

State University of New York

EECE 301 Signals & Systems Prof. Mark Fowler

Note Set #35

• C-T Systems: CT Filters - Passive

Introductory Comments

Recall that we already talked about ideal CT filters:

- $|H(\omega)|$ is Constant in Pass band
- $|H(\omega)|$ is zero in Stop band (Transition Band has zero width)
- $\angle H(\omega)$ is linear in Pass band

We also saw that such ideal filters can not really exist because they would need to be non-causal!!

Here we'll take a brief look at some of the kinds of CT filters that <u>can</u> be made...

• Note... all CT filter behavior exploits the fact that capacitors and inductors have an impedance that varies with frequency!

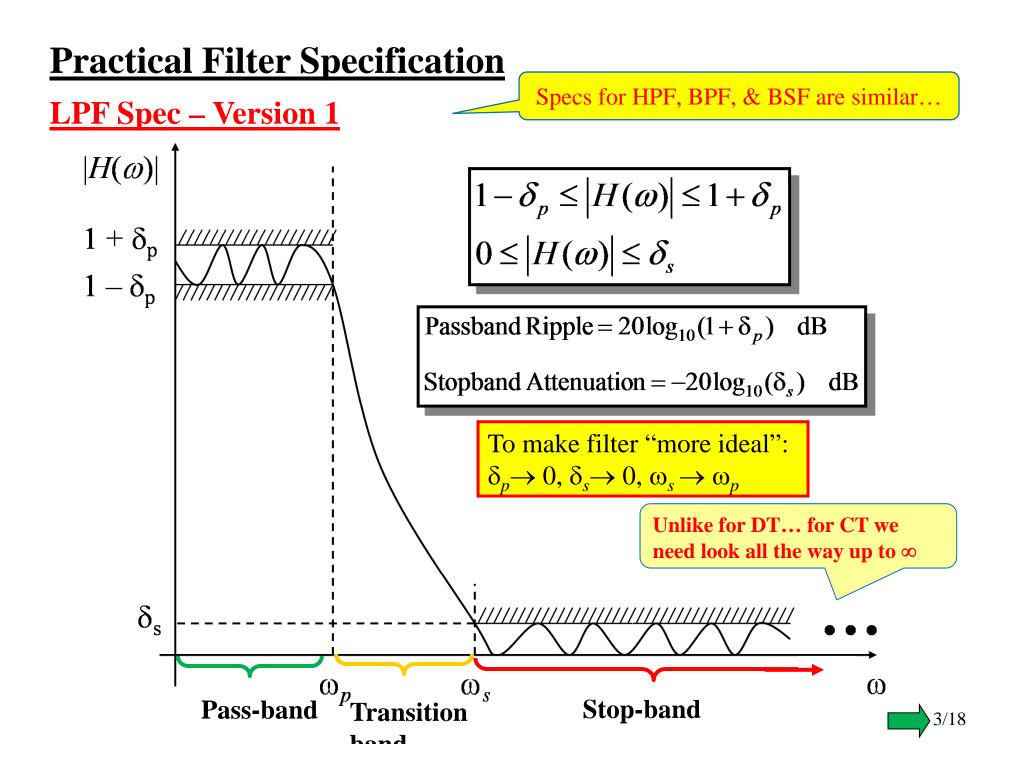
And we'll illustrate how to describe such filters using:

