EE3025 Presentation

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June 12, 2021



Outline

Problem

Assignment-01 Assignment-02

Solution

Filter Output y(n)

Matrix Equations

Filter Design

The Bilinear Transform
The FIR Filter



Question

Assignment-01

► Let

$$x(n) = \{\frac{1}{2}, 2, 3, 4, 2, 1\}$$

$$h(n) = \left(-\frac{1}{2}\right)^n u(n) + \left(-\frac{1}{2}\right)^{n-2} u(n-2)$$

- Compute X(k), H(k) and y(n) using FFT and IFFT
- Wherever possible, express all the above equations as matrix equations.

Assignment-02

- Filter Design
 - Filter Design Specifications
 - ► The Bilinear Transform: I
 - The FIR Filter



Computing X(k), H(k) Using FFT

▶ input signal x(n)

$$x(n) = \{1, 2, 3, 4, 2, 1\}$$

► Impulse Response of the System is

$$h(n) = \left(-\frac{1}{2}\right)^n u(n) + \left(-\frac{1}{2}\right)^{n-2} u(n-2)$$

 \triangleright FFT of a Input Signal x(n) is

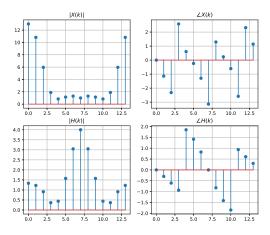
$$X(k) = \sum_{n=0}^{N-1} x(n)e^{-j2\pi kn/N}, \quad k = 0, 1...N-1$$

 \triangleright FFT of a Impulse Response h(n) is

$$H(k) = \sum_{n=0}^{N-1} h(n)e^{-j2\pi kn/N}, \quad k = 0, 1, \dots, N-1$$



Plots of Magnitude and Phase of X(k) and H(k)

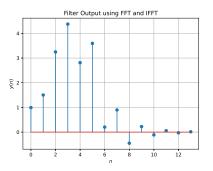




Filter Output Using FFT and IFFT

 \triangleright y(n) can be computed by doing IFFT for Y(k)

$$y(n) = \frac{1}{N} \sum_{n=0}^{N-1} Y(k) e^{j2\pi nk/N}, \quad k = 0, 1, \dots, N-1$$





expressing all the above equations as matrix equations

► FFT of signal X(n)

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n)e^{-j2\pi kn/N}, \quad k = 0, 1, \dots, N-1$$

Let $W_N^{nk} = e^{-j2\pi kn/N}$ then this can be ex- pressed in terms of matrices as:

$$\begin{bmatrix} X(0) \\ X(1) \\ X(2) \\ X(3) \\ X(4) \\ X(5) \end{bmatrix} = \begin{bmatrix} 1 & 1 & 1 & 1 & 1 \\ 1 & W_N^1 & W_N^2 & W_N^3 & W_N^4 & W_N^5 \\ 1 & W_N^2 & W_N^4 & W_N^6 & W_N^8 & W_N^{10} \\ 1 & W_N^3 & W_N^6 & W_N^9 & W_N^{12} & W_N^{15} \\ 1 & W_N^4 & W_N^8 & W_N^{12} & W_N^{16} & W_N^{20} \\ 1 & W_N^5 & W_N^{10} & W_N^{15} & W_N^{20} & W_N^{25} \end{bmatrix} \begin{bmatrix} x(0) \\ x(1) \\ x(2) \\ x(3) \\ x(4) \\ x(5) \end{bmatrix}$$



Computing X(K) from Matrix Equations

On solving we get,

$$\implies X(0) = 13 + 0j,$$

$$X(1) = -4 - 1.732j,$$

$$X(2) = 1 + 0j,$$

$$X(3) = -1 + 0j,$$

$$X(4) = 1 + 0j,$$

$$X(5) = -4 + 1.732j$$

the impluse

$$h(n) = \frac{-1}{2}^{n} u(n) + \frac{-1}{2}^{n-2} u(n-2)$$

assuming that length of h(n) is same as length of x(n) i.e., N=6. Similarly like x(n), solving h(n) using matrix method for each value of k we get,



Computing H(K) from Matrix Equations

$$\begin{bmatrix} H(0) \\ H(1) \\ H(2) \\ H(3) \\ H(4) \\ H(5) \end{bmatrix} = \begin{bmatrix} h(0) + h(1) + h(2) + h(3) + h(4) + h(5) \\ h(0) + h(1)e^{-j\pi/3} + \dots + h(5)e^{-j5\pi/3} \\ h(0) + h(1)e^{-2j\pi/3} + \dots + h(5)e^{-2j5\pi/3} \\ h(0) + h(1)e^{-3j\pi/3} + \dots + h(5)e^{-3j5\pi/3} \\ h(0) + h(1)e^{-4j\pi/3} + \dots + h(5)e^{-4j5\pi/3} \\ h(0) + h(1)e^{-5j\pi/3} + \dots + h(5)e^{-5j5\pi/3} \end{bmatrix}$$

On solving we get

$$H(0)=1.28125+0 j$$
 , $H(1)=0.51625$ - $0.5141875 j$, $H(2)=-0.078125+1.1095625 j$, $H(3)=3.84375+0 j$, $H(4)=-0.071825$ - $1.1095625 j$, $H(5)=0.515625+0.5141875 j$



