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Register No.:

SAINTGITS COLLEGE OF ENGINEERING (AUTONOMOUS)

810A2

Name:

(AFFILIATED TO APJ ABDUL KALAM TECHNOLOGICAL UNIVERSITY, THIRUVANANTHAPURAM)

FIFTH SEMESTER B.TECH DEGREE EXAMINATION (S), FEBRUARY 2023 ELECTRONICS AND COMMUNICATION ENGINEERING

(2020 SCHEME)

Course Code : 20ECT303

Course Name: Digital Signal Processing

Max. Marks : 100

Duration: 3 Hours

(5)

PART A

(Answer all questions. Each question carries 3 marks)

- 1. Derive the relationship between DFT and Z Transform.
- 2. Find the 4 point DFT of sequence $x(n)=\{1,1,0,0\}$.
- 3. Draw the 4 point radix 2 DIF FFT butterfly structure for DFT.
- 4. Bring out the computational advantage of performing an N-point DFT using radix-2 FFT algorithm.
- 5. What is Gibb's phenomenon?
- 6. Give the equation for the order N and cut off frequency of Butterworth filter.
- 7. Determine the direct form realization of FIR system function H(z)=1+2z⁻¹ -3z⁻² -4z⁻³ +5z⁻⁴.
- 8. Obtain the direct form-I realization for the IIR system described by difference equation y(n)=0.5y(n-1)-0.25y(n-2)+x(n)+0.4x(n-1).
- 9. Compare the fixed point and floating point arithmetic.
- 10. Compare Von Neumann and Harvard architecture.

PART B

(Answer one full question from each module, each question carries 14 marks) MODULE I

- 11. a) Perform the circular convolution using concentric circle method of the following sequences $x(n)=\{1,1,2,1\}$; $h(n)=\{1,2,3,4\}$. (9)
 - b) Find the IDFT of $Y(K) = \{1, 0, 1, 0\}$.

OR

- 12. a) Find the linear convolution of the sequences $x[n]=\{2,1,0,1,3,2,0,1,2,2\}$ and $h[n]=\{1,1,1\}$ using overlap save (8) method.
 - b) Explain the following properties of DFT a) Linearity b) Complex conjugate c) Circular convolution. (6)

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MODULE II

13.	Using DIF FFT algorithm compute the 8 point DFT of the sequence		
	$x(n)=1; 0 \le n \le 7$	(14)	
	0; otherwise.		

OR

14. Find the DFT of a sequence $x(n)=\{1,2,3,4,4,3,2,1\}$ using DIT FFT algorithm. (14)

MODULE III

 15. Design a band pass filter using frequency sampling method with the following specifications. Sampling frequency F=8000Hz, Cut off frequencies fc1=1000Hz, fc2=3000Hz. Determine the filter coefficients for N=7.

OR

16. Design a Butterworth filter using the bilinear transformation for the following specifications
0.8 ≤ |H(e^{jw})| ≤ 1; 0 ≤ w ≤ 0.2π (14) |H(e^{jw})| ≤ 0.2; 0.6π ≤ w ≤ π.
Assume T=1 sec.

MODULE IV

17.	a)	Obtain the cascade realization of FIR filter system function	<u> </u>
		$H(z) = 1 + \frac{5}{2} z^{-1} + 2z^{-2} + 2z^{-3}.$	(7)

b) Realize the IIR filter system with difference equation, y(n)=-0.1y(n-1)+0.72y(n-2)+0.7x(n)-0.252x(n-2) in parallel form. (7)

OR

18. Obtain the direct form I, direct form II, cascade form and parallel form realization for the following IIR filter system, (14) y(n)= -0.1y(n-1)+0.2y(n-2)+3x(n)+3.6x(n-1)+0.6x(n-2).

MODULE V

- 19. a) Explain the effect of coefficient quantization in IIR filter structures. (7)
 - b) With an example illustrate the error introduced by truncation and rounding in fixed point representation of numbers. (7)

OR

20. Explain the architecture of TMS320C67xx DSP with block diagram. (14)