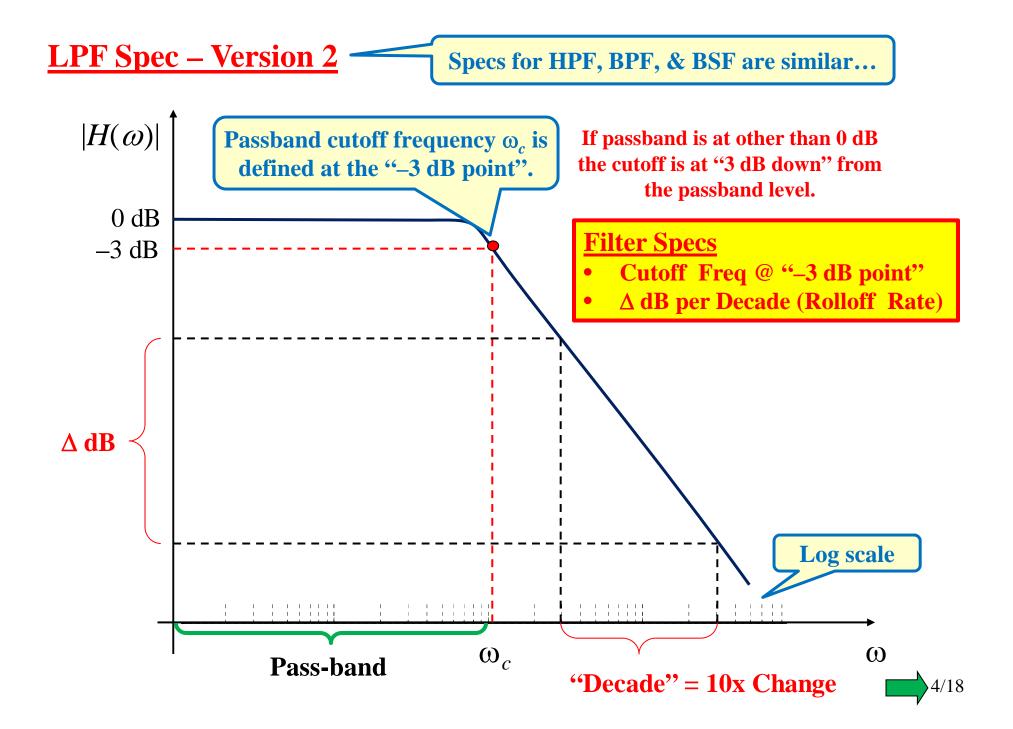
- Transfer Function
- Frequency Response
- Pole-Zero Diagrams

Also... keep in mind that although DT filters only need to be examined over $-\pi$ to π rad/*sample* (their Freq Resp repeats outside of that)... CT filters need to be examined for how they behave over $-\infty$ to ∞ rad/*second*. Thus, we will mostly plot them on a log frequency axis... with dB for the magnitude.



"CT Filters" are also called "Analog Filters"





CT Filter Types

Recall that DT filters were categorized as recursive (IIR) vs. non-recursive (FIR).

CT filters don't have a corresponding categorization... they all have infinite duration impulse responses!!!

Instead the main way to categorize CT filters is: <u>Passive</u> vs. <u>Active</u>

<u>Passive</u>: These filters use only "passive components" (resistors, capacitors, and inductors) and do not contain any op amps or transistors.

• One main advantage of such filters is that they can be used in places where access to a power supply is not available (e.g., inside a stereo speaker to separate the audio into bass and treble before sending it to the woofer & tweeter).

Active: These filters use op amps (and/or transistors) together with resistors, capacitors, and inductors.

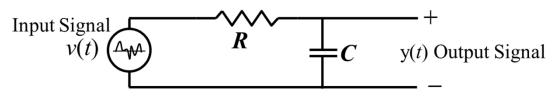
• Allows filters to be designed without inductors

Expensive

- Op amp characteristics enable design by cascading several "stages"
 - Large Input Impedance
 - Small Output Impedance



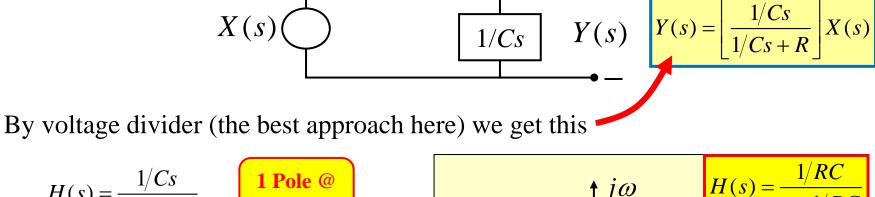
"First-Order" Lowpass Filter: RC Circuit

