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**Question Paper Code : 80576**

B.E./B.Tech. DEGREE EXAMINATIONS, APRIL/MAY 2024.

Fifth/Eight Semester

Electrical and Electronics Engineering

EE 8591 – DIGITAL SIGNAL PROCESSING

(Common to : Electronics and Instrumentation Engineering/ Instrumentation and Control Engineering)

(Regulations 2017)

Time : Three hours

Maximum : 100 marks

Answer ALL questions.

PART A — (10 × 2 = 20 marks)

1. State the aliasing effect.
2. Consider the analog signal  $X_a(t) = 3 \cos 210 \pi t$ .
  - (a) Determine the minimum sampling rate required to avoid aliasing.
  - (b) Suppose that the signal is sampled at the rate  $F_s = 300$  Hz. What is the discrete-time signal obtained after sampling?
3. Determine the fourier series of the signals
  - (a)  $x(n) = \cos \sqrt{2} \pi n$
  - (b)  $x(n) = \cos \pi n / 5$ .
4. Find the circular convolution of the following two sequences:  
 $x_1(n) = \{1, 3, 4, 1\}$   
    ↑  
 $x_2(n) = \{1, 2, 3, 1\}$   
    ↑
5. Consider the system described by the difference equation  
 $y(n) = ay(n-1) - ax(n) + x(n-1)$   
Obtain the direct form II realization of the system.

6. Illustrate the procedure for low pass to band stop filter transformation.
7. Define zero limit cycle oscillation.
8. A low pass filter has the desired response as given below.

$$H_d(e^{j\omega}) = \begin{cases} e^{-j3\omega} & 0 \leq \omega < \pi/2 \\ 0 & \pi/2 \leq \omega \leq \pi \end{cases}$$

Determine  $H_d(k)$  using frequency sampling technique.

9. Sketch the process involved in Echo cancellation.
10. Differentiate between fixed and floating point DSP Processor.

PART B — (5 × 13 = 65 marks)

11. (a) (i) Compute the convolution  $y(n)$  for  $x(n) = \{1, 2, 3, 2, 1\}$  and  $h(n) = \{1, -2, -2, 1\}$  using graphical method. (8)
- (ii) Determine the stability for each of the following linear systems (5)

$$(1) \quad y_1(n) < \sum_{k=0}^{\infty} \left(\frac{3}{4}\right)^k x(n-k)$$

$$(2) \quad y_2(n) = \sum_{k=0}^{\infty} 2^k x(n-k).$$

Or

- (b) (i) Compute the cross correlation  $r_{xy}(l)$  of the sequences  $x(n) = \{0, 1, -2, 3, -4\}$ ;  $h(n) = \{1, 2, 1, 0.5\}$ . (7)
- (ii) Check whether the given signals are power, energy signal or neither

$$(1) \quad x(n) = j^n + j^{-n} \quad (2)$$

$$(2) \quad x(n) = xj^n \quad (2)$$

$$(3) \quad x(n) = 2^n \cos \pi n \quad (2)$$

12. (a) (i) Determine the Z-transform and ROC for the following sequence

$$x(n) = a^n (\cos \omega_n n) u(n-1). \quad (6)$$

- (ii) Find the Z-transform of the sequence  $x(n) = \frac{1}{2}(n^2 + n)u(n)$ . (7)

Or

- (b) (i) Using partial fraction expansion, determine  $x(n)$  if  

$$X(z) = 1/(1 - 1.5z^{-1} + 0.5z^{-2}). \quad (6)$$

- (ii) Consider the signal  $x(n) = \left(\frac{1}{2}\right)^n u(n) + \left(\frac{-1}{4}\right)^n u(n)$  Determine  $X(z)$  and  $RoC$ . (7)

13. (a) (i) Compute the FFT using DIT algorithm for the sequence given by  
 $x(n) = \{2, 1, 3, 1, 4, 1, 5, 1\}$ . (8)

- (ii) A finite-duration sequence of length  $L$  is given as  

$$x(n) = \begin{cases} 1, & 0 \leq n \leq L-1 \\ 0, & \text{otherwise} \end{cases}$$

Determine the  $N$ -point DFT of this sequence for  $N \geq L$ . (5)

Or

- (b) (i) Compute the inverse using DIF algorithm for the given  
 $X(k) = \{20, -5.828 - j2.414, 0, -0.172 - j0.414, 0, -0.172 + j0.414,$   
 $0, -5.828 + j2.414\}$  (8)

- (ii) An FIR digital filter has the unit impulse response sequence given by  $h(n) = \{1, 2, 1\}$ . Determine the output sequence in response to the input sequence  $x(n) = \{1, 2, 4, 4, 3, 1\}$  using overlap-save method. (5)

14. (a) Design a digital low pass Butterworth filter using bilinear transformation technique for the following specification. (13)

Passband ripple (or peak-to-peak ripple): -0.5 dB

Passband edge: 1.2 kHz

Stopband attenuation: -40 dB

Stopband edge: 2.0 kHz

Or

- (b) The desired response of a low pass filter is

$$H_d(e^{j\omega}) = \begin{cases} e^{-j5\omega}, & -3\pi/4 \leq \omega \leq 3\pi/4 \\ 0, & 3\pi/4 \leq |\omega| \leq \pi \end{cases}$$

Determine  $H(e^{j\omega})$  for  $M=11$  using Hanning window. (13)

15. (a) What are digital signal processors? Explain the architecture with illustrative diagram.

Or

- (b) List the addressing formats and functional modes. Explain their characteristics with examples.

PART C — (1 × 15 = 15 marks)

16. (a) By means of the DFT and IDFT, determine the sequence  $x(n)$  corresponding to the circular convolution of the sequences  $x_1(n)$  and  $x_2(n)$   $x_1(n) = \{1, 2, 1, 2\}$  and  $x_2(n) = \{1, 3, 3, 1\}$ .

Or

- (b) Realize the following system using cascade and parallel realization

$$H(z) = \frac{(z+1)(z+2)(z-2)}{(z-1)(z+0.5)(z-0.2)}$$