

Active filters & how they work

Ref: } Text, various places; Abramowitz & Stegun, Hbk. of Math. Funct., Ch. 29
 Millman, Microelectronics, Sec. 16-6
 Kowalski, Nuclear Electronics, Appendix 8

Filters are conveniently analyzed using the Laplace transform techniques mentioned earlier. One deals with the circuit in the Laplace transformed complex frequency space using circuit analysis which is closely analogous to what was done before using complex impedance. In particular, we get the following:

Further-
more

$$\hat{f}(s) = \int_0^{\infty} e^{-st} f(t) dt$$

{ where $\hat{f}(s)$ is the transformed quantity.

$$\frac{\hat{f}(s)}{c} = \frac{f(t)}{c \delta(t)}$$

$s = \sigma + j\omega$

$c \theta(0) \rightarrow$ const. times unit step fcn. (written here explicitly)

$$s \hat{f}(s) - f(0) \quad \frac{df}{dt} \quad (+0 \text{ means approached from above})$$

$$\frac{1}{s} \hat{f}(s) \quad \int_0^t f(\tau) d\tau$$

$$\frac{b}{(s+a)^2 + b^2} \quad e^{-at} \sin bt$$

$$\frac{s+a}{(s+a)^2 + b^2} \quad e^{-at} \cos bt$$

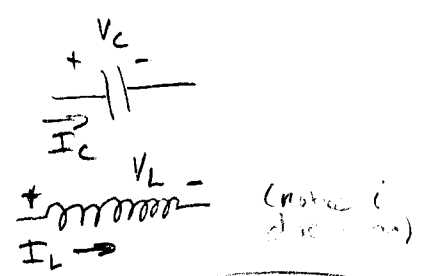
$$\frac{1}{s+a} \quad e^{-at}$$

poles of $\hat{f}(s)$ in s-space determine the behavior of $f(t)$.
 The δ -function response looks like $f(t)$ up to a constant and a $\theta(0)$.
 More poles? use superposition.

$$\hat{V}_R(s) = R \hat{I}_R(s) \quad V_R(t) = R I_R(t)$$

$$* \hat{V}_C(s) = \frac{1}{sC} \hat{I}_C(s) \quad V_C(t) = \frac{1}{C} \int_0^t I_C(\tau) d\tau$$

$$** \hat{V}_L(s) = s \hat{I}_L(s) \quad V_L(t) = L \frac{d}{dt} I_L(t)$$

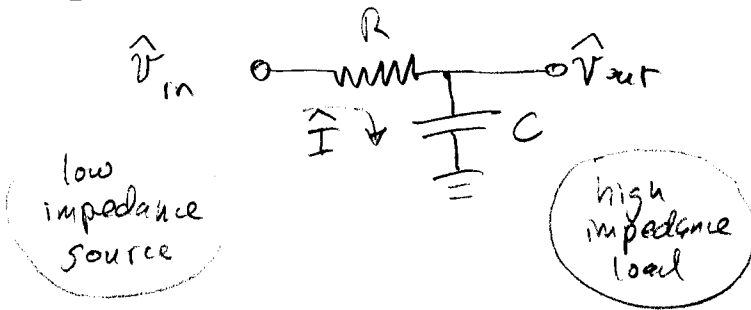


*,** Note: These require $\begin{cases} V_C = 0 \text{ when } t=0 (*) \\ I_L = 0 \text{ " " } (**) \end{cases}$
 (Other initial conditions can also be done)

$$t e^{-at} \quad \hat{f}(s) = \frac{1}{(s+a)^2}$$

Examples -

We saw for the low pass filter



transfer fcn:

