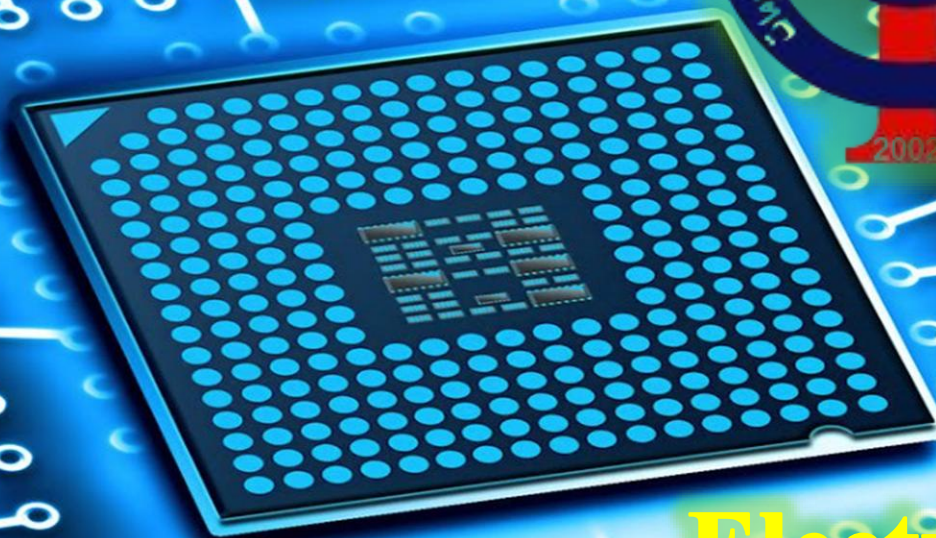




College of  
Information  
Engineering

2002



# Electronics

Lect. #14

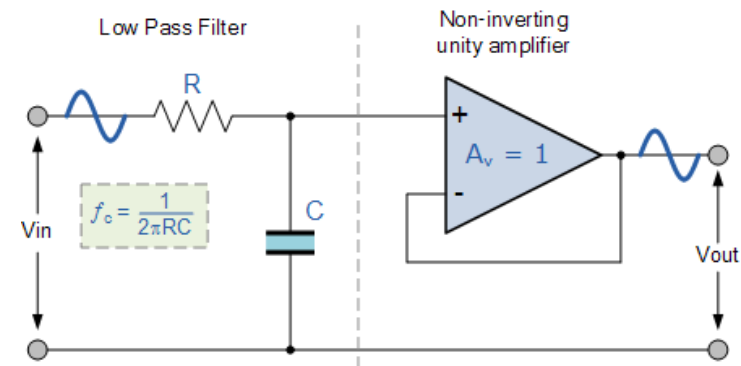
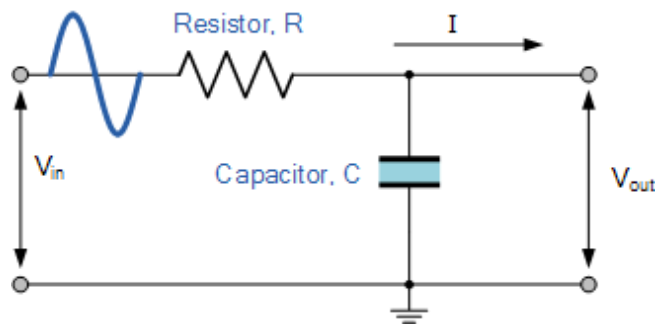
Lect. Hamzah M. Marhoon

## Filters

- A filter is an **electronic** circuit or system that **selects** specific frequencies of a signal and **attenuates** (reduces) others. In other words, filters **allow** certain frequency ranges to pass while **blocking** unwanted frequencies, **shaping** the signal according to the **required** application.
- Filters are used in almost every electronic and communication system. The main applications, can be summarized as follows:
  - **Noise Reduction:** Removes **unwanted** high-frequency or low-frequency noise from signals.
  - **Communication Systems:** Selects **desired** channels and rejects interference.
  - **Power Supplies:** Smooths DC output by removing AC ripple.
  - **Medical Instruments:** Filters ECG/EEG signals to remove noise.
  - **Digital Signal Processing:** Shapes signals in image, audio, and speech processing.
  - **Radio Receivers:** Allow only a specific frequency band to pass.

