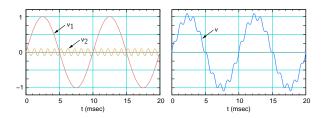
EE101: Op Amp circuits (Part 3)



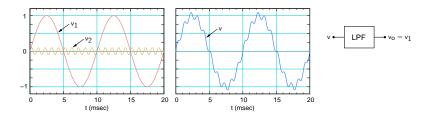
M. B. Patil mbpatil@ee.iitb.ac.in

Department of Electrical Engineering Indian Institute of Technology Bombay

Consider $v(t) = v_1(t) + v_2(t) = V_{m1} \sin \omega_1 t + V_{m2} \sin \omega_2 t$.

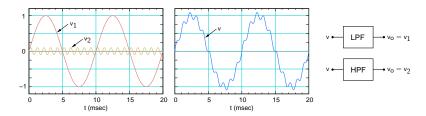


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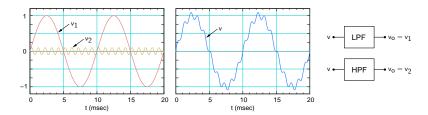


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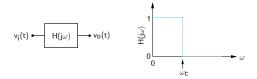
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There are some other types of filters, as we will see.

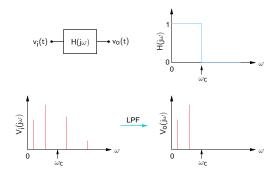


Ideal low-pass filter



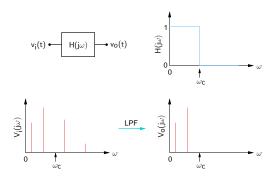
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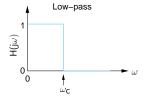


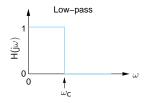
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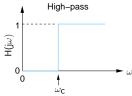
All components with $\omega < \omega_c$ appear at the output without attenuation.

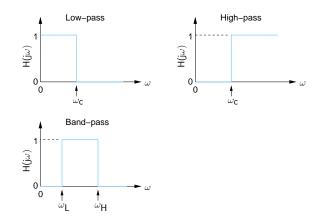
All components with $\omega>\omega_{c}$ get eliminated.

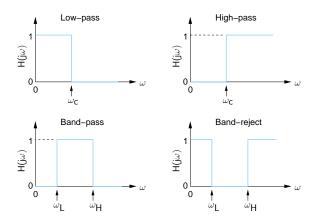
(Note that the ideal low-pass filter has $\angle H(j\omega) = 1$, i.e., $H(j\omega) = 1 + j0$.)

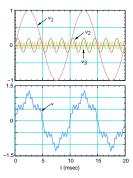


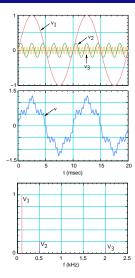


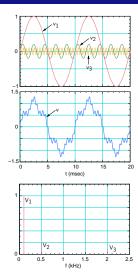




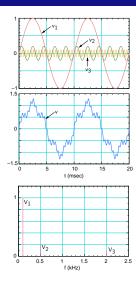


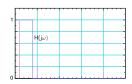


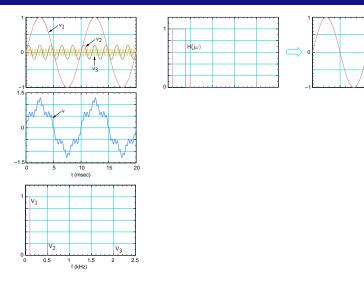


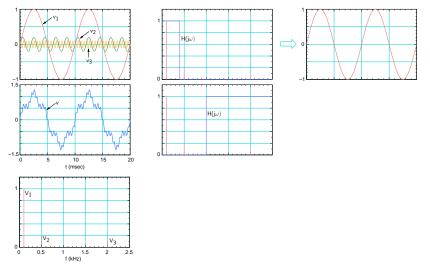


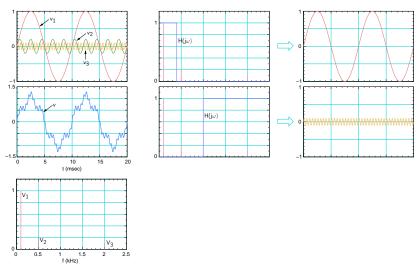
Let us see the effect of a few filters on v(t).