$$\tilde{H}(\omega) \equiv \frac{\tilde{V}_{out}}{\tilde{V}_{in}}$$

$$\hat{H}(s) \equiv \frac{\hat{V}_{out}}{\hat{V}_{in}}$$



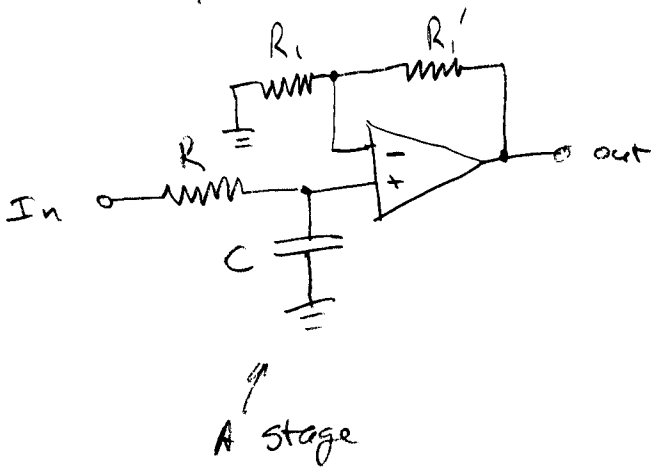
$$\left. \begin{aligned} \hat{V}_{out} &= \frac{1}{sC} \hat{I} \\ \hat{V}_{in} &= \hat{I}R + \frac{1}{sC} \hat{I} \end{aligned} \right\} \Rightarrow \hat{H}(s) = \frac{\frac{1}{sC}}{R + \frac{1}{sC}} = \frac{\omega_c}{s + \omega_c}$$

δ -function response is exponential and

$$|H(\omega)| = \frac{1}{\sqrt{1 + (\omega/\omega_c)^2}} \quad \text{where } \omega_c = \frac{1}{RC} \text{ (-3dB point)}$$

(above ω_B A_v falls at 20dB/decade)

→ Now we want to improve the filter response - we could "activate" the filter and cascade many stages to get more dB/decade;



$$\omega_B = \frac{1}{RC}, \quad A_{v0} = 1 + \frac{R'}{R_1} \text{ (arbitrary)}$$

$$\left| \frac{A_v}{A_{v0}} \right| = \frac{1}{\sqrt{1 + (\omega/\omega_c)^2}}$$

where A_{v0} = "passband transfer fcn" (at $\omega=0$ here)

So



gives 60 dB/decade
3-pole RC filter.

(ok up to a few MHz with suitable op amps)

Improve things further by making RLC filters (or "activated" filters with just R's & C's)

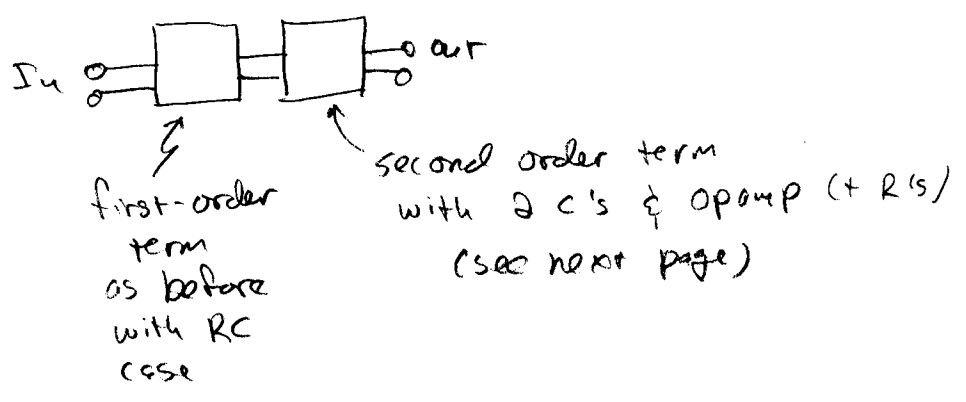
Study has shown that certain arrangements of poles are advantageous for different situations - e.g. the Butterworth filter has a sharper cutoff near ω_B than a cascaded RC filter with the same number of poles while retaining a smooth A_V in the "pass band".