## Filters Categories

- **Passive filter:** is an electronic network constructed exclusively from **passive components** such as **resistors**, **inductors**, and **capacitors**. It operates without any **external power** source and relies just on the inherent **impedance characteristics** of its components to **shape** or **control** the frequency content of a signal. Passive filters cannot provide **amplification**; therefore, their **output** amplitude is always **less** than or **equal** to the **input**.
- **Active filter:** is a frequency-selective electronic circuit that incorporates active **components**, typically operational amplifiers, along with **resistors** and **capacitors**. Active filters require an **external power supply** and can provide **signal amplification**, improved selectivity, and greater control over filter parameters.



## Low-Pass Filter

- **Low-Pass Filter:** is a circuit that allows low frequencies to pass through and blocks or reduces high frequencies.
- The **passband** is the range of frequencies that can pass through the filter with almost no loss.
- The critical (**cutoff**) frequency,  $f_c$  is the point where the output drops to 70.7% (-3 dB) of the original signal. This marks the end of the passband.
  - The -3 dB is a **standard** engineering criterion used to define the **cutoff** frequency of a filter and to calculate bandwidth.
  - Because at -3 dB the power becomes exactly half, which is a natural physical reference.
  - It gives a clear and uniform way to define bandwidth and filter limits.

$$P \propto V^2$$

$$\frac{1}{\sqrt{2}} = 0.707$$

$$(0.707)^2 = 0.5$$



$$\text{dB} = 10 \log \left( \frac{P_{out}}{P_{in}} \right)$$

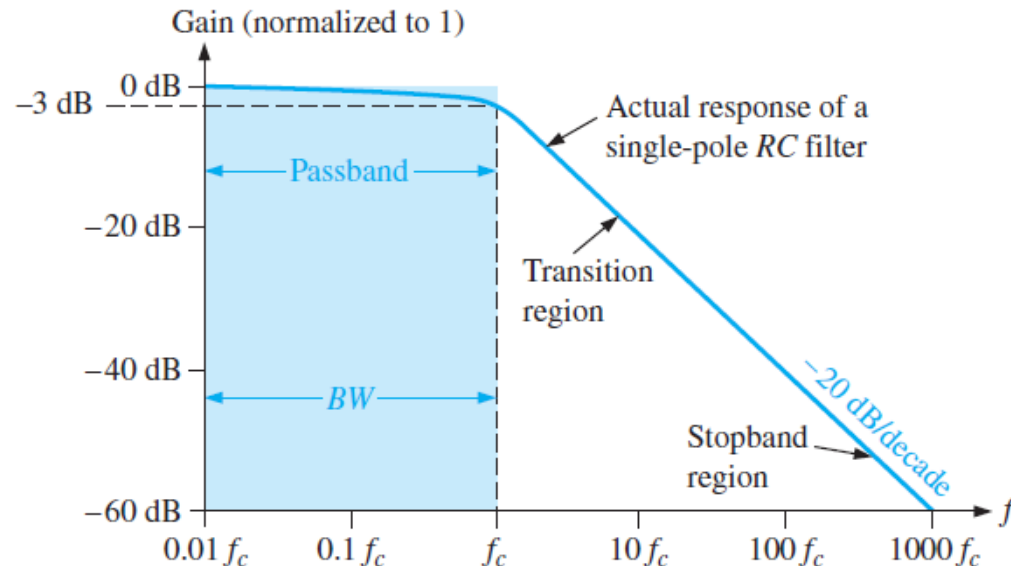
$$\text{dB} = 20 \log \left( \frac{V_{out}}{V_{in}} \right)$$

Plot Type	Cutoff Definition
Gain in dB vs frequency	Look for -3 dB
Gain (voltage ratio) vs frequency	Look for 0.707 × max gain
Power gain vs frequency	Look for 0.5 × max power

