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Fifth Semester B.E. Degree Examination, June/July 2023 Digital Signal Processing

Max. Marks: 100

Note: Answer any FIVE full questions, choosing ONE full question from each module.

Module-1

- 1 a. Compute the N-point DFT of the sequence, $x(n) = an, 0 \leq n \leq N-1$. (06 Marks)
- b. Obtain the relationship between DFT and discrete-time Fourier transform of aperiodic signals. (06 Marks)
- c. State and prove the following properties related to N-point DFT:
 - i) Linearity
 - ii) Multiplication of two DFTs. (08 Marks)

OR

- 2 a. Compute the 8-point DFT of the sequence, $x[n] = \{1, 1, 1, 1, 0, 0, 0, 0\}$. (08 Marks)
- b. Compute the N-point DFT of the following:
 - i) $x_1(n) = \cos\left(\frac{2\pi n}{N}\right), 0 \leq n \leq N-1$
 - ii) $x_2(n) = 1, 0 \leq n \leq N-1$. (12 Marks)

Module-2

- 3 a. The 8-point DFT of a length – 8 complex sequence: $V(n) = x(n) + jh(n)$ is given by:
 $V(0) = -2 + j3, V(1) = 1 + j5, V(2) = -4 + j7$
 $V(3) = 2 + j6, V(4) = -1 - j3, V(5) = 4 - j$
 $V(6) = 3 + j8, V(7) = j6$
 Without computing IDFT of $V(K)$ determine the 8-point DFTs of the sequences $x(n)$ and $h(n)$. (08 Marks)
- b. Discuss the filtering of long data sequences using overlap-save method. (06 Marks)
- c. Let $x(n)$ be a finite length sequence with $X(k) = (0, 1 + j, 1, 1 - j)$. Using the properties of DFT, find DFTs of the following sequences:
 - i) $x_1(n) = e^{j\pi/2n} x(n)$
 - ii) $x_2(n) = \cos\left(\frac{\pi}{2}n\right) x(n)$
 - iii) $x_3(n) = x((n-1))_4$. (06 Marks)

OR

- 4 a. Why do we adopt FFT algorithms for the computation of N-point DFT. (04 Marks)
- b. Perform $x(n) * h(n)$, for the sequences $x(n)$ and $h(n)$ given below using overlap-add method:
 $h(n) = (1, 1, 1)$
 $x(n) = (1, 2, 0, -3, 4, 2, -1, 1, -2, 3, 2, 1, -3)$.
 Consider the initial data blocks size to be 6. (08 Marks)
- c. Prove the following :
 - i) $\text{DFT}\{x(n)W_N^{-\ell n}\} = X((K - \ell))_N$
 - ii) $\sum_{n=0}^{N-1} |x(n)|^2 = \frac{1}{N} \sum_{k=0}^{N-1} |X(k)|^2$. (08 Marks)

Module-3

- 5 a. Compute the 8-point DFT of the sequence, $x(n) = \{1, 1, 1, 1, -1, -1, -1, -1\}$, using DIT-FFT algorithm. (08 Marks)
- b. Derive the second order Goertzel filter which can be used for computation of DFT samples. (08 Marks)

Important Note : 1. On completing your answers, compulsorily draw diagonal cross lines on the remaining blank pages.
2. Any revealing of identification, appeal to evaluator and/or equations written eg, 42+8 = 50, will be treated as malpractice.

- c. In the computation of N-point DFT using FFT algorithms, how many
- Complex multiplications are involved?
 - Complex additions are involved?
 - Real registers are used?
 - Butterflies (basic building blocks) are used?

(04 Marks)

OR

- 6 a. Develop DIF-FFT algorithm for the computation of 8-point DFT. (08 Marks)
 b. Write an explanatory note on Chirp-z transform. (07 Marks)
 c. Find the 4-point real sequence $x(n)$, if its 4-point DFT samples are $X(0) = 6$, $X(1) = -2 + j2$, $X(2) = -2$. Use DIF-FFT algorithm. (05 Marks)

Module-4

- 7 a. An LTI system is characterized by the difference equation:
 $y(n] - 0.5y(n-2) - 2y(n-3) = 1.5x(n-2)$.
 Realize the system in i) Direct form - I ii) Direct form - II structures. (08 Marks)
 b. Derive the expressions for order and cut-off frequency related to analog low-pass Butterworth filter. (08 Marks)
 c. An analog filter has the transfer function; $H_a(s) = \frac{1}{s+2}$. Transform $H_a(s)$ to a digital filter, $H(z)$, using impulse invariance technique. Consider the sampling rate to be 2Hz. (04 Marks)

OR

- 8 a. A digital lowpass filter is required to meet the following specifications:
 Monotonic passband and stopband.
 -3.01dB cutoff frequency of 0.5π rad. And, stopband attenuation of at least 15dB at 0.75π rad. Find the system function $H(z)$ using Bilinear transformation. (12 Marks)
 b. Obtain a parallel realization for the transfer function $H(z)$ given below.

$$H(z) = \frac{8 - 4z^{-1} + 11z^{-2} - 2z^{-3}}{\left(1 - \frac{1}{4}z^{-1}\right)\left(1 - z^{-1} + \frac{1}{2}z^{-2}\right)}$$

(08 Marks)

Module-5

- 9 a. A filter is to be designed with the following desired frequency response:

$$H_d(\omega) = \begin{cases} 0, & -\pi/4 < \omega < \pi/4 \\ e^{-j2\omega}, & \pi/4 < |\omega| < \pi \end{cases}$$

Use a rectangular window to design the related FIR filter and find its frequency response. (10 Marks)

- b. Write the relevant equations and stop band attenuations obtained from following windows:
 i) Rectangular ii) Hanning iii) Hamming. (06 Marks)
 c. Realize on FIR filter with system function, $H(z) = 1 + \frac{1}{2}z^{-1} + \frac{1}{4}z^{-2} + \frac{1}{8}z^{-3}$. (04 Marks)

OR

- 10 a. Realize the linear-phase FIR filter having the following impulse response:

$$h(n) = \delta(n) + \frac{1}{4}\delta(n-1) - \frac{1}{8}\delta(n-2) + \frac{1}{4}\delta(n-3) + \delta(n-4).$$

(06 Marks)

- b. A three-stage FIR lattice structure has the following coefficients: $K_1 = 0.65$, $K_2 = -0.34$ and $K_3 = 0.8$. Evaluate its impulse response by tracing a unit impulse $\delta(n)$ at its input through the lattice structure. Also, draw its direct form-I structure. (10 Marks)
 c. The frequency response of an FIR filter is given by $H(\omega) = e^{-j2\omega}(1 + 1.8\cos 2\omega + 1.2\cos \omega)$. Determine impulse response coefficients of FIR filter. (04 Marks)