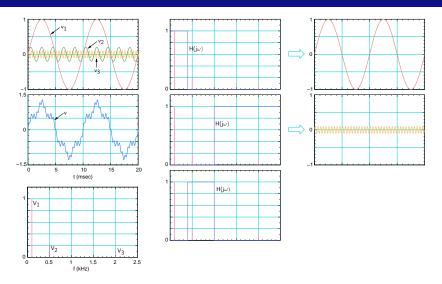


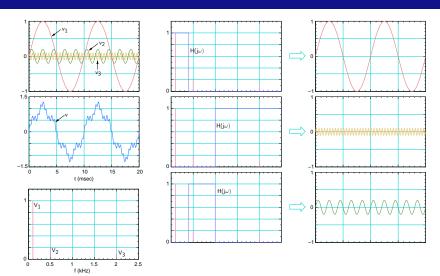


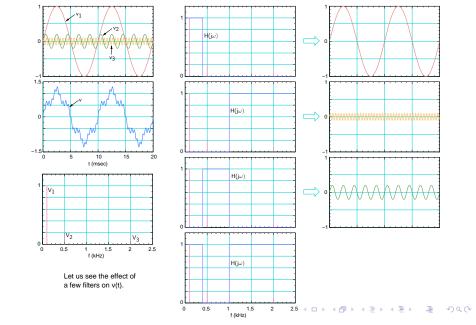


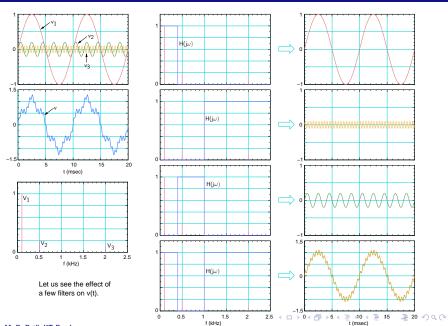












Practical filter circuits

* In practical filter circuits, the ideal filter response is approximated with a suitable $H(j\omega)$ that can be obtained with circuit elements. For example,

$$H(s) = \frac{1}{a_5s^5 + a_4s^4 + a_3s^3 + a_2s^2 + a_1s + a_0}$$

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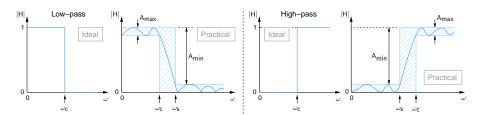
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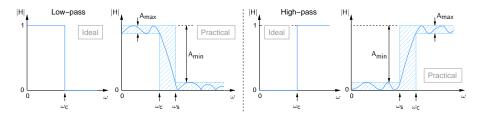
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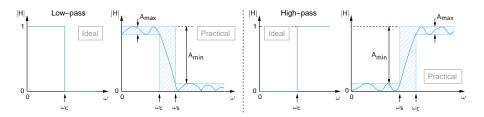
- * Some commonly used approximations (polynomials) are the Butterworth, Chebyshev, Bessel, and elliptic functions.
- * Coefficients for these filters listed in filter handbooks. Also, programs for filter design are available on the internet.



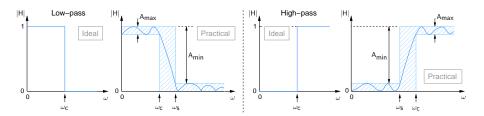




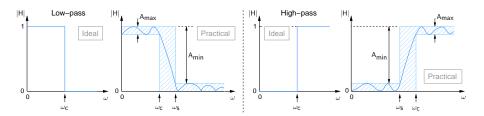
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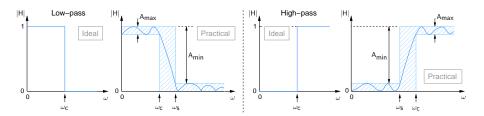


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- * $\omega_c < \omega < \omega_s$: transition band.



For a low-pass filter,
$$H(s) = \frac{1}{\displaystyle\sum_{i=0}^{n} a_i (s/\omega_c)^i}$$
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Coefficients (a_i) for various types of filters are tabulated in handbooks. We now look at $|H(j\omega)|$ for two commonly used filters.