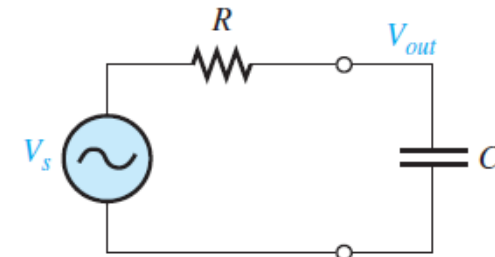
# Low-Pass Filter

- After the passband, the signal enters the **transition region**, where the filter starts **reducing** the signal.
- Beyond that is the **stopband**, where the filter **strongly** blocks the frequencies.
- A low-pass filter lets low frequencies (from DC up to  $f_c$ ) pass and reduces high frequencies.
- In an **ideal** low-pass filter, everything below  $f_c$  passes perfectly, and everything above it is completely blocked, this is called a brick-wall response.
- The **bandwidth** of an ideal low-pass filter is simply the cutoff frequency  $f_c$ .

Frequency	Increase	Attenuation
$f_c = 1 \text{ kHz}$	—	-3 dB
$10 \times f_c = 10 \text{ kHz}$	1 decade	-20 dB
$100 \times f_c = 100 \text{ kHz}$	2 decades	-40 dB
$1000 \times f_c = 1 \text{ MHz}$	3 decades	-60 dB



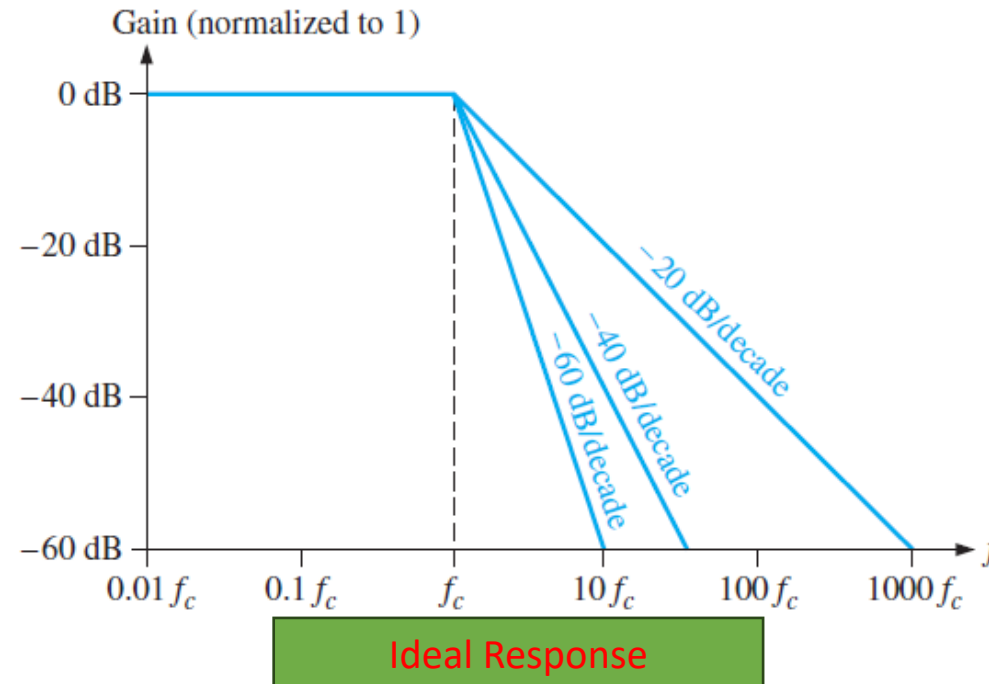
(a) Comparison of an ideal low-pass filter response (blue area) with actual response. Although not shown on log scale, response extends down to  $f_c = 0$ .



