key filter with gain = K



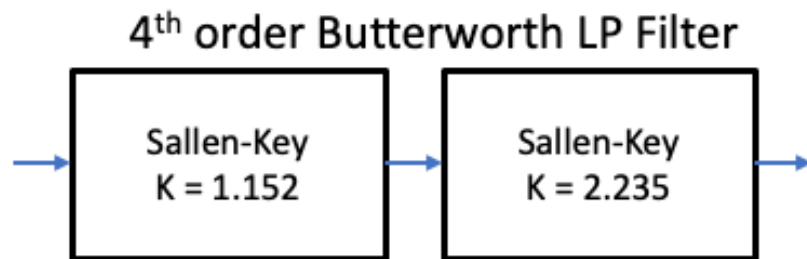
- ❖ Keep same R and C values, fix Q by changing gain of op-amp K
- ❖ Left as an exercise to proof:

$$H(s) = \frac{V_{out}(s)}{V_{in}(s)} = K \times \frac{\frac{1}{\omega_0 Q}}{1 + (3 - K)RC s + R^2 C^2 s^2}$$

- ❖ Therefore, cutoff frequency  $f_c$  is same as before:  $f_c = \frac{1}{2\pi\omega_0} = \frac{1}{2\pi RC}$ .
- ❖ And,  $Q = \frac{1}{\omega_0} \times \frac{1}{(3-K)RC} = \frac{1}{3-K}$ .
- ❖ Therefore, to get a Butterworth filter with this topology,  $Q = 0.707$ , and
- ❖  $K = 3 - \frac{1}{Q} = 1.586$ .

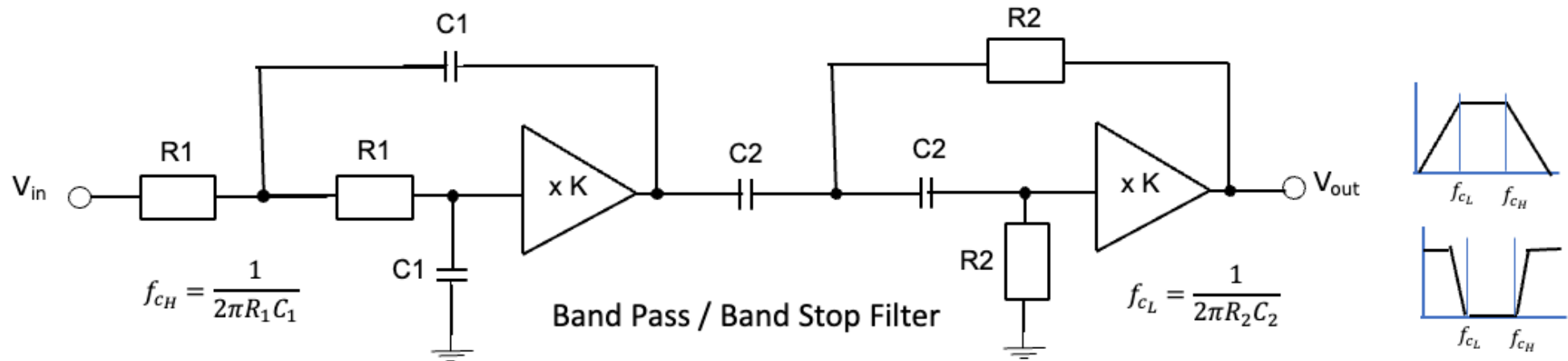
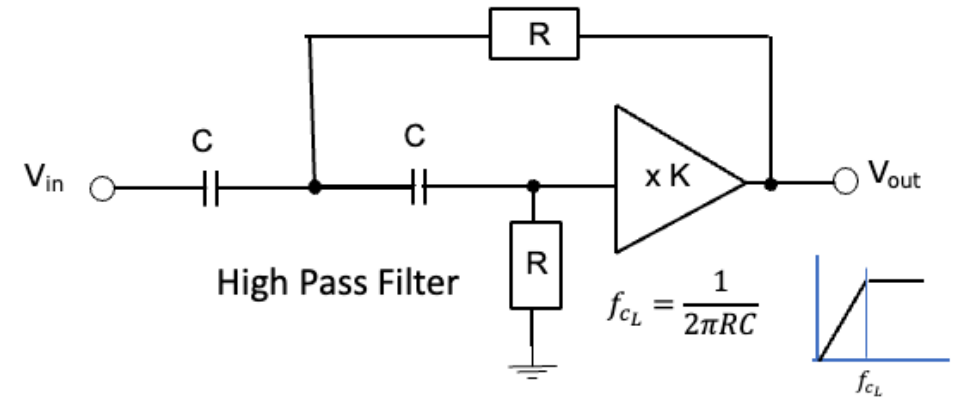
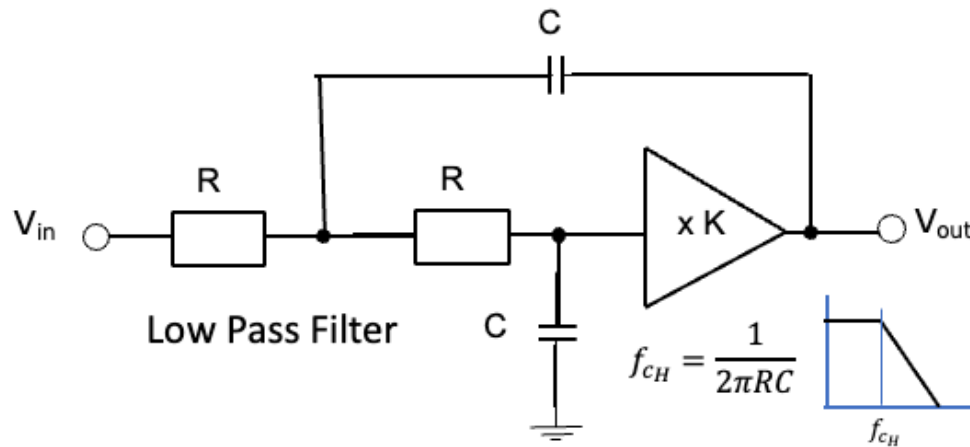
# General Procedure: Butterworth LP filter