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$$|H(j\omega)|=rac{1}{\sqrt{1+\epsilon^2C_n^2(\omega/\omega_c)}}$$
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$$C_n(x) = \cos\left[n\cos^{-1}(x)\right] \text{ for } x \le 1,$$

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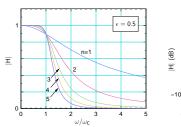
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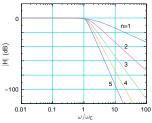
H(s) for a high-pass filter can be obtained from H(s) of the corresponding low-pass filter by $(s/\omega_c) \to (\omega_c/s)$.



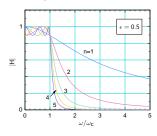
Practical filters (low-pass)

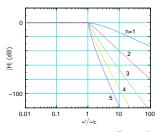
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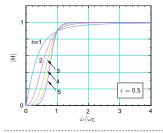
Chebyshev filters:

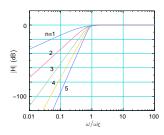




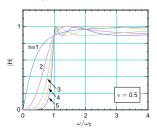
Practical filters (high-pass)

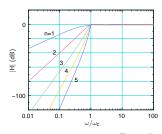
Butterworth filters:





Chebyshev filters:

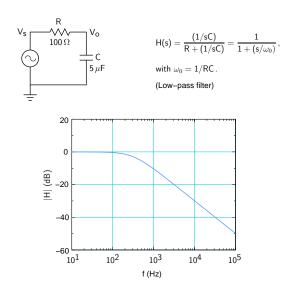




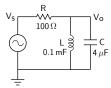


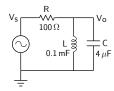


$$\label{eq:Hsigma} \begin{split} & \text{H(s)} = \frac{(1/\text{sC})}{\text{R} + (1/\text{sC})} = \frac{1}{1 + (\text{s}/\omega_0)} \\ & \text{with } \omega_0 = 1/\text{RC} \ . \\ & \text{(Low-pass filter)} \end{split}$$

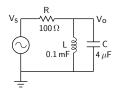


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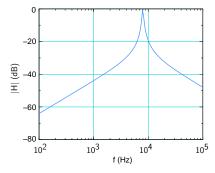




$$\begin{split} &H(s) = \frac{(sL) \parallel (1/sC)}{R + (sL) \parallel (1/sC)} = \frac{s(L/R)}{1 + s(L/R) + s^2LC} \\ &\text{with } \omega_0 = 1/\sqrt{LC} \,. \\ &\text{(Band-pass filter)} \end{split}$$



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(SEQUEL file: ee101_lc_1.sqproj)

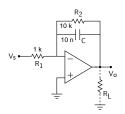
* Op Amp filters can be designed without using inductors. This is a significant advantage since inductors are bulky and expensive. Inductors also exhibit nonlinear behaviour (arising from the core properties) which is undesirable in a filter circuit.

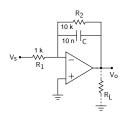
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- * However, at high frequencies (\sim MHz), Op Amps no longer have a high gain \rightarrow passive filters.

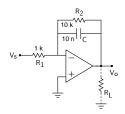
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- * Also, if the power requirement is high, Op Amp filters cannot be used \rightarrow passive filters.





Op Amp filters are designed for Op Amp operation in the linear region \rightarrow Our analysis of the inverting amplifier applies, and we get,

$$\mathbf{V_o} = -rac{R_2 \parallel (1/sC)}{R_1}\,\mathbf{V_s}\,\,\,\,(\mathbf{V_s}\,\, ext{and}\,\,\mathbf{V_o}\,\, ext{are phasors})$$
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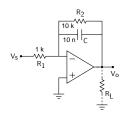


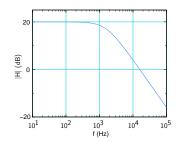
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This is a low-pass filter, with $\omega_0=1/R_2C$.