(b) Basic low-pass circuit

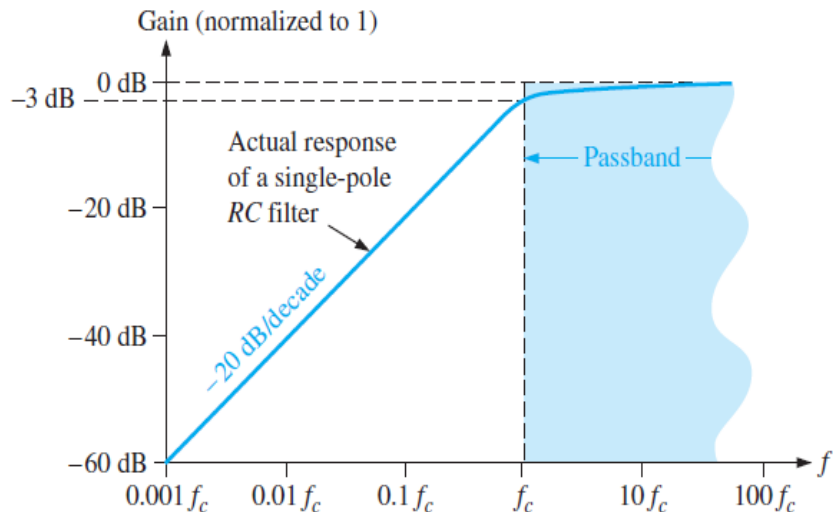
## Low-Pass Filter

- A decade in electronics means, A  $10\times$  (ten-fold) increase or decrease in frequency, for example:
  - From 1 kHz to 10 kHz → one decade
  - From 10 kHz to 100 kHz → one decade
  - From 100 kHz to 1 MHz → one decade
  - From 2 MHz to 20 MHz → one decade
- The  $-20$  dB/decade roll-off means for every  $10\times$  increase in frequency, the output of the filter decreases by 20 dB.

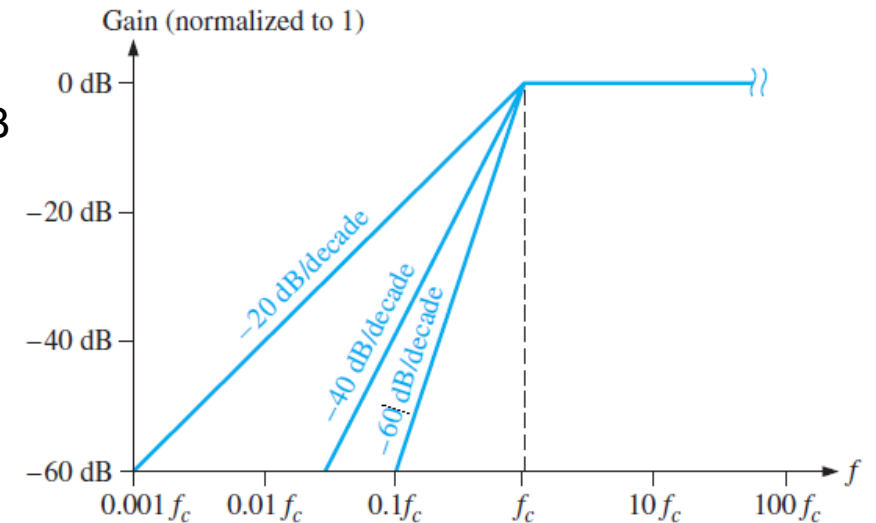
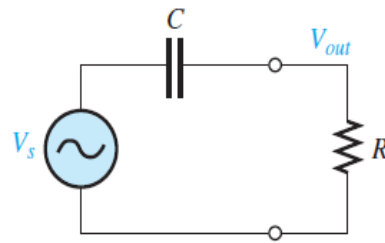


# High-Pass Filter

- A high-pass filter blocks or attenuates low frequencies and passes high frequencies.
- At frequencies below  $f_c$ , the filter provides strong attenuation.
- At frequencies above  $f_c$ , the filter allows signals to pass.
- The **ideal high-pass filter** would have a **sharp**, instantaneous **cutoff** at  $f_c$ , but this is not possible in real circuits.
- In reality, the filter response increases **gradually**, not **instantly**.