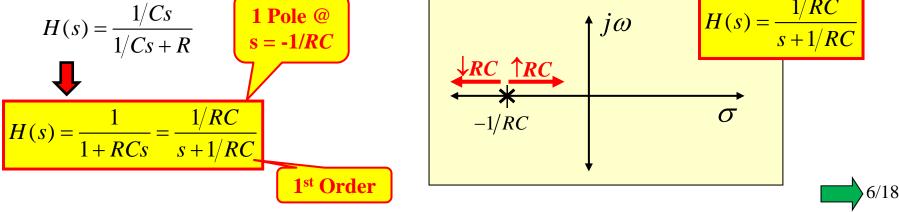


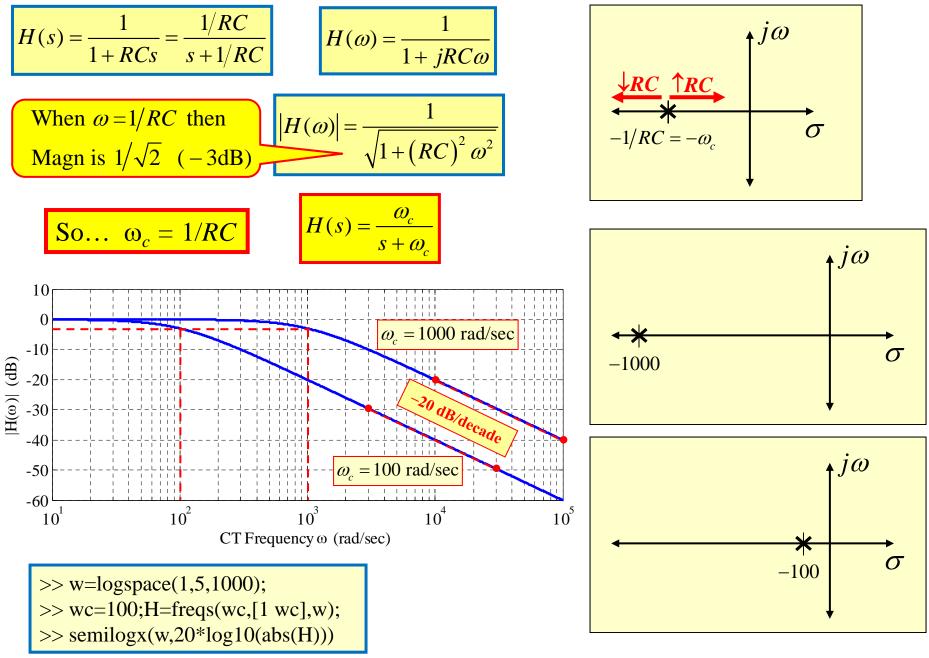
We already analyzed this filter using phasor ideas... but we'll take another look here.

To analyze this filter in the s-domain:

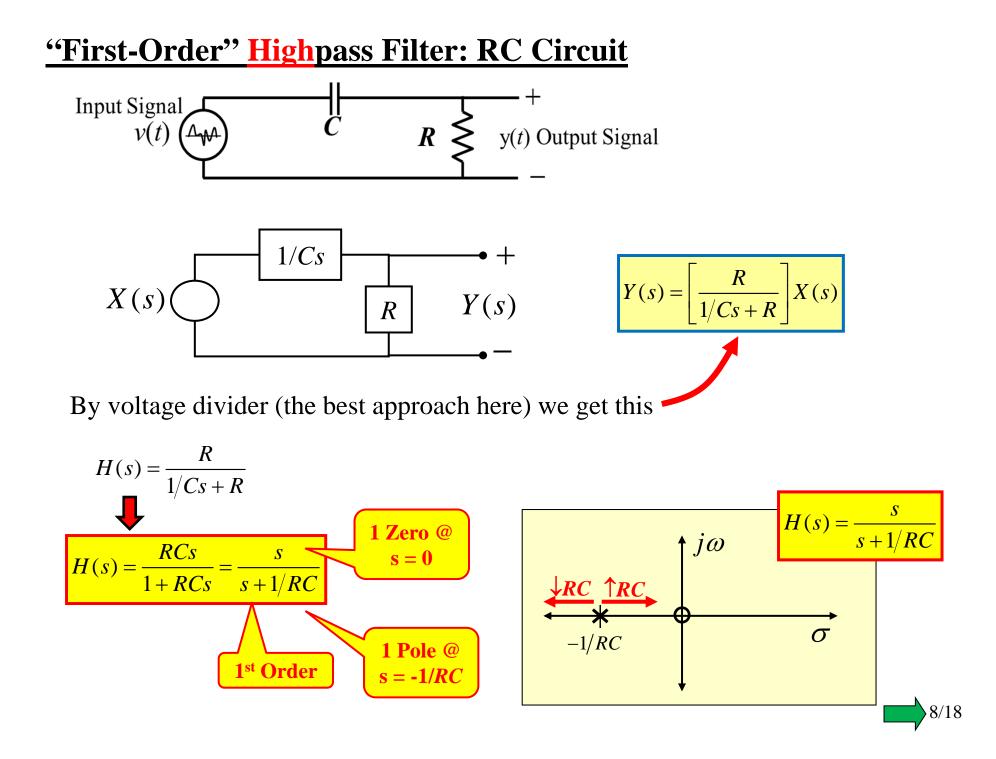
- Replace input and output by their LT symbols
- Replace components by their <u>s-domain impedances</u>
- Solve for output Y(s) in terms of input X(s)... the thing that multiplies X(s) is the TF H(s)

