B.TECH/ ECE/6TH SEM/ ECEN 3202/2018 DIGITAL SIGNAL PROCESSING & APPLICATIONS x(n)={2,4,5,7,0,1}? (ECEN 3202)

Time Allotted : 3 hrs

Full Marks : 70

Figures out of the right margin indicate full marks.

Candidates are required to answer Group A and <u>any 5 (five)</u> from Group B to E, taking <u>at least one</u> from each group.

Candidates are required to give answer in their own words as far as practicable.

Group – A (Multiple Choice Type Questions)

- 1. Choose the correct alternative for the following: $10 \times 1 = 10$
 - (i) The digital system in y[n] = e^{x(n)} is
 (a) linear and time variant
 (b) non linear and time invariant
 (c) non linear and time variant
 (d) linear and time invariant.
 - (ii) The z-transform of $\delta(n-m)$ is (a) z^{-n} (b) 1/(z-n) (c) z^{-m} (d) 1/(z-m).
 - (iii) If $x[n] = \{1,0,0,1\}$, The DFT value of X(0) is (a) 1+j (b) 1-j (c) 0 (d) 2.
 - (iv) The convolution of u(n) and u(n-4) at n=5 is (a) 2 (b) 5 (c) 1 (d) 0.
 - (v) Pole of Chebyshev filter lies on a(a) rectangle(b) circle(c) parabola(d) ellipse.
 - (vi) An FIR filter is
 (a) unstable and linear phase
 (b) stable and linear phase
 (c) stable and non linear phase
 (d) unstable and non linear phase.
 - (vii) An LTI system is said to be causal if and only if

(a) impulse response is non-zero for positive values of n(b) impulse response is zero for positive values of n

- (c) impulse response is non-zero for negative values of n
- (d) impulse response is zero for negative values of n.

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(viii) What is the z-transform of the finite duration signal

(a) $2 + 4z + 5z^2 + 7z^3 + z^4$	(b) $2 + 4z + 5z^2 + 7z^3 + z^5$
(c) $2 + 4z^{-1} + 5z^{-2} + 7z^{-3} + z^{-5}$	(d) $2z^2 + 4z + 5 + 7z^{-1} + z^{-3}$.

- (ix) If x(n) and X(k) are N-point DFT pair, then X(k+N)=?
 (a) X(-k)
 (b) -X(k)
 (c) X(k)
 (d) None of the mentioned.
- (x) Which of the following rule is used in the bilinear transformation?
 (a) Simpson's rule
 (b) Backward difference
 (c) Forward difference
 (d) Trapezoidal rule.

Group – B

- 2. (a) A system has unit sample response h(n) given by $h(n) = -1/4 \delta(n+1) + 1/2 \delta(n) 1/4 \delta(n-1)$. Is the system BIBO stable? Is the filter causal? Justify your answer.
 - (b) Find the step response of the system described by difference equation y(n) 1/6 y(n-1) 1/6 y(n-2) = x(n)
 - (c) Determine the convolution of the following pair of signals $x_1(n) = u(n-2)$ $x_2(n) = \delta(n) + (1/2)^n u(n)$ 3+ 5 + 4 = 12
- 3. An analog signal $x_a(t)=4sin(40\pi t)+5cos(80\pi t)$ is sampled with sampling rate of 400 samples per second to get the discrete time signal x(n).
- (a) What is the Nyquist sampling rate for $x_a(t)$?
- (b) Determine the folding frequency.
- (c) What are the frequencies, in radian, in the resulting discrete time signal x(n)?
- (d) If x(n) is passed through an ideal D/A converter, what is the reconstructed signal $y_a(t)$?

2 + 1 + 4 + 5 = 12

Group – C

- 4.(a) What is FFT? Explain Radix 2 FFT algorithm?
- (b) Derive the expression of DIF algorithm for 8 point DFT.
- (c) Determine 8 point DFT of $x(n) = \{1, 1, 2, 3\}$ using DIF butterfly structure.

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5.(a) Determine the output response y(n) if h(n) ={1,2,1} and x(n) = {2,1,3,4} by i) circular convolution method

ii) circular convolution using zero padding to obtain linear convolution.

(b) Find the DFT of sequence $x(n) = \{1,2,3,4,4,3,2,1\}$ using DIT-FFT algorithm. 6 + 6 = 12

Group – D

- 6. (a) Write down the comparison between Butterworth filter and Chebyshev filter.
 - (b) Derive the expression for order of analog low pass Butterworth filter.
 - (c) For the given specifications design an analog Butterworth filter.

 $\begin{array}{ll} 0.9 \leq |\mathrm{H}(\mathrm{j}\Omega)| \leq 1 & 0 \leq \Omega \leq 0.2\pi \\ |\mathrm{H}(\mathrm{j}\Omega)| \leq 0.2 & 0.4\,\pi \leq \Omega \leq \pi \end{array}$

3 + 3 + 6 = 12

- 7.(a) State some advantages of digital filters. State some disadvantages of impulse invariance technique to design digital filters.
- (b) Design a digital filter with the following specifications: $\alpha_p=4$; $\alpha_s=30$; $f_p=400$ Hz and $f_s=600$ Hz. Where α_p and α_s are pass band and stop band attenuations at the cut off frequencies at f_p and f_s respectively. Use Bilinear Transformation Technique.

4 + 8 = 12

Group – E

- 8.(a) Realize the system given by difference equation y(n) = -0.1y(n-1) + 0.72y(n-2) + 0.7x(n) - 0.252x(n-1) in DirectformII.
- (b) Realize the system function H(Z)= $1/2 + 1/3(Z^{-1}) + Z^{-2} + 1/4(Z^{-3}) + Z^{-4} + 1/3(Z^{-5}) + 1/2(Z^{-6})$. Use cascade form.

6 + 6 = 12

- 9. Write short notes on any two of following.
 - (i) Windowing Technique.
 - (ii) Bilinear Transformation.
 - (iii) Multirate Signal Processing
 - (iv) Architecture of TMS 320C5416

6 * 2 = **1**2