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Fifth Semester B.E. Degree Examination, June/July 2019
Digital Signal Processing

Time: 3 hrs.

Max. Marks:100

Note: Answer FIVE full questions, selecting at least TWO questions from each part.

PART - A

- 1 a. Define DFT and establish the relationship between the Fourier series co-efficient of a continuous time signal and DFT. (04 Marks)
- b. An program to find DFT of a complex valued sequence $x(n)$ is given, how can this program be used to find IDFT of $X(K)$. (03 Marks)
- c. Find the N-point DFT of the sequence:
 $x(n)=a.n$ for $0 \leq n \leq N - 1$ (07 Marks)
- d. If the time-domain expression:

$$W(n) = \frac{1}{2} + \frac{1}{2} \cos \left[\frac{2\pi}{N} \left(n - \frac{N}{2} \right) \right]$$
 What is the DFT of the windowed sequence,
 $y(n) = x(n).W(n)$
 Express the answer in terms of $X(K)$. (06 Marks)
- 2 a. For DFT pair shown, compute the values of the boxed quantities using appropriate properties:
 $\{ \boxed{x_0}, 3, -4, 0, 2 \} \xleftrightarrow{\text{DFT}} \{ 5, \boxed{x_1}, -1.28-j3.49, \boxed{x_3}, 8.78-j1.4 \}$ (03 Marks)
- b. Consider the sequence $x_1(n) = \{0, 1, 2, 3, 4\}$, $x_2(n) = \{0, 1, 0, 0, 0\}$, $S(n) = \{1, 0, 0, 0, 0\}$ and this 5-point DFT's.
 (i) Determine the sequence $y(n)$ so that $Y(K) = X_1(K) \cdot X_2(K)$.
 (ii) Is there a sequence $x_3(n)$ so that $S(K) = X_1(K) \cdot X_3(K)$ (07 Marks)
- c. Two finite sequences $x(n) = [x(0), x(1), x(2), x(3)]$ and $h(n) = [h(0), h(1), h(2), h(3)]$ have DFT's given by:
 $X(K) = \text{DFT}\{x(n)\} = \{1, j, -1, -j\}$
 $H(K) = \text{DFT}\{h(n)\} = \{0, 1+j, 1, 1-j\}$
 Using the properties of DFT, find the following :
 (i) $X_1(K) = \text{DFT}\{h(0), -h(1), h(2), -h(3)\}$
 (ii) $X_2(K) = \text{DFT}\{x(0), h(0), x(1), h(1), x(2), h(2), x(3), h(3)\}$ (06 Marks)
- d. If $x(n) = \{1, 2, 0, 3, -2, 4, 7, 5\}$, evaluate the following : (i) $\sum_{K=0}^7 X(K)$ (ii) $\sum_{K=0}^7 |X(K)|^2$. (04 Marks)

- 3 a. If we perform DFT of an N-length sequence six times, what will be the resulting sequence? (03 Marks)
- b. Find the output $y(n)$ of a filter whose impulse response $h(n) = \{1, -1\}$ and $x(n) = \{1, 1, 1, 1, 1, 3, 1, 1, 4, 2, 1, 1, 3, 1\}$ Using Overlap save method and 5-point circular convolution. (08 Marks)
- c. Explain In-place computation? How many complex additions and multiplications are required for $N = 16$ points, if DFT is computed directly and if FFT is used. (05 Marks)
- d. Compute DFT of the sequence, $x(n) = \cos\left(\frac{n\pi}{2}\right)$ where $N = 4$, using DIT-FFT algorithm. (04 Marks)
- 4 a. Mention any four differences between DIT and DIF-FFT algorithms. (04 Marks)
- b. Given $x_1(n) = \left\{1, \frac{1}{2}, \frac{1}{4}, \frac{1}{8}\right\}$ and $x_2(n) = \{1, 1, 1, 1\}$.
- (i) Compute DFT of $x_1(n)$ using DIT-FFT.
- (ii) Compute DFT of $x_2(n)$ using DIF-FFT.
- (iii) Using above results, compute \otimes_N by IDIF-FFT. (10 Marks)
- c. Consider the sequence $x(n) = u(n) - u(n - 8)$, using chirp z-transform, determine the values $x(Z_0)$ and $X(Z_1)$, where $Z_0 = e^{j\frac{2\pi}{8}}$ and $Z_1 = e^{j\frac{4\pi}{8}}$. (06 Marks)

PART - B

- 5 a. An Butterworth low pass filter has to meet the following specifications:
- (i) Passband gain, $K_p = -1$ dB @ $\Omega_p = 4$ rad/sec.
- (ii) Stop band attenuation greater than or equal to 20 dB @ $\Omega_s = 8$ rad/sec.
- Determine the transfer function $H_a(s)$ of the lowest-order Butterworth filter to meet the above specifications. (10 Marks)
- b. Design a Chebyshev I filter to meet the following specifications :
- (i) Passband ripple : ≤ 2 dB
- (ii) Passband edge : 1 rad/sec
- (iii) Stopband attenuation : ≥ 20 dB
- (iv) Stop band edge : 1.3 rad/sec. (10 Marks)
- 6 a. Realize the following Transfer function,
- $$H(z) = \left\{ \frac{0.7 - 0.25z^{-1} - z^{-2}}{1 + 0.1z^{-1} - 0.72z^{-2}} \right\}$$
- Using Direct form - I, Direct form - II and Cascade form's. (12 Marks)
- b. The Coefficients of a 3-stage FIR lattice structure are given by $K_1 = 0.1$, $K_2 = 0.2$, $K_3 = 0.3$. Find the co-efficients of Direct form - I, FIR filter and draw its block diagram. (08 Marks)

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- 7 a. An analog signal contains frequencies up to 10 kHz. This signal is sampled @ 50 KHz. Design an FIR filter having a linear phase characteristics and a transition band of 5 kHz. The filter should provide minimum 50 dB attenuation at the end of transition band, with respect to Fig. Q7 (a). (10 Marks)

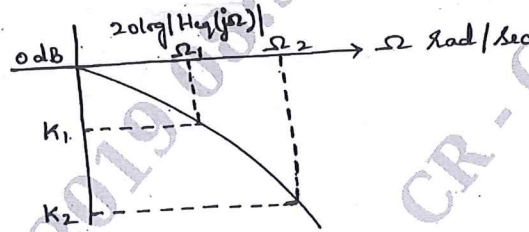


Fig. Q7 (a)

- b. Design a 17-tap linear phase FIR filter with a cut-off frequency $\omega_c = \frac{\pi}{2}$. The design is to be done based on frequency sampling technique. (10 Marks)
- 8 a. An Digital IIR low pass filter is required to meet the following frequency-domain specifications:
 Pass band ripple : ≤ 1 dB
 Pass band edge frequency : 0.33π rad
 Stop band attenuation : ≥ 40 dB
 Stop band edge frequency : 0.5π rad
 The digital filter is to be designed by applying Bilinear transformation on an analog system function. Determine the order, N of Butterworth and Chebyshev filter's needed to meet the specification's in the digital implementation. (10 Marks)
- b. An Third-order Butterworth low pass filter has the transfer function:

$$H(s) = \frac{1}{(s+1)(s^2+s+1)}$$

Design $H(z)$ using Impulse Invariant technique. (10 Marks)

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