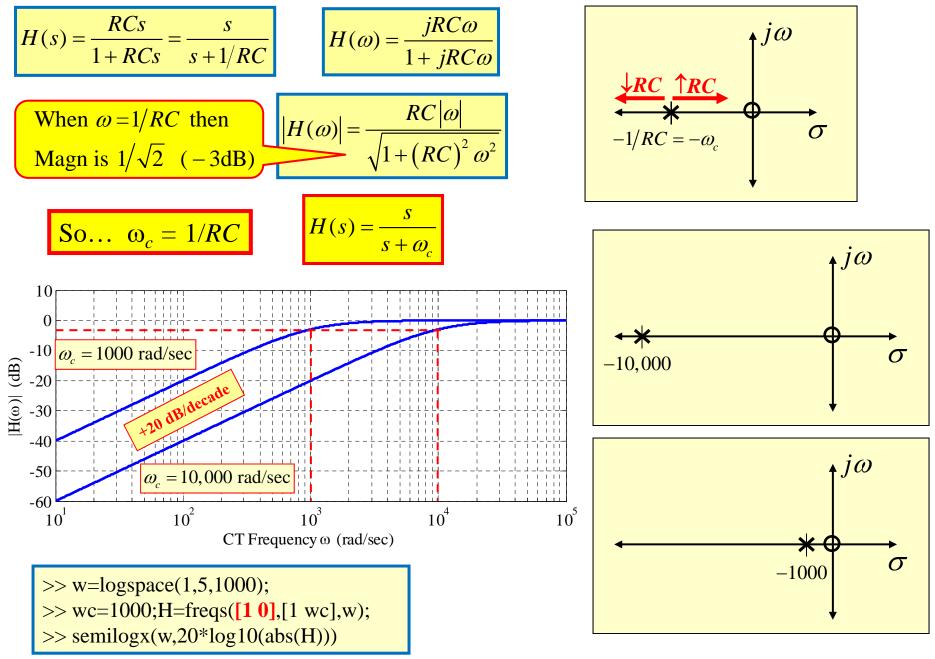




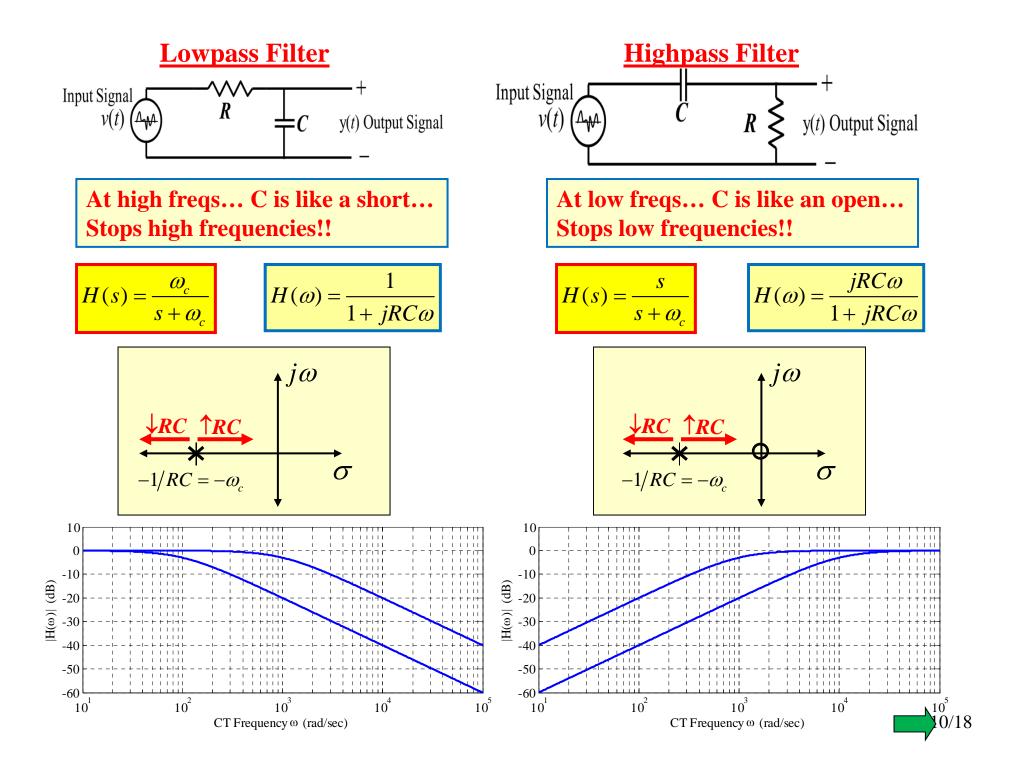


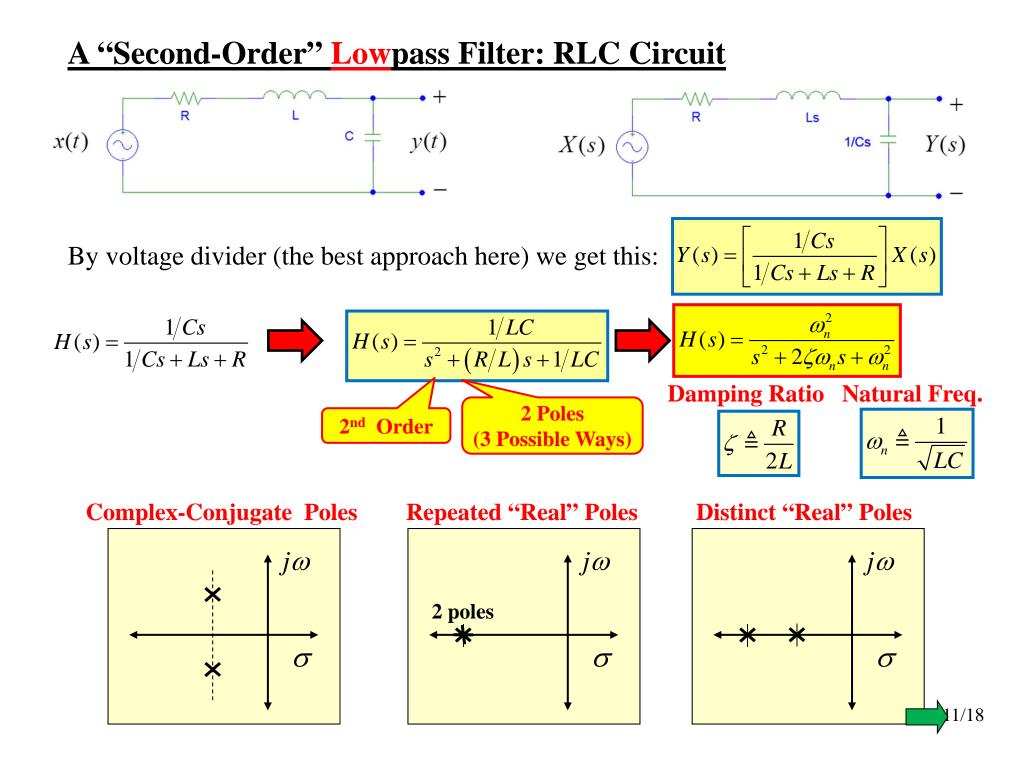