- At  $0.1 f_c$ , attenuation  $\approx -20$  dB
- At  $0.01 f_c$ , attenuation  $\approx -40$  dB
- At  $0.001 f_c$ , attenuation  $\approx -60$  dB



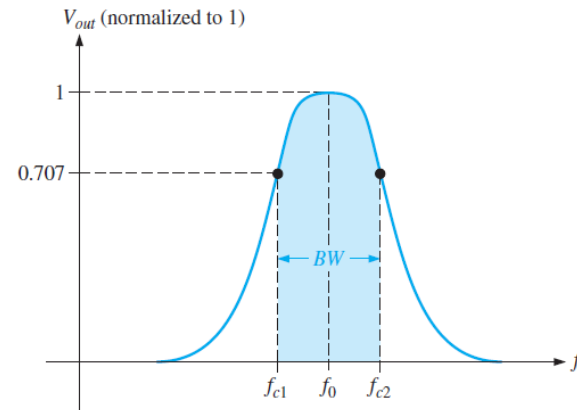
(a) Comparison of an ideal high-pass filter response (blue area) with actual response

(b) Basic high-pass circuit

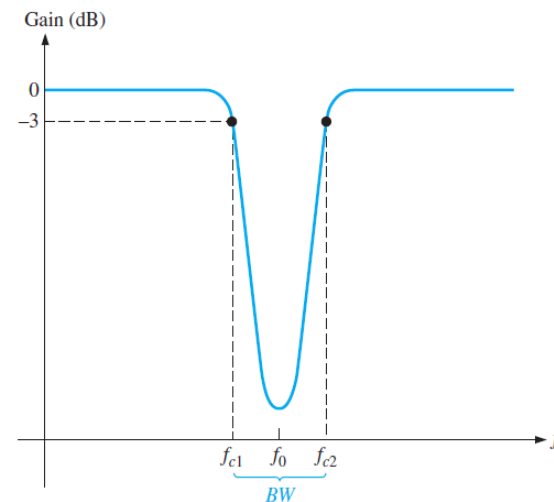
Ideal Response

## Band-Pass and Band-Stop Filters

- A band-pass filter is an electronic circuit that **passes** only the **frequencies** within a **specific range**, called the passband, and attenuates all frequencies below and above that range.

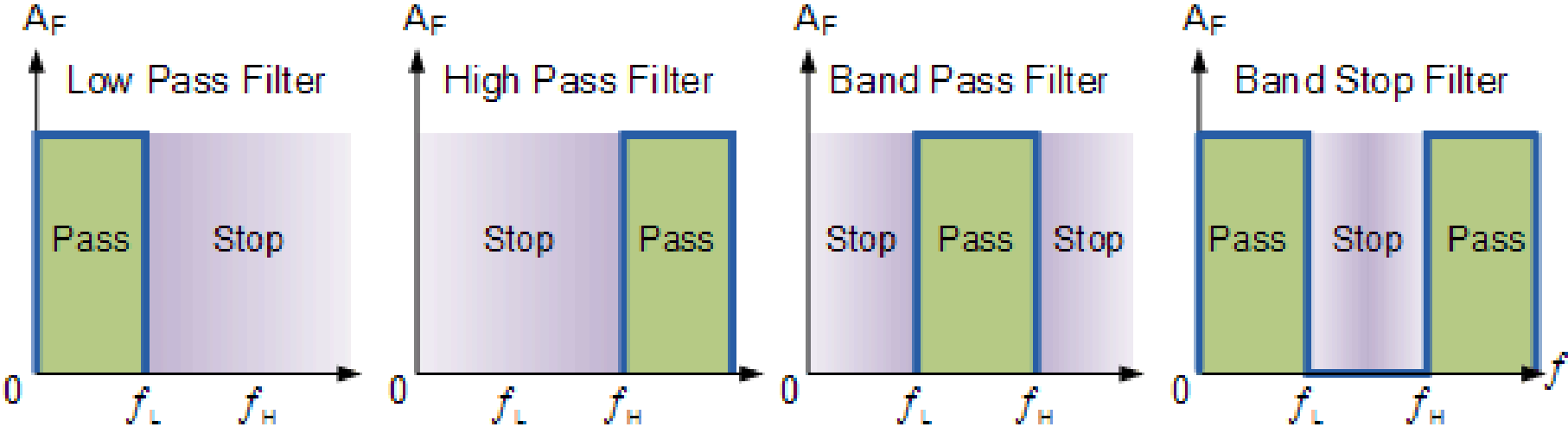


- A band-stop filter (also called a band-reject or **notch filter**) is an electronic circuit that attenuates or rejects a specific band of frequencies while allowing frequencies below and above that band to pass with little or no attenuation.