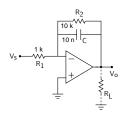


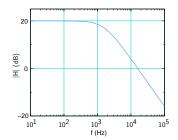
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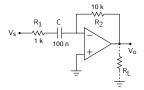
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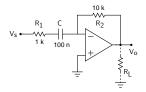
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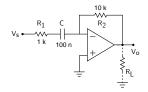
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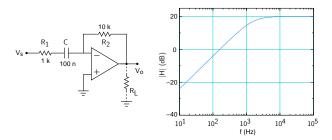


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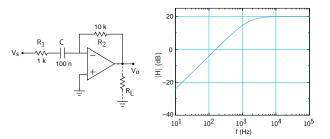
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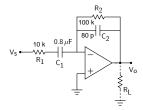
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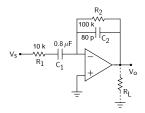


$$H(s) = -\frac{R_2}{R_1 + (1/sC)} = \frac{sR_2C}{1 + sR_1C}$$
.

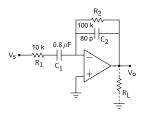
This is a high-pass filter, with $\omega_0=1/R_1\,{\it C}$.

(SEQUEL file: ee101_op_filter_2.sqproj)



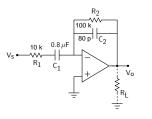


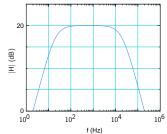
$$H(s) = -\frac{R_2 \parallel (1/sC_2)}{R_1 + (1/sC_1)} = -\frac{R_2}{R_1} \, \frac{sR_1\,C_1}{(1+sR_1\,C_1)(1+sR_2\,C_2)} \, .$$



$$H(s) = -\frac{R_2 \parallel (1/sC_2)}{R_1 + (1/sC_1)} = -\frac{R_2}{R_1} \, \frac{sR_1\,C_1}{(1+sR_1\,C_1)(1+sR_2\,C_2)} \,.$$

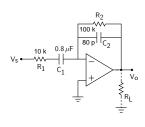
This is a band-pass filter, with $\omega_L=1/R_1\, C_1$ and $\omega_H=1/R_2\, C_2$.

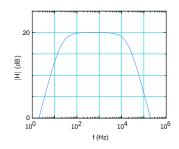




$$H(s) = -\frac{R_2 \parallel (1/sC_2)}{R_1 + (1/sC_1)} = -\frac{R_2}{R_1} \frac{sR_1C_1}{(1 + sR_1C_1)(1 + sR_2C_2)} \,.$$

This is a band-pass filter, with $\omega_L=1/R_1\, C_1$ and $\omega_H=1/R_2\, C_2$.



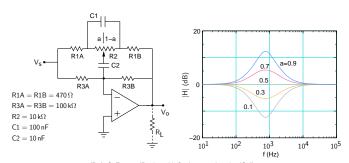


$$H(s) = -\frac{R_2 \parallel (1/sC_2)}{R_1 + (1/sC_1)} = -\frac{R_2}{R_1} \, \frac{sR_1\,C_1}{(1+sR_1\,C_1)(1+sR_2\,C_2)} \,.$$

This is a band-pass filter, with $\omega_L=1/R_1\, C_1$ and $\omega_H=1/R_2\, C_2$.

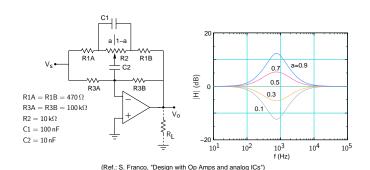
(SEQUEL file: ee101_op_filter_3.sqproj)

Graphic equalizer



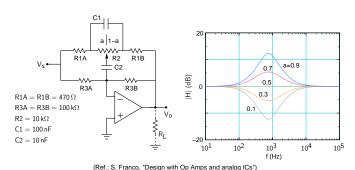
(Ref.: S. Franco, "Design with Op Amps and analog ICs")

Graphic equalizer



* Equalizers are implemented as arrays of narrow-band filters, each with an adjustable gain (attenuation) around a centre frequency.

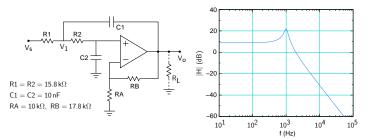
Graphic equalizer



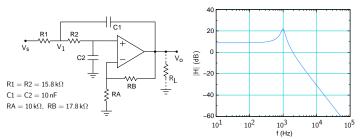
(IVel.: 5. I failed, Design with Op Amps and analog les)

- * Equalizers are implemented as arrays of narrow-band filters, each with an adjustable gain (attenuation) around a centre frequency.
- * The circuit shown above represents one of the equalizer sections.