Computing Y(K) from Matrix Equations

We can now compute Y(k) using below equation

$$Y(k) = X(k)H(k)$$

So, Y(k) is obtained element wise multiplication of X(k) and H(k)

$$\begin{bmatrix} Y(0) \\ Y(1) \\ Y(2) \\ Y(3) \\ Y(4) \\ Y(5) \end{bmatrix} = \begin{bmatrix} 16.6562 + 0j \\ -2.95312 + 1.16372j \\ -0.07812 + 1.10959j \\ -3.84375 - 9.27556j \\ -0.07812 - 1.10959j \\ -2.95312 - 1.16372j \end{bmatrix}$$



Computinh y(n) from IFFT of Y matrix

y(n) is given by IFFT of Y matrix. this y(n) can be calculated by a python code

$$\begin{bmatrix} y(0) \\ y(1) \\ y(2) \\ y(3) \\ y(4) \\ y(5) \end{bmatrix} = \begin{bmatrix} 1.125 + 0j \\ 2.28125071 + 0j \\ 2.6250019 - 1.11022302 \times 10^{-16}j \\ 4.37499667 - 1.47104551 \times 10^{-15}j \\ 2.6562481 + 6.10622664 \times 10^{-16}j \\ 3.59375262 - 1.60982339 \times 10^{-15}j \end{bmatrix}$$



Filter Design Specifications

▶ Tolerances:

The magnitude of resonance in Pass-band is called Pass-band Tolerance and similarly magnitude of resonance in Stop-band is called Stop-band Tolerance.

Passband:

The Frequency band within which signals are transmitted by filter without attenuation.

Stopband:

The Frequency band within which signals are not transmitted by filter or with a large attenuation.



The Bilinear Transform

It is a technique in Signal Processing that is used to Continuous-Time System Representations to Discrete-Time System Representations and vice-versa. The bilinear transform is a first-order approximation of the natural logarithm function that is an exact mapping of the z-plane to the s-plane. When the Laplace transform is performed on a discrete-time signal (with each element of the discrete-time sequence attached to a correspondingly delayed unit impulse), the result is precisely the Z transform of the discrete-time sequence with the substitution of $z=e^{st}$

Small Proof:

$$z = e^{sT} \ s = \frac{1}{T} \ln(z)$$

$$s = \frac{2}{T} \left[\frac{z-1}{z+1} + \frac{1}{3} \left(\frac{z-1}{z+1} \right)^3 + \frac{1}{5} \left(\frac{z-1}{z+1} \right)^5 + \frac{1}{7} \left(\frac{z-1}{z+1} \right)^7 + \cdots \right]$$

$$s = \frac{2}{T} \frac{1-z^{-1}}{1+z^{-1}}$$



The Bilinear Transform

$$\begin{split} H_d(z) &= \left. H_a(s) \right|_{s = \frac{2}{T} \frac{z-1}{z+1}} = H_a\left(\frac{2}{T} \frac{z-1}{z+1}\right) \\ H_d\left(e^{j\omega_d T}\right) &= H_a\left(\frac{2}{T} \frac{e^{j\omega_d T}-1}{e^{j\omega_d T+1}}\right) \\ H_d\left(e^{j\omega_d T}\right) &= H_a\left(\frac{2}{T} \cdot \frac{\left(e^{j\omega_d T/2}-e^{-j\omega_d T/2}\right)}{\left(e^{j\omega_d T/2}+e^{-j\omega_d T/2}\right)}\right) \\ H_d\left(e^{j\omega_d T}\right) &= H_a\left(j\frac{2}{T} \cdot \frac{\sin(\omega_d T/2)}{\cos(\omega_d T/2)}\right) \\ H_d\left(e^{j\omega_d T}\right) &= H_a\left(j\frac{2}{T} \cdot \tan\left(\omega_d T/2\right)\right) \\ \omega_a &= \frac{2}{T} \tan\left(\omega_d \frac{T}{2}\right) \end{split}$$



The FIR Filter

The lowpass filter has a pass frequency ω_l and transition band $\Delta\omega$ with Stop Band Tolerance δ is $h_{lp}(n) = \frac{\sin(n\omega_l)}{n\pi}w(n)$

$$w(n) = \begin{cases} \frac{I\left[\beta N \sqrt{1 - \left(\frac{n}{N}\right)^2}\right]}{I_0(\beta N)}, & -N \le n \le N, \quad \beta > 0 \\ 0 \text{ otherwise} \end{cases}$$

where w(n) is Kaiser Window and $I_0(x)$ is the modified Bessel function of the first kind of order zero in x, and N are the window shaping factors and they are selected as follows

$$N \ge \frac{A-8}{4.57\Delta\omega}$$

$$A = -20\log_{10}\delta$$

$$\beta N = \begin{cases} 0.1102(A-8.7) & A > 50\\ 0.5849(A-21)^{0.4} + 0.07886(A-21) & 21 \le A \le 50\\ 0 & A < 21 \end{cases}$$



The FIR Filter

The FIR Bandpass Filter: The centre of the passband of the desired bandpass filter was found to be $\omega_c=0.275\pi$ The impulse response of the desired bandpass filter is obtained from the impulse response of the corresponding lowpass filter as

$$h_{bp}(n) = 2h_{lp}(n)\cos(n\omega_c)$$

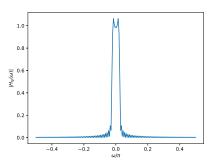


Figure 1: FIR Low Pass Filter



The FIR Filter

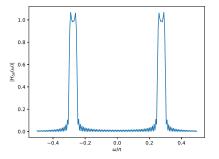


Figure 2: FIR Band Pass Filter

Thank You

