



## Experiment No. # 9

### Digital Filter Design: Windowing, Linear phase FIR

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1) **Aim:**

- a) Digital Filter Design.

2) **Software used:**

- a) MATLAB.

3) **Theory**

- a) J. Proakis and D. Manolakis, Digital signal processing: principles, algorithms, and applications

4) **Procedure**

a) **Windowing Based design: Rectangle, Hamming, Blackman**

- i) Let's consider a low pass filter, whose response is given by

$$H_d(w) = e^{(-j \cdot w \cdot (\frac{M-1}{2}))} \dots, 0 \leq |w| \leq w_c \quad (1)$$

$$H_d(w) = 0, \dots \text{Otherwise.}$$

where M is corresponding delay in time domain.

- ii) The corresponding filter impulse response can be found by

$$h_d(n) = \frac{1}{2\pi} \int_{-w_c}^{w_c} e^{(-j \cdot w \cdot (n - \frac{M-1}{2}))} dw \quad (2)$$

and is given by

$$h_d(n) = \frac{\sin(w_c (n - (\frac{M-1}{2})))}{(\pi (n - (\frac{M-1}{2})))} \quad (3)$$

- iii) This impulse response is windowed using the windows mentioned below.

A) Rectangle Window:  $w(n) = 1$

B) Hamming Window:  $w(n) = 0.42 - 0.5 \cdot \cos(\frac{2\pi n}{M-1}) + 0.08 \cdot \cos(\frac{4\pi n}{M-1})$

C) Blackman Window:  $w(n) = 0.54 - 0.46 \cdot \cos(\frac{2\pi n}{M-1})$

where  $n = 0, 1, \dots, M-1$

- iv) The finite impulse response  $h(n)$  is then windowed impulse response given by

$$h(n) = h_d(n) \cdot w(n) \quad (4)$$

where  $n = 0, 1, 2, \dots, M-1$ .

**b) Filtering Hamming sound using filtering:**

- i) Let's read an audio file and sample it using 32 KHz. Let's also consider this audio file has a hamming sound or Hiss having frequency greater than  $(\frac{\pi}{2})$  8 KHz. Follow the steps and remove the hissing sound.
- ii) Steps:
  - Take 256 samples of the audio file and locate the hissing frequency (Approximately at  $66^{th}$  and  $192^{th}$  sample).
  - Use the low pass filter designed in previous exercise, having cut off frequency  $\frac{\pi}{2}$  i.e (or ideal filter having ones for first 64 samples and zeros for next 64 sample over a span of 0 to  $\pi$  frequency).
  - Multiply the filter response and spectrum of audio signal for  $N = 256$  samples.
  - Repeat above step for the whole audio file.
  - Check whether hiss sound removed and also conclude for distortion audible if any.

**c) Notch Filtering of Hiss:**

- i) Let's read an audio file and sample it using 32 KHz. Let's also consider this audio file has a hissing sound at 8 KHz. Follow the steps and remove the hissing sound.
- ii) Steps:
  - Take 8 samples of the audio file and locate the hissing frequency (Approximately at  $3^{rd}$  and  $7^{th}$  sample).
  - Design a notch filter at frequency  $\frac{\pi}{4}$  i.e (By placing complex conjugate pairs of zeros at  $\pm \frac{\pi}{4}$  frequency).
  - Convolve the impulse response  $h(n)$  with the distorted file Or Multiply the filter response  $H(w)$  ( Having zero values at corresponding frequencies) and spectrum of audio signal for  $N = 8$  samples. Here the transfer function  $H(w)$  and the impulse response  $h(n)$  having complex conjugate zeros at frequencies  $\frac{\pi}{4}$  is given by

$$H(Z) = (1 - \exp(j \cdot \frac{\pi}{4})Z^{-1})(1 - \exp(-j \cdot \frac{\pi}{4})Z^{-1}) = 1 - 2 \cos(\frac{\pi}{4})Z^{-1} + Z^{-2} \quad (5)$$

$$h(n) = [1, -2 \cos(\frac{\pi}{4}), 1] \quad (6)$$

- Check whether hiss sound removed and also conclude for distortion audible if any.

**5) Tasks and Observations:**

- a) Window Based Filtering: Use Eq.(1),(2),(3),(4)
    - i) Simulate various window(rectangle, hamming, blackman's) based Low pass FIR filter with cutoff frequencies  $w_c = \frac{\pi}{2}, \frac{\pi}{4}, \frac{\pi}{6}$ , show the frequency response of the windowing functions and plot the magnitude(in dB), phase response of the designed FIR filter.
    - ii) Simulate various window(rectangle, hamming, blackman's) based High pass FIR filter with cutoff frequencies  $w_c = \frac{\pi}{2}, \frac{\pi}{4}, \frac{\pi}{6}$ , show the frequency response of the windowing functions and plot the magnitude, phase response of the designed FIR filter.
    - iii) Simulate various window(rectangle, hamming, blackman's) based band pass FIR filter with cutoff frequencies  $w_c(low) = \frac{\pi}{4}$ ,  $w_c(high) = \frac{\pi}{2}$ , show the frequency response of the windowing functions and plot the magnitude, phase response of the designed FIR filter.
  - b) Remove the hissing sound using above generated filter.
  - c) Remove the hissing sound using Notch filter.
  - d) Repeat the experiment in simulink.
- 6) Plot all the signals in all of the above steps and justify them.
- 7) **Conclusion** Write/plot your own.