

## 6.1 Preliminaries

In this section, the low-pass filter (LPF) is designed. Firstly, you can choose the sampling frequency as **50kHz** and the time duration as **0-10ms** to set the simulation environments. Then, you consider a signal that consists of two sinusoidal with different frequencies such that one has **8kHz** and the other one has **24kHz**. Finally, we can separate these signals via the LPF to obtain lower frequency sinusoidal.

You may use the following codes for LPF;

```
>> freqs = [0 0.4 0.45 1];  
>> amps = [1 1 0 0];  
>> b = firpm(100,freqs,amps);  
>> y = filter(b,1,x);
```

where **x** is the generated signal by yours and **y** is the filtered signal. You can observe the frequency response of these signals by using the Fourier Transform and observe the filter response using `>> freqz(b)` command. You can also find basic knowledge about above built-in functions by simply writing `>> help firpm` and `>> help filter` on the command window in MatLab. Before you start the lab experiment, you had better answer the following questions;

- 1- What is the value of normalized frequency that corresponds to 1 in *freqs* for the above code?
- 2- How do we choose these normalized frequencies in the *freqs*?
- 3- What is the value of cutoff frequency for the LPF in the normalized frequency domain?
- 4- If we want to design a high-pass filter (HPF) to obtain higher frequency sinusoidal from *x*, how can *amps* and/or *freqs* be changed?

## 6.2 Filtering LabWork

- (a) Set the sampling frequency as **1kHz** and the time duration as **0-0.5s**.
- (b) Generate the below signal;

$$x(t) = \sin(200\pi t) + 2\sin(400\pi t) + 0.5\sin(600\pi t)$$

- (c) Design LPF, HPF and band-pass (BPF) to filter for the lowest frequency, for the highest frequency and for the middle frequency components in  $x(t)$ , respectively. Note that you have to use *firpm()* command to obtain these filters as in preliminaries, otherwise you will not get any credit.
- (d) Plot the frequency response of each designed filter.
- (e) Plot the  $x(t)$  and filtered signals which are obtained from (c) and their frequency domain representations in a single figure by using **subplot(42x)** where x indicates the order of each plot.
- (f) Design the ideal filters that includes LPF, HPF and BPF as obtained from (c) and plot them into another figure by using *subplot()* command.