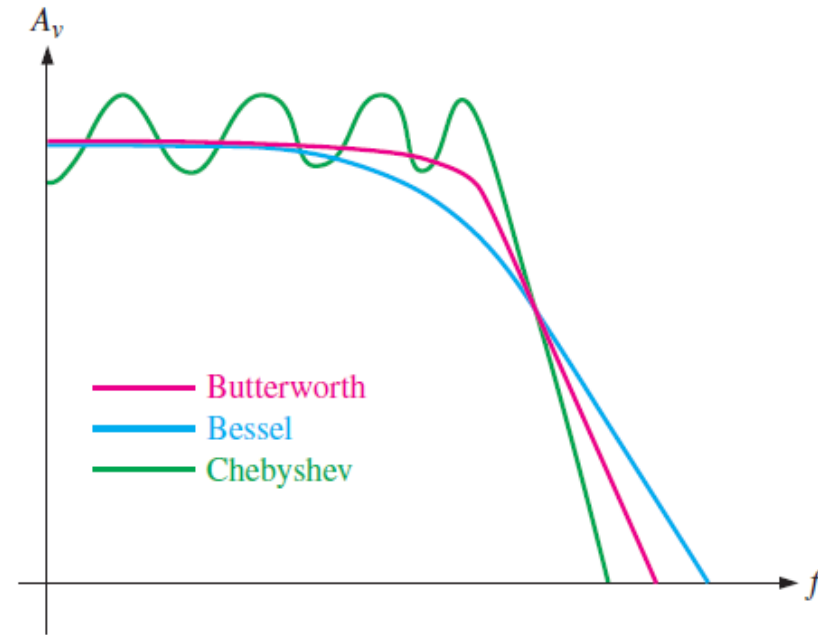
# Band-Pass and Band-Stop Filters

Ideal Filters



## Filter response characteristics

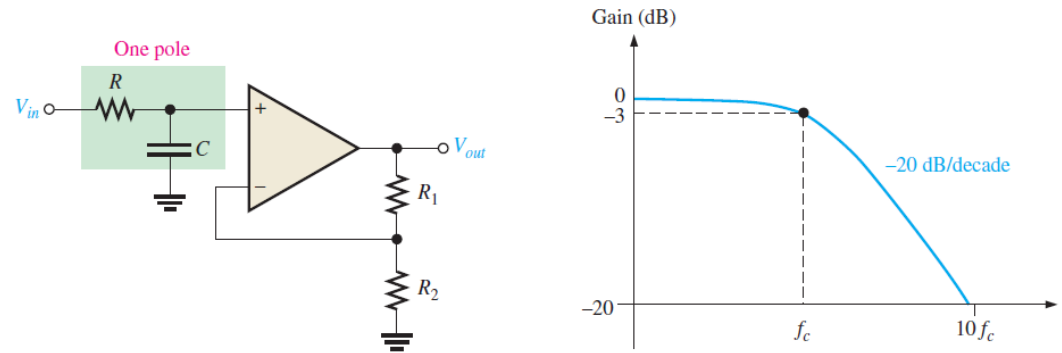
- The **Butterworth** characteristic provides a **maximally flat** amplitude response in the passband, meaning there is **no ripple** and the gain is very smooth up to the cutoff frequency. Its roll-off rate is  $-20$  dB/decade per pole.
- The **Bessel** characteristic provides the most **linear phase response** and therefore the best time-domain behavior. It minimizes signal distortion, preserves waveform shape, and avoids overshoot or ringing. Its roll-off is slower than Butterworth or Chebyshev.
- The **Chebyshev** characteristic achieves a **faster** roll-off by allowing ripple either in the passband (Type I) or stopband (Type II).



