Lab 7: Filters

Preamble

Other formats

This document is available in https://cpjobling.github.io/eg-247-textbook/labs/lab07/index for printing.

Acknowledgements

This lab is based on <u>Filter Design Using Matlab Demo by David Dorran</u> (http://dadorran.wordpress.com/2013/10/18/filter-design-using-matlab-demo/).

There is a <u>YouTube video (http://www.youtube.com/watch?</u>

<u>v=vfH5r4cKukg&list=PLJ8LTUMGG9U4vAGind2_Bh4TUfgg1y0F4&feature=</u>

that illustrates what we are going to be using

Aims

This optional lab exercise demonstrates the design and simulation of digital filters. I is not assessed, but you may find it useful preparation for the project.

Setup

Before you start

If you haven't already, create a suitable folder structure on your file-store for your labs.

I suggest

```
P:\workspace
signals-and-systems-lab
lab01
lab02
lab03
lab04
lab05
lab06
lab07
```

Use folder p:\workspace\signals-and-systems-lab\lab07 for this lab.

Preparation

Download the example filter design script filters.m

(https://github.com/cpjobling/eg-247-

<u>textbook/blob/master/portfolio/lab07/filters.m)</u> from this repository. Save it to your folder for lab07.

Open the script as a MATLAB Live Script and execute the embedded code step-by step and read and understand the commentary.

Lab Exercise

Lab Exercise 15: Interactive Filter Design

MATLAB provides a filter design tool with a graphical user interface called fdatool.

We want you to use this tool to design and test a low-pass, band-pass and high-pass Butterworth filter with sampling frequency equal to 44.1 kHz. The filter should implement the first, second and third stage in a three-stage graphic equalizer with a low pass filter with a cut-off frequency of 31.5 Hz, a pass-band filter for the middle filter (f_1 to f_2) of about one octave and centre-frequency f_c equal to 63 hz and a high-pass filter with pass-frequency of 125 Hz.

The aim of this exercise is to determine the order of the Butterworth filters to be used in your design and the Q factor needed (where $Q=f_c/(f_2-f_1)$) for the pass-band filters required to implement the mid-range of your 10-stage graphic equalizer.

The centre pass-band filter should be designed so that f_2 & f_1 satisfies $f_c = \sqrt{(f_1 \, f_2)}$. Your goal is to find the Δf value for this filter that achieves a flat frequency response when it is combined with equal weight to the low-pass and high-pass filters.

What to hand in (EG-3068 Only)

Claim

Up to 3 marks each can be claimed for the design evidenced by a suitable Live Script and filter design file. You should use the filterDesigner from the DSP System Toolbox (https://uk.mathworks.com/products/dsp-system.html? s tid=srchtitle) and save your designs to disk.

Up to 2 marks can be claimed if you have a Simulink model showing the filters set with a gain of 10, 0 and -10 dB respectively. You can start with the model used in the Project Descriptor (../project/) (Three Band EQ Model.slx).

You may find it useful to use the **Filter Realization Wizard** block (part of the Simulink Collection from the DSP System Toolbox) which combines the fiterDesigner with a block that can be used in simulation.

Submission

You should submit the following to the Lab 07: Filters Assignment on Canvas.

- 1. Complete the labwork self-assessment claim form and declaration.
- 2. Evidence of your filter design as a m-file or MLX file.
- 3. Simulink model of your three part filter with gain settings -10dB, 0dB and 10dB.
- 4. The audio file that you used for testing.

Deadline

The deadline for claims and submission is 4:00 pm, 3rd April