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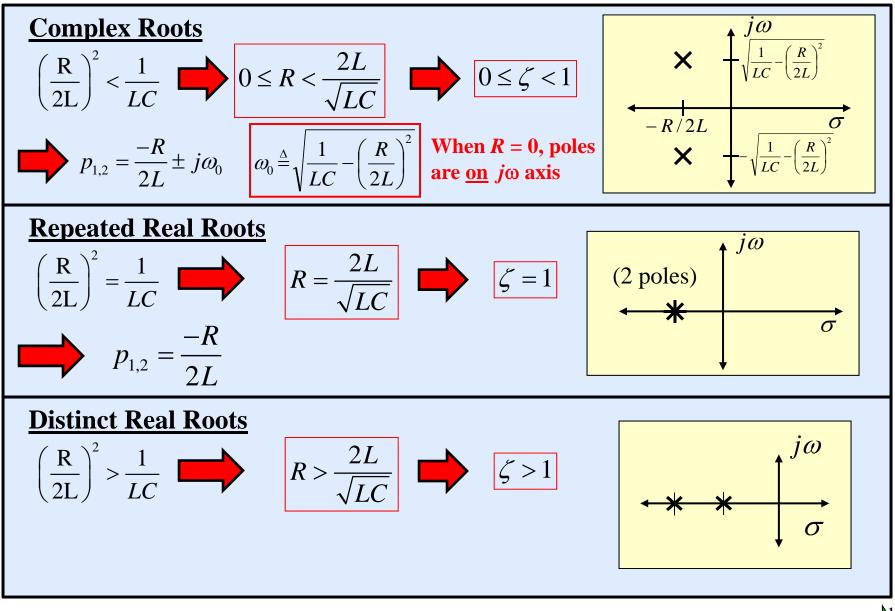


