Introduction to Filters

Colophon

An annotatable worksheet for this presentation is available as Worksheet 15 Worksheet 15 https://cpjobling.github.io/eg-247-textbook/fourier_transform/4/worksheet15.html).

- The source code for this page is content/fourier_transform/4/ft4.ipynb (https://github.com/cpjobling/eg-247-textbook/blob/master/content/fourier_transform/4/ft4.ipynb).
- You can view the notes for this presentation as a webpage (<u>HTML (https://cpjobling.github.io/eg-247-textbook/fourier_transform/4/ft4.html)</u>).
- This page is downloadable as a <u>PDF (https://cpjobling.github.io/eg-247-textbook/fourier_transform/4/ft4.pdf)</u> file.

Scope and Background Reading

This section is Based on the section **Filtering** from Chapter 5 of <u>Benoit Boulet</u>, <u>Fundamentals of Signals and Systems (https://ebookcentral.proquest.com/lib/swansea-ebooks/reader.action?</u>

ppg=221&docID=3135971&tm=1518715953782) from the **Recommended Reading List**.

This material is an introduction to analogue filters. You will find much more in-depth coverage on <u>Pages 11-1—11-48 of Karris (https://ebookcentral.proquest.com/lib/swansea-ebooks/reader.action?</u>
ppg=429&docID=3384197&tm=1518716026573).

Agenda

- Frequency Selective Filters
- Ideal low-pass filter
- Butterworth low-pass filter

- · High-pass filter
- Bandpass filter

Introduction

- Filter design is an important application of the Fourier transform
- Filtering is a rich topic often taught in graduate courses so we give only an introduction.
- Our introduction will illustrate the usefulness of the frequency domain viewpoint.
- We will explore how filters can shape the spectrum of a signal.

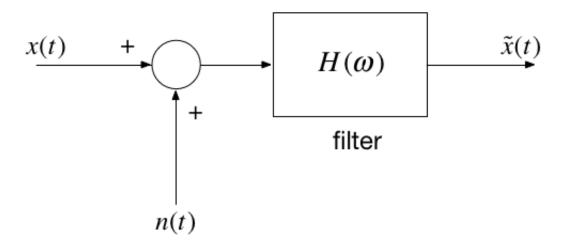
Other applications of the Fourier transform are sampling theory (introduced next week) and modulation.

Frequency Selective Filters

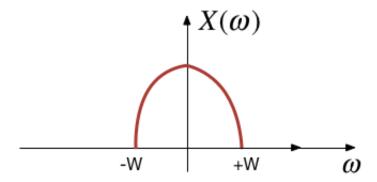
An ideal frequency-selective filter is a system that let's the frequency components of a signal through undistorted while frequency components at other components are completely cut off.

- The range of frequencies which are let through belong to the pass Band
- The range of frequencies which are cut-off by the filter are called the stopband
- A typical scenario where filtering is needed is when noise n(t) is added to a signal x(t) but that signal has most of its energy outside the bandwidth of a signal.

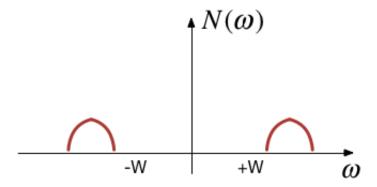
Typical filtering problem



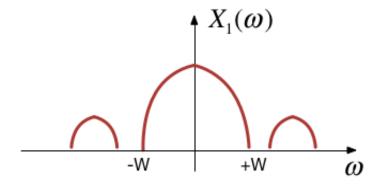
Signal



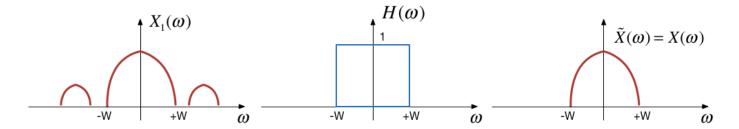
Out-of Bandwidth Noise



Signal plus Noise



Filtering



Motivating example

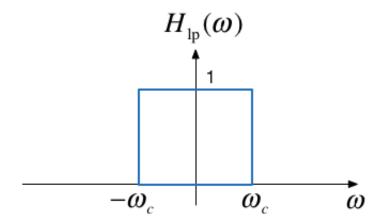
See the video and script on <u>Canvas Week 7 (https://canvas.swansea.ac.uk/courses/646/pages/week-7-classroom-activities?module_item_id=398892).</u>

Ideal Low-Pass Filter (LPF)

An ideal low pass filter cuts-off frequencies higher than its $\it cut$ -off frequency, $\it \omega_{\it c}$.