 (SEQUEL file: ee101_op_filter_4.sqproj)

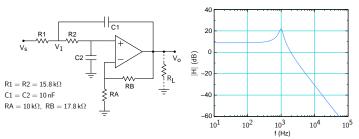


(Ref.: S. Franco, "Design with Op Amps and analog ICs")



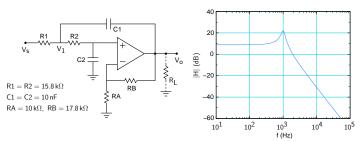
(Ref.: S. Franco, "Design with Op Amps and analog ICs")

$$V_{+} = V_{-} = V_{o} \, rac{R_{A}}{R_{A} + R_{B}} \equiv V_{o}/K \, .$$



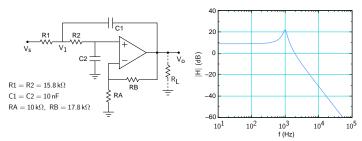
(Ref.: S. Franco, "Design with Op Amps and analog ICs")

$$\begin{split} V_+ &= V_- = V_o \, \frac{R_A}{R_A + R_B} \equiv V_o / K \, . \\ \text{Also, } V_+ &= \frac{\left(1/s C_2\right)}{R_2 + \left(1/s C_2\right)} \, V_1 = \frac{1}{1 + s R_2 C_2} \, V_1 \, . \end{split}$$



(Ref.: S. Franco, "Design with Op Amps and analog ICs")

$$\begin{split} V_+ &= V_- = V_o \, \frac{R_A}{R_A + R_B} \equiv V_o / K \, . \\ \text{Also, } V_+ &= \frac{(1/sC_2)}{R_2 + (1/sC_2)} \, V_1 = \frac{1}{1 + sR_2C_2} \, V_1 \, . \\ \text{KCL at } V_1 &\to \frac{1}{R_1} (V_s - V_1) + sC_1 (V_o - V_1) + \frac{1}{R_2} (V_+ - V_1) = 0 \, . \end{split}$$



(Ref.: S. Franco, "Design with Op Amps and analog ICs")

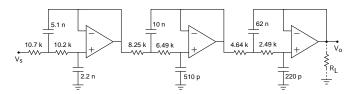
$$\begin{split} V_{+} &= V_{-} = V_{o} \, \frac{R_{A}}{R_{A} + R_{B}} \equiv V_{o} / K \, . \\ \text{Also, } V_{+} &= \frac{(1/sC_{2})}{R_{2} + (1/sC_{2})} \, V_{1} = \frac{1}{1 + sR_{2}C_{2}} \, V_{1} \, . \\ \text{KCL at } V_{1} &\to \frac{1}{R_{1}} (V_{s} - V_{1}) + sC_{1} (V_{o} - V_{1}) + \frac{1}{R_{2}} (V_{+} - V_{1}) = 0 \, . \end{split}$$

Combining the above equations, $H(s) = \frac{K}{1 + s [(R_1 + R_2)C_2 + (1 - K)R_1C_1] + s^2R_1C_1R_2C_2}$

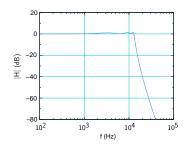
(SEQUEL file: ee101_op_filter_5.sqproj)



Sixth-order Chebyshev low-pass filter (cascade design)

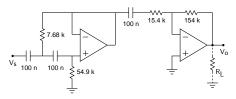


(Ref.: S. Franco, "Design with Op Amps and analog ICs")

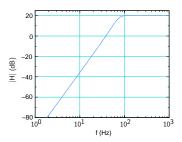


SEQUEL file: ee101_op_filter_6.sqproj

Third-order Chebyshev high-pass filter

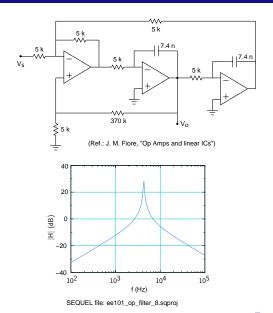


(Ref.: S. Franco, "Design with Op Amps and analog ICs")



SEQUEL file: ee101_op_filter_7.sqproj

Band-pass filter example



Notch filter example