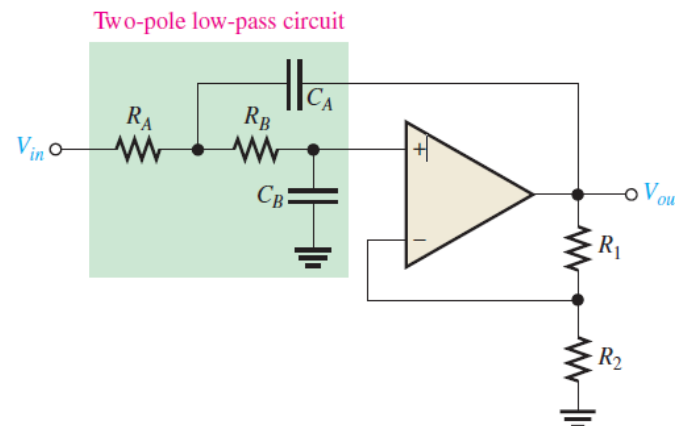
Filter Type	Passband	Roll-Off	Phase Response	Notes
Butterworth	Flat (no ripple)	Medium	Moderate	Most common, smooth response
Bessel	Flat	Slow (shallow)	<b>Best</b> (linear phase)	Best for preserving waveform shape
Chebyshev	Ripple	<b>Fastest</b>	Worst $t + \Delta t$	Sharp cutoff, but wavy response

## Active Low-Pass Filters

- A single-pole (1<sup>st</sup> Order) active low-pass filter is a filter that uses a single RC network to establish a low-pass cutoff frequency and an operational amplifier to provide voltage gain in the passband. It has one pole, which produces a roll-off of -20 dB per decade beyond the cutoff frequency. The op-amp is typically configured as a non-inverting amplifier, and its closed-loop gain is determined by the feedback resistors.



- A Sallen-Key low-pass filter is a widely used second-order (two-pole) active filter configuration that uses an operational amplifier and two RC networks to achieve sharper frequency selectivity, it provides a -40 dB/decade roll-off above the cutoff frequency, which is twice as steep as a single-pole filter.



$$f_c = \frac{1}{2\pi\sqrt{R_A R_B C_A C_B}}$$

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

For:

$$R_A = R_A = R$$

$$C_A = C_B = C$$

## Example #1:

Determine the critical (cutoff) frequency  $f_c$  of the Sallen–Key low-pass filter shown in the figure below. Then choose the value of  $R_1$  to obtain an approximate Butterworth response.

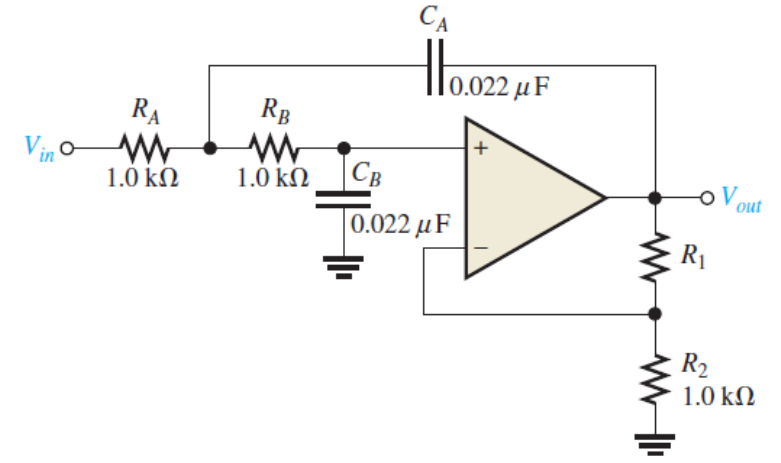
□ For the Sallen–Key low-pass filter in the shown figure:

- $R_A = R_B = R_2 = 2.2 \text{ k}\Omega$
- $C_A = C_B = 0.01 \text{ }\mu\text{F}$

$$f_c = \frac{1}{2\pi RC}$$
$$f_c = \frac{1}{2\pi(2.2 \times 10^3)(0.01 \times 10^{-6})}$$
$$f_c \approx 7.23 \text{ kHz}$$

□ For a Butterworth characteristic with this Sallen–Key configuration, the design condition is:

$$\frac{R_1}{R_2} = 0.586$$
$$R_1 = 586 \text{ }\Omega$$





**Thanks for listening...**