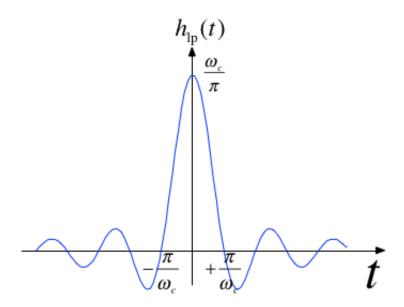
$$H_{\rm lp}(\omega) = \begin{cases} 1, & |\omega| < \omega_c \\ 0, & |\omega| \ge \omega_c \end{cases}$$

Frequency response of LPF



Impulse response

$$h_{\mathrm{lp}}(t) = \frac{\omega_c}{\pi} \mathrm{sinc}\left(\frac{\omega_c}{\pi}t\right)$$



Filtering is Convolution

The output of an LTI system with impulse response

$$h(t) \Leftrightarrow H(\omega)$$

subject to an input signal

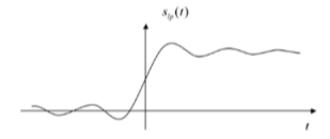
$$x(t) \Leftrightarrow X(\omega)$$

is given by

$$y(t) = h(t) * x(t) \Leftrightarrow Y(\omega) = H(\omega)X(\omega)$$

Issues with the "ideal" filter

This is the step response:



(reproduced from Boulet Fig. 5.23 p. 205)

Ripples in the impulse resonse would be undesireable, and because the impulse response is non-causal it cannot actually be implemented.

Butterworth low-pass filter

N-th Order Butterworth Filter

$$|H_B(\omega)| = \frac{1}{\left(1 + \left(\frac{\omega}{\omega_c}\right)^{2N}\right)^{\frac{1}{2}}}$$

Remarks

· DC gain is

$$|H_B(j0)| = 1$$

· Attenuation at the cut-off frequency is

$$|H_B(j\omega_c)| = 1/\sqrt{2}$$

for any N

More about the Butterworth filter: Wikipedia Article (http://en.wikipedia.org/wiki/Butterworth_filter)

Example 5: Second-order BW Filter

The second-order butterworth Filter is defined by is Characteristic Equation (CE):

$$p(s) = s^2 + \omega_c \sqrt{2}s + \omega_c^2 = 0^*$$

Calculate the roots of p(s) (the poles of the filter transfer function) in both Cartesian and polar form.

Note: This has the same characteristic as a control system with damping ratio $\zeta = 1/\sqrt{2}$ and $\omega_n = \omega_c$!

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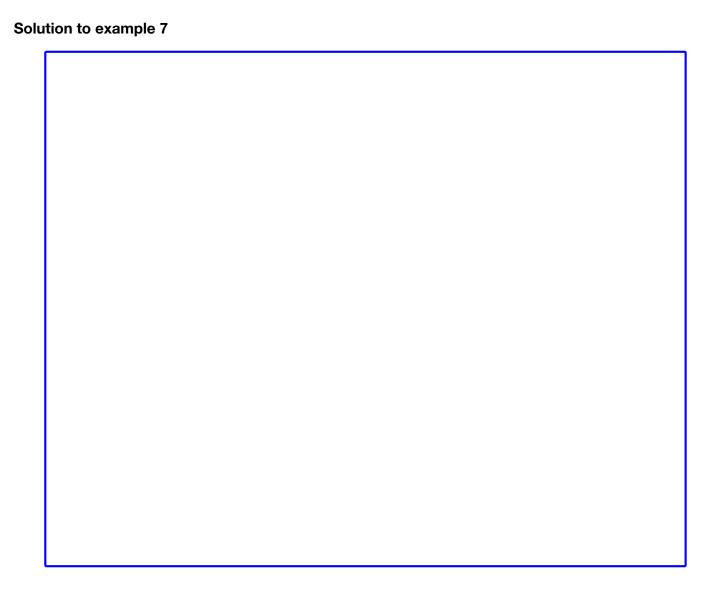
Example 6

Derive the differential equation relating the input x(t) to output y(t) of the 2nd-Order Butterworth Low-Pass Filter with cutoff frequency ω_c .