1. Determine the required cutoff frequency  $f_c$ .
2. Calculate R and C product with:  $RC = \frac{1}{2\pi f_c}$ .
3. Pick a suitable value of C >> input capacitance of op-amp (say in nF range).
4. Calculate value of R to give the required cutoff frequency.
5. Determine order of filter depending on required attenuation rate. Filter attenuation rate is - 20 x n dB/decade, for an nth order filter.
6. Round n to the nearest high even number. You will need n/2 Sallen-Key filter stages.
7. Use the table below to design gain of each stage of the filter. For example, for a 4<sup>th</sup> order Butterworth filter, we need two Sallen-Key stages with gain of 1.152 followed by 2.235.
8. Choose resistors for op-amps feedback paths to provides specified gain values.



| ORDER n | Gain values K              |
|---------|----------------------------|
| 2       | 1.586                      |
| 4       | 1.152, 2.235               |
| 6       | 1.068, 1.586, 2.483        |
| 8       | 1.038, 1.337, 1.889, 2.610 |

# Other Sallen-Key filter circuits



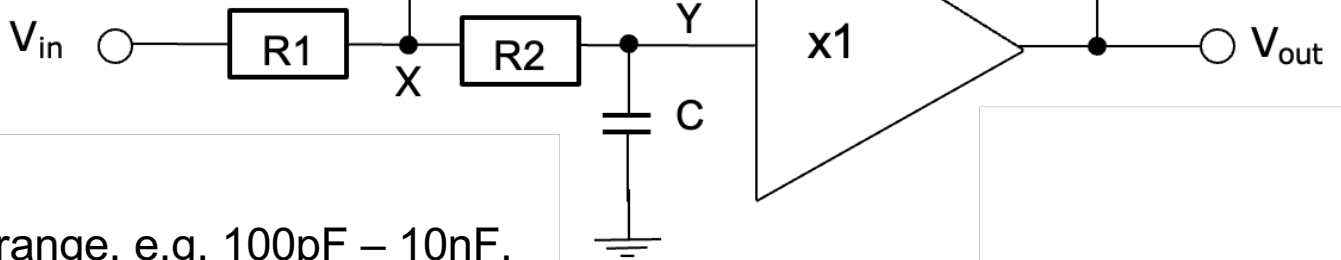
# Using different values for R1 and R2

- ❖ Revisit transfer function of filter from slide 7::

$$\frac{V_{out}}{V_{in}} = \frac{1}{1 + \frac{1}{\omega_0 Q} s + \frac{1}{\omega_0^2} s^2} = \frac{1}{1 + \textcolor{red}{C}(\textcolor{red}{R}_1 + \textcolor{red}{R}_2) s + \textcolor{green}{C}^2 \textcolor{green}{R}_1 \textcolor{green}{R}_2 s^2}$$

$$Q = \frac{\sqrt{R_1 R_2}}{R_1 + R_2}$$

$$\omega_0 = \frac{1}{C \sqrt{R_1 R_2}} \text{ Hz}$$



- ❖ Design step:
  - Choose C in a reasonable range, e.g. 100pF – 10nF.
- ❖ Write down eq.1 for a given cutoff frequency.
- ❖ Write down eq. 2 for a given Q value.
- ❖ Solve the two equations for R1 and R2 values.