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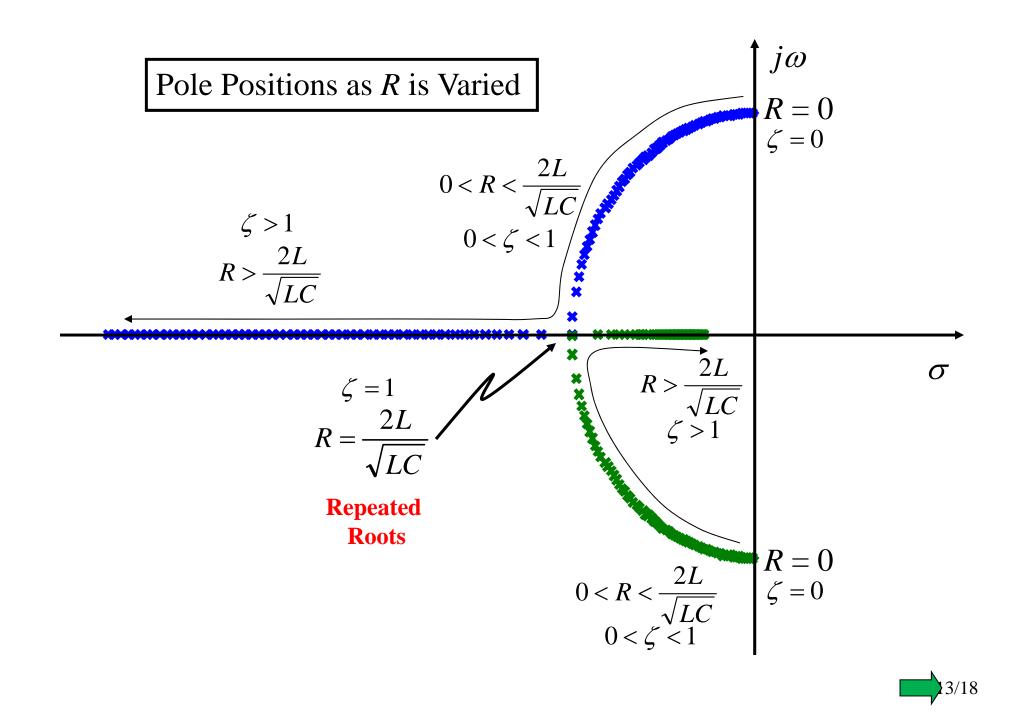


