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Sixth Semester B.E. Degree Examination, June/July 2024 Signals and Digital Signal Processing

Time: 3 hrs.

Max. Marks: 100

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

Module-1

- 1 a. Distinguish between
- i) Continuous and Discrete time signal
 - ii) Even and odd signal
 - iii) Periodic and Non-periodic signal
 - iv) Energy and power signal
- (08 Marks)
- b. Let $y(t)$ and $x(t)$ are given in Fig Q1(b). Sketch the following signal $z(t) = x(2t) * y(1/2t + 1)$

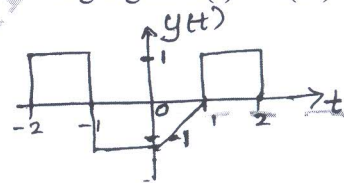
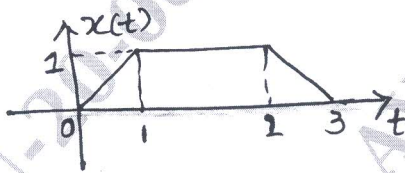


Fig Q1(b)

(06 Marks)

- c. Check whether the following signals are periodic or not. If periodic, find the fundamental period.
- i) $x_1(n) = \cos 2\pi n$
 - ii) $x_2(t) = \cos 2\pi t \cdot \sin 4\pi t$.
- (06 Marks)

OR

- 2 a. Determine whether the following signals are linear, time – invariant memory causal, stable.
- i) $y(n) = x(n^2)$
 - ii) $y(t) = \frac{d}{dt} [e^{-t} x(t)]$.
- (08 Marks)
- b. Evaluate the continuous time convolution integral given below :
- $$y(t) = e^{-2t} u(t) * u(t + 2)$$
- (06 Marks)
- c. Compute the convolution of the sequences.
- $$x(n) = \alpha^n u(n) ; y(n) = \beta^n u(n)$$
- When $\alpha \neq \beta$; and $\alpha = \beta$.
- (06 Marks)

Module-2

- 3 a. Compute 4 point DFT of causal three samples sequence given by
- $$x(n) = \frac{1}{3} ; 0 \leq n \leq 2$$
- $$= 0 ; \text{else}$$
- (06 Marks)
- b. Compute 6-point DFT of the sequence $x(n) = [4, 3, 2, 1, 0, 0]$. Also plot magnitude and phase spectrum.
- (08 Marks)
- c. Prove the following properties of DFT
- i) Linearity
 - ii) Circular time shift.
- (06 Marks)

OR

- 4 a. Consider a FIR filter with impulse response $h(n) = [3, 2, 1, 1]$ if the input is $x(n) = [1, 2, 3, 3, 2, 1, -1, -2, -3, 5, 6, -1, 2, 0, 2, 1]$. Find the output $y(n)$. Use overlap-add method, assuming the length of block is 7. (10 Marks)
- b. Find the IDFT of the given sequence $x(k) = [3, 2 + j, 1, 2 - j]$. (05 Marks)
- c. Perform circular convolution of $x_1(n) = \{2, 1, 2, 1\}$ and $x_2(n) = \{1, 2, 3, 4\}$ using circular shift method. (05 Marks)

Module-3

- 5 a. Develop decimation in time algorithm for finding FFT. Draw signal flow graph for $N = 8$ for DT algorithm. (10 Marks)
- b. Find the 8-point DFT of the following sequence using radix-2 DIF-FFT algorithm. $x(n) = [2, 1, 2, 1]$. (10 Marks)

OR

- 6 a. Tabulate the number of complex multiplications and complex additions required for the direct computation of DFT and FFT algorithm for $N = 8, 16, 32$. (08 Marks)
- b. Find the 8-point DFT of the sequence $x(n) = [1, 1, 1, 1, 0, 0, 0, 0]$ using DIT-FFT radix-2 algorithm. Draw the signal flow graph. (12 Marks)

Module-4

- 7 a. Let $H(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$ represent the transfer function of a low-pass filter with a passband of 1 rad/sec. Use frequency transformation to find the transfer function of the following analog filters.
- A low pass filter with passband of 10 rad/sec
 - A high pass filter with cut-off frequency of 10 rad/sec. (06 Marks)
- b. Compare Butterworth and Chebyshev filter approximations. (04 Marks)
- c. Design a butterworth analog high pass filter that will meet the following specifications :
- Maximum passband attenuation = 2dB
 - Passband edge frequency = 200 rad/sec
 - Minimum stopband attenuation = 20dB
 - Stopband edge frequency = 100 rad/sec. (10 Marks)

OR

- 8 a. Transform $H(s) = \frac{s+a}{(s+a)^2 + b^2}$ into digital filter using impulse invariant technique. (08 Marks)
- b. Design the digital filter using Chebyshev approximation and bilinear transformation to meet the following specifications. Passband ripple = 1dB, for $0 \leq \omega \leq 0.15\pi$ stopband attenuation ≥ 20 dB for $0.45\pi \leq \omega \leq \pi$. (12 Marks)

Module-5

- 9 a. The desired frequency response of the lowpass filter is given by

$$H_d(e^{j\omega}) = H_d(\omega) = \begin{cases} e^{-j3\omega} & ; \quad |\omega| < 3\pi/4 \\ 0 & ; \quad 3\pi/4 < |\omega| < \pi \end{cases}$$

Determine the frequency response of FIR filter if the hamming window is used, with $N = 7$.
(10 Marks)

- b. Design an ideal band pass filter with frequency response

$$H_d(e^{j\omega}) = 1, \text{ for } \pi/4 \leq |\omega| \leq 3\pi/4. \text{ Use rectangular window with } N = 11 \text{ in the design.}$$

(10 Marks)

OR

- 10 a. Obtain the direct form – I and direct form – II, cascade and parallel realizations for the following system.

$$y(n) = 0.75y(n-1) - 0.125y(n-2) + 6x(n) + 7x(n-1) + x(n-2) \quad (10 \text{ Marks})$$

- b. Given the FIR filter with following difference equation

$$y(n) = x(n) + \frac{3}{4}x(n-1) + \frac{17}{8}x(n-2) + \frac{3}{4}x(n-3) + x(n-4). \text{ Draw direct form – I and cascade form.} \quad (06 \text{ Marks})$$

- c. Realize the linear phase filter with the impulse response.

$$h(n) = \delta(n) - \frac{1}{2}\delta(n-1) - \frac{1}{4}\delta(n-2) - \frac{1}{4}\delta(n-2) + \frac{1}{4}\delta(n-3) - \frac{1}{2}\delta(n-4) + \delta(n-5). \quad (04 \text{ Marks})$$