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Example 7

Determine the frequency response $H_B(\omega) = Y(\omega)/X(\omega)$



Magnitude of frequency response of a 2nd-order Butterworth Filter

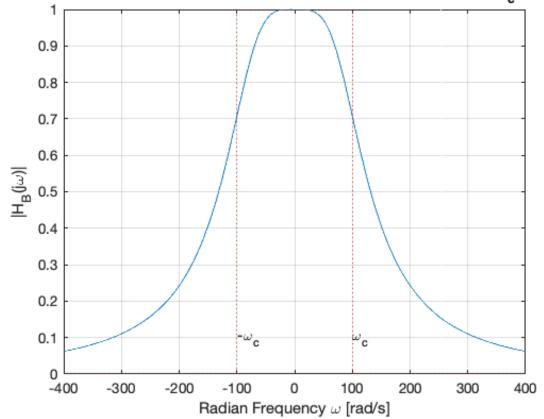
Transfer function

Continuous-time transfer function.

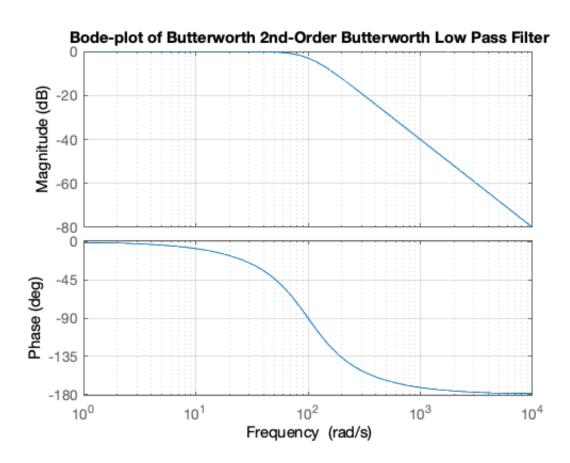
Magnitude frequency response

```
In [4]: w = -400:400;
mHlp = 1./(sqrt(1 + (w./wc).^4));
plot(w,mHlp)
grid
ylabel('|H_B(j\omega)|')
title('Magnitude Frequency Response for 2nd-Order LP Butterworth Fi
lter (\omega_c = 100 rad/s)')
xlabel('Radian Frequency \omega [rad/s]')
text(100,0.1,'\omega_c')
text(-100,0.1,'-\omega_c')
hold on
plot([-400,-100,-100,100,400],[0,0,1,1,0,0],'r:')
hold off
```

Magnitude Frequency Response for 2nd-Order LP Butterworth Filter ($\omega_{_{\mathrm{C}}}$ = 100 rad



Bode plot



Example 8

Determine the impulse and step response of a butterworth low-pass filter.

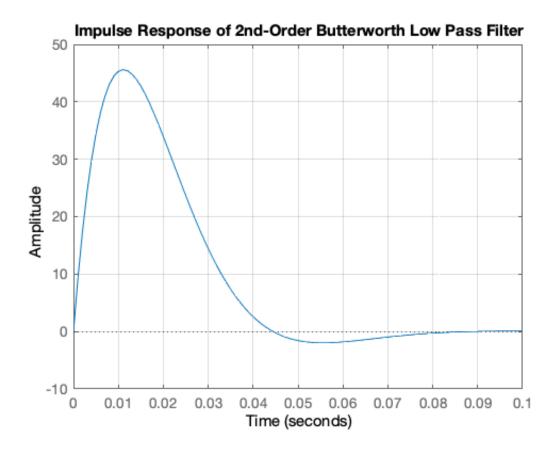
You will find this Fourier transform pair useful:

$$e^{-at} \sin \omega_0 t \ u_0(t) \Leftrightarrow \frac{\omega_0}{(j\omega + a)^2 + \omega_0^2}$$

Solu	Solution to example 8				

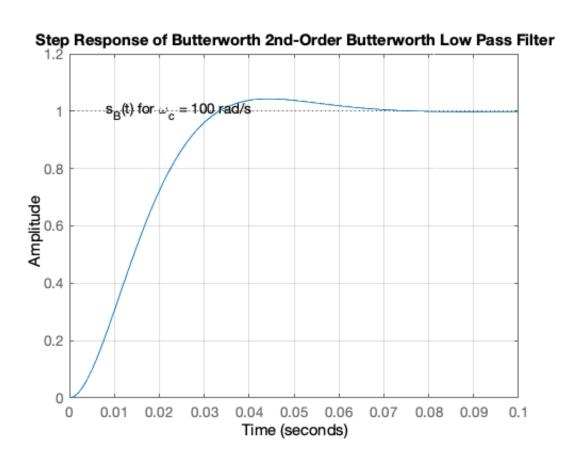
Impulse response

```
In [6]: impulse(H,0.1)
    grid
    title('Impulse Response of 2nd-Order Butterworth Low Pass Filter')
```



Step response

```
In [7]: step(H,0.1)
    title('Step Response of Butterworth 2nd-Order Butterworth Low Pass
    Filter')
    grid
    text(0.008,1,'s_B(t) for \omega_c = 100 rad/s')
```