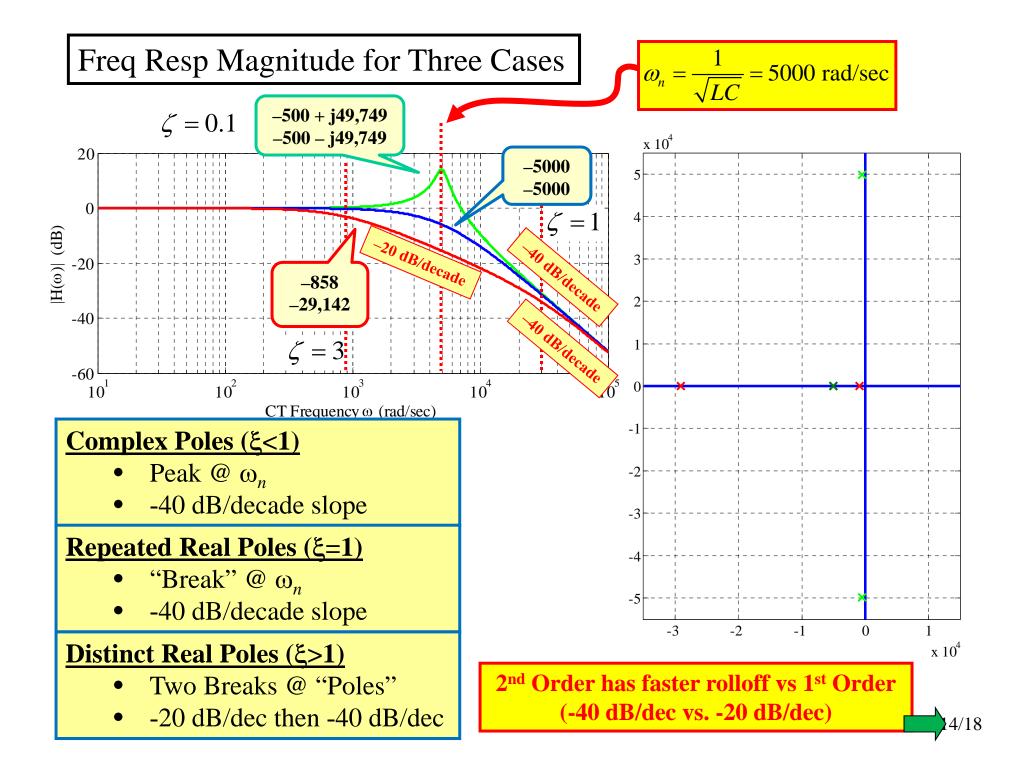


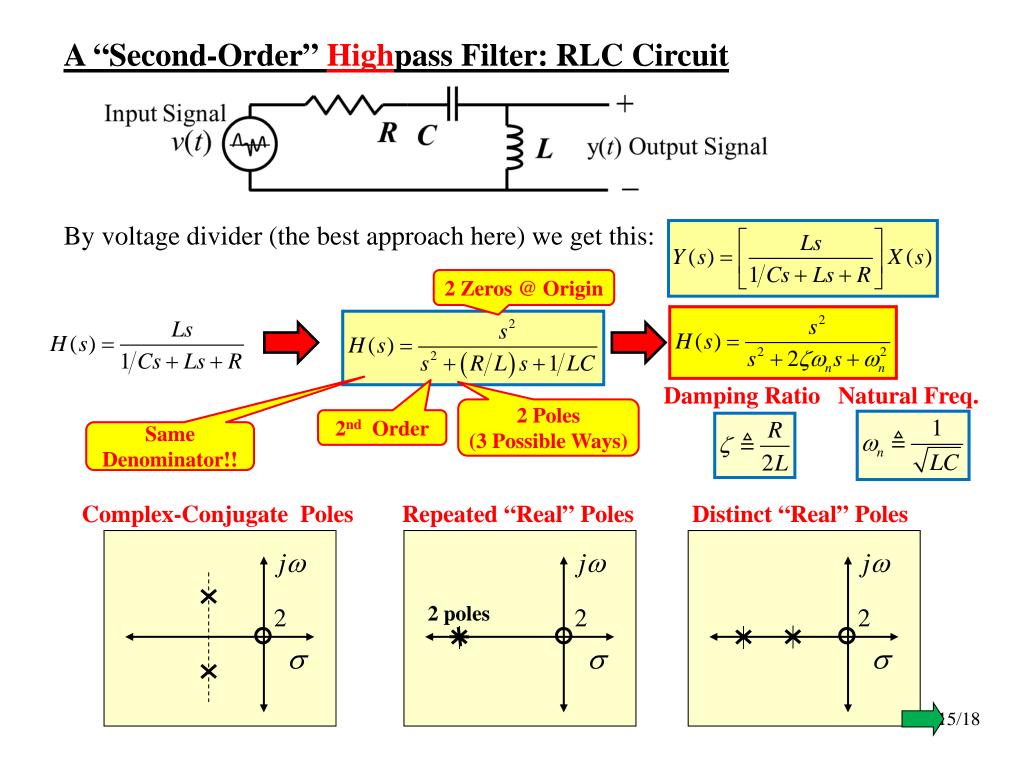
The poles are the roots of $s^2 + (R/L)s + 1/LC$: $p_{1,2} = -R/2L \pm \sqrt{(R/2L)^2 - 1/LC}$

