



P.E.S. College of Engineering, Mandya - 571 401

(An Autonomous Institution affiliated to VTU, Belagavi)

Sixth Semester, B.E. - Electrical and Electronics Engineering

Semester End Examination; May/June - 2019

Digital Signal Processing

Time: 3 hrs

Max. Marks: 100

Note: Answer **FIVE** full questions, selecting **ONE** full question from each unit.

UNIT - I

- 1 a. Define N-point DFT and IDFT of a sequence and obtain the relation between DFT and Z-transform. 6
- b. Find the 8-point DFT of the sequence $x(n) = \{2, 2, 2, 2\}$. 8
- c. Compute the IDFT of the sequence $X(k) = (2, (1+j), 0, 1-j)$. 6
- 2 a. Compute the N-point DFT of the sequence, 8
 - i) $x(n) = a^n, \quad 0 \leq n \leq N-1$
 - ii) $x(n) = an, \quad 0 \leq n \leq N-1$
- b. State sampling theorem and explain reconstruction of discrete time signals. 6
- c. Find the IDFT of 4-point sequence $X(k) = (4, -j2, 0, 2j)$ using the DFT. 6

UNIT - II

- 3 a. State and prove: i) Linearity Property ii) Circular convolution in time property. 8
- b. Let $x(n) = (1, 2, 0, 3, -2, 4, 7, 5)$. Evaluate the following : 7
 - i) $X(0)$
 - ii) $X(4)$
 - iii) $\sum_{k=0}^7 x(k)$
 - iv) $\sum_{k=0}^7 |x(k)|^2$
- c. Compute the 4-point DFT of the sequence $x(n) = (1, 0, 1, 0)$. Also find $Y(n)$, if $Y(k) = X((k-2))_4$. 5
- 4 a. Find the 4-point circular convolution of the sequence $x_1(n) = \left(\underset{\uparrow}{1}, 2, 3, 1\right)$ and $x_2(n) = \left(\underset{\uparrow}{4}, 3, 2, 2\right)$ 8

using time domain approach and verify the result using frequency domain approach.
- b. State and prove Parseval's theorem. 6
- c. Two length 4 sequences are defined below;

$$x(n) = \cos\left(\frac{\pi n}{2}\right), \quad n = 0, 1, 2, 3$$

$$h(n) = 2^n, \quad n = 0, 1, 2, 3$$
 - i) Calculate $x(n) \otimes_4 h(n)$ by doing circular convolution directly
 - ii) Calculate $x(n) \otimes_4 h(n)$ by doing linear convolution

UNIT - III

- 5 a. In the direct computations of N-point of a sequence, how many; 5
 - i) Complex multiplications
 - ii) Complex additions
 - iii) Real multiplications
 - iv) Real addition
 - v) Trigonometric functions are required
- b. Find 8-point DFT of a sequence $x(n) = \{1, 1, 1, 1, 0, 0, 0, 0\}$ using DIT-FFT radix-2 algorithm. Use butterfly diagram. 8

- c. Given $X(k) = \{20, -5.282 - j2.414, 0, -0.172 - j0.414, 0, -0.172 + j0.414, 0, -5.828 + j2.414\}$. Find $x(n)$ using inverse radix - 2 FFT algorithm. 7
- 6 a. State and prove the following properties of phase factor (twiddle factor) W_N : 4
 - i) Periodicity ii) Symmetry
- b. Given the sequences $x_1(n)$ and $x_2(n)$ below, compute the circular convolution $x_1(n) \otimes_N x_2(n)$ for $N = 4$. Use DIT-FFT algorithm. $x_1(n) = (2, 1, 1, 2)$ and $x_2(n) = (1, 1, 1, 1)$. 8
- c. First five points of the eight points DFT of a real valued sequence is given by $X(0) = 0$, $X(1) = 2 + j2$, $X(2) = -j4$, $X(3) = 2 - j2$, $X(4) = 0$. Determine the remaining points. Hence find the original sequence $x(n)$ using Decimation in frequency FFT algorithm. 8

UNIT - IV

- 7 a. Realize a linear phase FIR filter with the following impulse response. Give the necessary equations: 8

$$h(n) = \delta(n) + \frac{1}{2}\delta(n-1) - \frac{1}{4}\delta(n-2) + \delta(n-4) + \frac{1}{2}\delta(n-3)$$
- b. Realize the following system function : 8

$$H(z) = 1 + \frac{3}{4}z^{-1} + \frac{17}{8}z^{-2} + \frac{3}{4}z^{-3} + z^{-4}$$
 in i) Direct form ii) Cascaded form
- c. Compare direct form-I and direct form-II realizations of IIR filters. 4
- 8 a. The transfer function of a discrete casual system is given as follows : 12

$$H(z) = \frac{1 - z^{-1}}{1 - 0.2z^{-1} - 0.15z^{-2}}$$
 - i) Find the difference equation ii) Draw cascade and parallel realization
 - iii) Calculate the impulse response of the system
- b. A linear time-invariant system is described by the following input-output relation : 8

$$2y(n) - y(n-2) - 4y(n-3) = 3x(n-2)$$
. Realize the system in the following forms :
 - i) Direct form-I realization ii) Direct form-II realization iii) Transposed realization of direct form II

UNIT - V

- 9 a. Design Butterworth filter for the following specifications: 10

$$0.8 \leq |H_a(s)| \leq 1 \text{ for } 0 \leq F \leq 1000\text{Hz} \qquad |H_a(s)| \leq 0.2 \text{ for } F \geq 5000\text{Hz}$$
- b. The system function of an analog filter is given as $H_a(s) = \frac{1}{(s+1)(s+2)}$. Obtain $H(z)$ using impulse invariant method. Take sampling frequency of 5 samples/s. 5
- c. Compare Butterworth and Chebyshev filter approximations. 5
- 10 a. Design the symmetric FIR low pass filter whose desired frequency response is given as, 10

$$H_d(\omega) \begin{cases} e^{-j\omega\tau} & \text{for } |\omega| \leq \omega_c \\ 0 & \text{otherwise} \end{cases}$$

The length of the filter should be 7 and $\omega_c = 1$ radians/sample. Use rectangular window.
- b. Design a normalised linear phase FIR filter having the phase delay of $\tau = 4$ and at least 40 dB attenuation in the stop band. Use Hanning window. 10