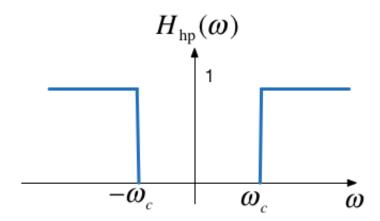


High-pass filter (HPF)

An ideal highpass filter cuts-off frequencies lower than its *cutoff frequency*, ω_c .

$$H_{\rm hp}(\omega) = \left\{ egin{array}{ll} 0, & |\omega| \leq \omega_c \\ 1, & |\omega| > \omega_c \end{array} \right.$$

Frequency response of HPF



Responses

Frequency response

$$H_{\rm hp}(\omega) = 1 - H_{\rm lp}(\omega)$$

Impulse response

$$h_{\rm hp}(t) = \delta(t) - h_{\rm lp}(t)$$

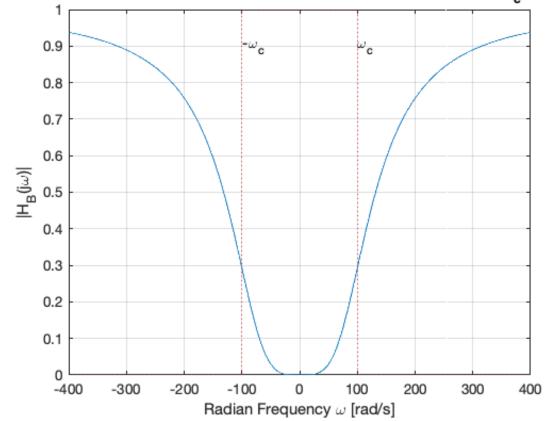
Example 9

Determine the frequency response of a 2nd-order butterworth highpass filter

olution to example	ution to example 9				

Magnitude frequency response

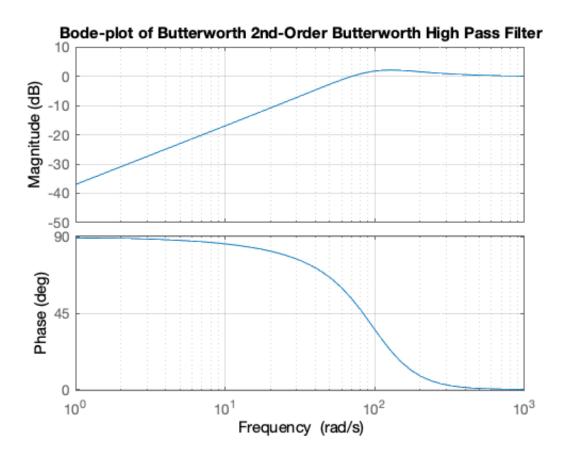
//agnitude Frequency Response for 2nd-Order HP Butterworth Filter (ω_{c} = 100 ra



High-pass filter

Hhp =

Continuous-time transfer function.

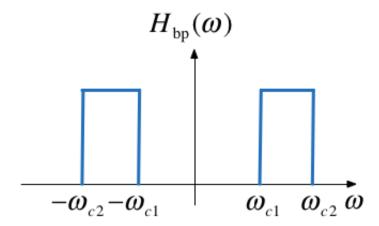


Band-pass filter (BPF)

An ideal bandpass filter cuts-off frequencies lower than its first *cutoff frequency* ω_{c1} , and higher than its second *cutoff frequency* ω_{c2} .

$$H_{\rm bp}(\omega) = \left\{ egin{array}{ll} 1, & \omega_{c1} < |\omega| < \omega_{c2} \\ 0, & {
m otherwise} \end{array} \right.$$

Frequency response of BPF



Bandpass filter design

A bandpass filter can be obtained by multiplying the frequency responses of a lowpass filter by a highpass filter.

$$H_{\rm bp}(\omega) = H_{\rm hp}(\omega)H_{\rm lp}(\omega)$$

- ullet The highpass filter should have cut-off frequency of ω_{c1}
- The lowpass filter should have cut-off frequency of ω_{c2}

To generate all the plots shown in this presentation, you can use <u>butter2_ex.mlx</u> (https://cpjobling.github.io/eg-247-textbook/fourier_transform/4/butter2_ex.mlx)

Summary

- Frequency-Selective Filters
- Ideal low-pass filter
- Butterworth low-pass filter
- · High-pass filter
- · Bandpass filter

Solutions

Solutions to Examples 5-9 are captured as a PenCast in <u>filters.pdf (https://cpjobling.github.io/eg-247-textbook/fourier_transform/solutions/filters2.pdf)</u>.