$$\frac{A_V(s)}{A_{V0}} = \frac{1}{B_n(s)} \quad \xrightarrow{\text{magnitude}} \quad \left| \frac{A_V(\omega)}{A_{V0}} \right| = \frac{1}{|B_n(\omega)|} = \frac{1}{\sqrt{1 + (\omega/\omega_B)^{2n}}}$$

where

- | | | |
|-----|----------------------------------------------------------------------------------|-----------------------------|
| n | $B_n(s)$ (Butterworth polynomials) | |
| 1 | $(s/\omega_B) + 1$ | ← This one is our RC filter |
| 2 | $(s/\omega_B)^2 + 1.414(s/\omega_B) + 1$ | |
| 3 | $(s/\omega_B + 1)((s/\omega_B)^2 + s/\omega_B + 1)$ | |
| 4 | $((s/\omega_B)^2 + 0.765 s/\omega_B + 1)((s/\omega_B)^2 + 1.845 s/\omega_B + 1)$ | |
| | etc. | |

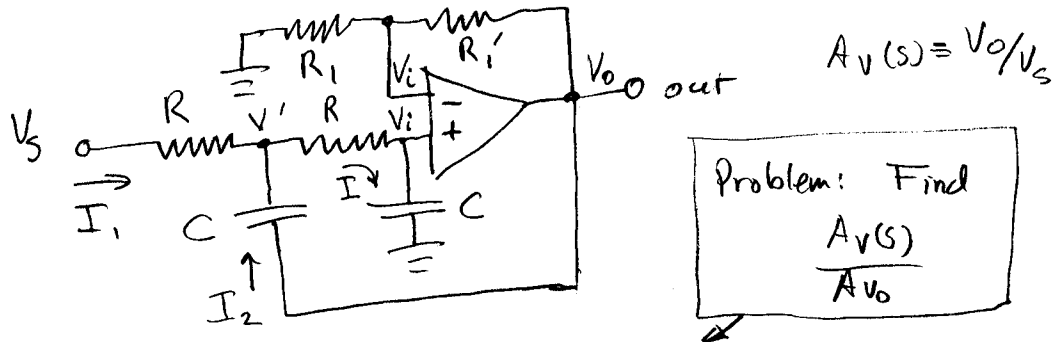
The actual filters can be made in sections with the first order terms realized by an RC circuit and the 2nd order terms by an LRC circuit or an opamp circuit with R's and C's only; e.g., for a 3-pole filter:



The 2nd order polynomial term looks in general like:

$$\frac{A_V(s)}{A_{V_0}} = \frac{1}{(s/\omega_B)^2 + 2k(s/\omega_B) + 1}$$

It can be realized using



Analyze the circuit using Kirchhoff's laws as before—work in s -space. I will not use the "hats" because I have to write V and I so much—You will see we are in s -space because things are frus of s .

For the opamp, $V_+ = V_- = V_i$. No current flows into \oplus or \ominus .

$A_{V_0} = \text{midband gain} = 1 + R_1'/R_1$ (a constant)

Now $I = V_i / \frac{1}{sC} = sC V_i$

$$\underline{V'} = I \left(R + \frac{1}{sC} \right) = sC V_i \left(R + \frac{1}{sC} \right) = \frac{V_o}{A_{V_0}} (sRC + 1)$$

$$\Rightarrow I = \frac{V_o}{A_{V_0}} \frac{sRC + 1}{R + 1/sC} = \frac{V_o}{A_{V_0}} sC$$

$$I_2 = (V_o - V') sC = I - I_1 \Rightarrow I_1 = I - (V_o - V') sC$$

$$V_s = V' + I_1 R = V' + I R - (V_o - V') sRC$$

$$= \frac{V_o}{A_{V_0}} (sRC + 1) + \frac{V_o}{A_{V_0}} sRC - V_o sRC + \frac{V_o}{A_{V_0}} (sRC + 1) sRC$$

Thus $\frac{A_{V_0}}{(V_o/V_s)} = sRC + 1 + sRC - A_{V_0} sRC + (sRC)^2 + sRC = \underline{(sRC)^2 + (3 - A_{V_0}) sRC + 1}$

or $\frac{A_V(s)}{A_{V_0}} = \frac{1}{(sRC)^2 + (3 - A_{V_0}) sRC + 1}$

choose $RC = \frac{1}{\omega_B}$
and $A_{V_0} = 3 - 2k = 1 + R_1'/R_1$
to get Filter!