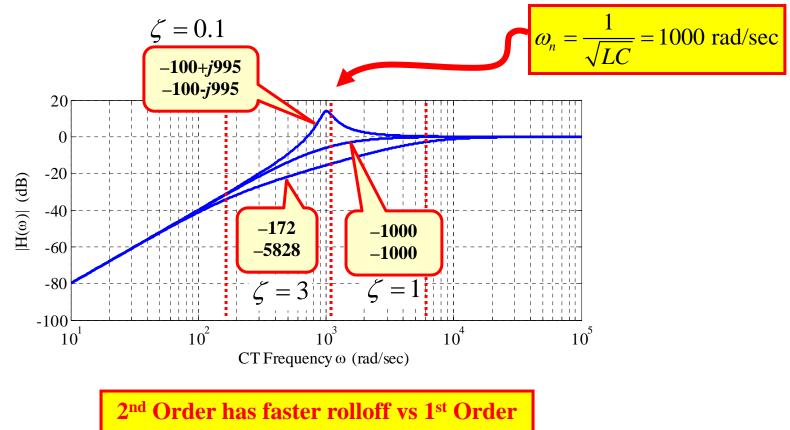


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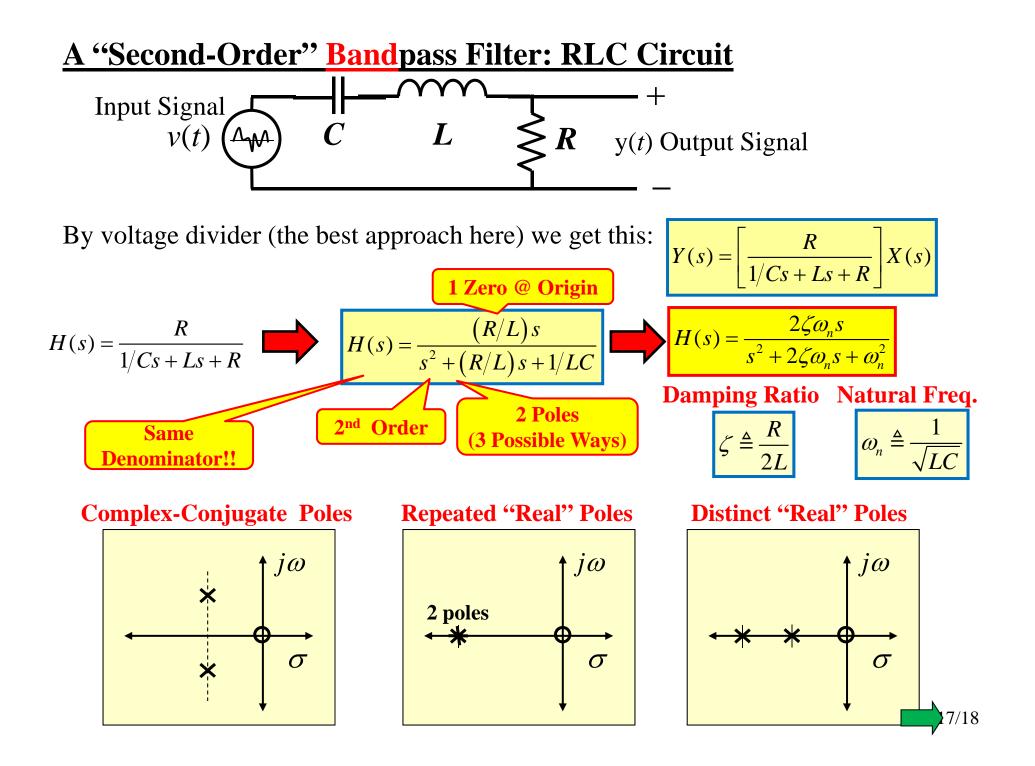
(-40 dB/dec vs. -20 dB/dec)

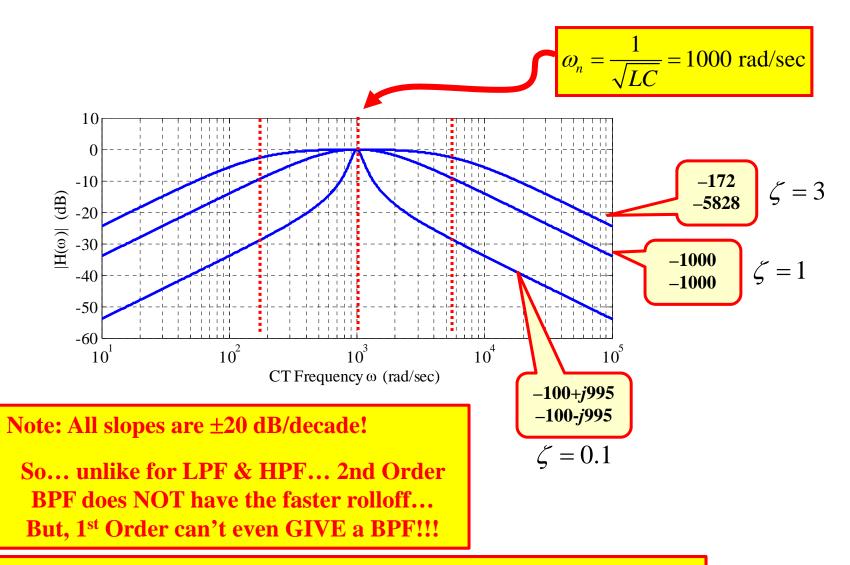
What we are seeing is that we get 20 dB of slope for each order!!! (For LPF and HPF... But see next for BPF...)

So a 3rd Order LPF would (eventually) rolloff at -60 dB/decade!!!

So... the main advantage of higher order filters is that your stop band is better due to the faster rolloff!!







What is happening is that the second order gives you two 20 dB/dec slopes "available"...

But for a BPF you need one going up and one going down... so each only gets one of the two 20 dB slopes!

