

Reg. No.

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B.E. / B.TECH. DEGREE EXAMINATIONS, MAY 2024

Fourth Semester

EC22403 – DISCRETE TIME SIGNAL PROCESSING*(Electronics and Communication Engineering)***(Regulation 2022)****TIME: 3 HOURS****MAX. MARKS: 100**

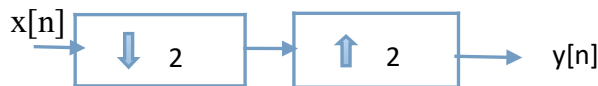
COURSE OUTCOMES	STATEMENT	RBT LEVEL
CO 1	Analyze the frequency spectrum of Discrete time signal using Discrete Fourier Transform and Fast Fourier Transform.	4
CO 2	Interpret the characteristics of FIR filters and articulate the design of Finite Impulse Response filters for filtering undesired signals.	3
CO 3	Observe the IIR filter characteristics and design IIR filters according to the user specifications.	3
CO 4	Assess the word length effects in signal processing systems.	4
CO 5	Explore the architecture of Digital Signal Processor and inspect the various applications of Digital Signal Processing.	3

PART- A (20 x 2 = 40 Marks)

(Answer all Questions)

	CO	RBT LEVEL
1. Obtain the circular convolution of the following sequences, $x[n]=\{1,2,1\}$ and $h[n]=\{1,-2,1\}$	1	3
2. How many additions and multiplications are required to compute 64-point radix-2 FFT algorithm?	1	2
3. If $x(n) = \{1,0,2,3\}$, then $X(k) = \{-----, -1+3j, -----, -----\}$	1	3
4. Compute the twiddle factor, W_{64}^{16}	1	2
5. Give the equations for causal and non-causal Hamming window.	2	2
6. For causal linear phase FIR filter, the partial filter coefficients are $h[0] = 1$, $h[1] = 3$ and $h[2] = 2$. Find the remaining coefficients of $h(n)$. Assume $N=5$.	2	2
7. Determine the frequency response of causal FIR filter with $h[n]=\{0.75,0.25,1,0.25,0.75\}$.	2	2
8. Draw the transversal structure of given system function $H(z) = 1 + 2z^{-1} - 3z^{-2} + 4z^{-3}$	2	3
9. Why is prewarping required in IIR filter design?	3	3
10. What is the advantage of Direct form-II structure over Direct form-I structure?	3	3
11. Interpret the stability condition of an IIR filter.	3	3

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|-----|--|---|---|
| 12. | What are the techniques that can be employed for digitizing the transfer function of an analog filter? | 3 | 2 |
| 13. | Express the fraction $7/8$ and $-7/8$ in sign magnitude and 2's complement representation. | 4 | 3 |
| 14. | What is the effect of quantization on pole locations? | 4 | 2 |
| 15. | Quantize 0.0625_{10} by rounding off to 3 bits and hence calculate the roundoff error. | 4 | 2 |
| 16. | Sketch the product quantization noise model of II order IIR system. | 4 | 3 |
| 17. | If the sampling rate of a sequence $x[n]$ is 20 samples/sec, what will be the sampling rate after decimating by a factor of 4? | 5 | 3 |
| 18. | Compare fixed point and floating point processor. | 5 | 2 |
| 19. | Draw a simple speech recognition model. | 5 | 3 |
| 20. | Find the output of following block diagram. | 5 | 2 |



PART- B (5 x 10 = 50 Marks)

- | | | Marks | CO | RBT LEVEL |
|---------|---|-------|----|-----------|
| 21. (a) | Compute 8-point DFT using DIF-FFT algorithm for the following sequence, | (10) | 1 | 4 |

$$x[n] = \begin{cases} 1 & \text{for } 0 \leq n \leq 7 \\ 0 & \text{for otherwise} \end{cases}$$

(OR)

- | | | | | |
|-----|---|------|---|---|
| (b) | Find the response of LTI system for $x[n] = \{1, 2, -1, 2, 3, -2, -3, -1, 1, 1, 1, 2, -1\}$ and $h[n] = \{1, 2, 1\}$ using overlap-add method | (10) | 1 | 4 |
|-----|---|------|---|---|

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|---------|--------------------------------------|------|---|---|
| 22. (a) | The desired response of a filter is, | (10) | 2 | 3 |
|---------|--------------------------------------|------|---|---|

$$H_d(e^{j\omega}) = 1 \cdot e^{-j3\omega}, \quad -3\pi/4 \leq \omega \leq 3\pi/4$$

$$0, \quad 3\pi/4 \leq |\omega| \leq \pi$$

Identify the filter type and order of the filter and determine the impulse response and frequency response of the filter using Hanning window.

(OR)

(b) Design a linear phase FIR-High pass filter with $\omega_c = 2\pi/3$ rad/sample for $N=7$ using frequency sampling method and determine filter coefficients. (10) 2 3

23. (a) Evaluate $H(z)$ of a digital low pass Butterworth filter using bilinear transformation for the following specifications. (10) 3 3

(i) -3dB cutoff at pass band edge of 0.5π rad/sample

(ii) magnitude down at least 15dB at $\omega = 0.75\pi$ rad/sample

(OR)

(b) Design an IIR analog Chebyshev low pass filter using a Impulse invariant transformation for the following specifications (10) 3 3

$$0.8 \leq |H(e^{j\omega})| \leq 1, \quad 0 \leq \omega \leq 0.2\pi$$

$$|H(e^{j\omega})| \leq 0.2, \quad 0.6\pi \leq \omega \leq \pi$$

24. (a) Consider the transfer function $H(z)=H_1(z).H_2(z)$ where $H_1(z)=1/(1-0.5 z^{-1})$ and $H_2(z)=1/(1-0.6 z^{-1})$. Find the output round off noise power. Assume word length is 4 bits(excluding sign bit). Draw the direct form-II structure. (10) 4 3

(OR)

(b) A digital system is characterized by the difference equation $y[n]=0.5y[n-1]+x[n]$. Determine the dead band of the system when $x[n]=0.875$ for $n=0$ (10) 4 3

$$= 0 \text{ for } n > 0$$

and $y[n]=0$ for $n < 0$. Assume 4- bit sign magnitude representation (Including sign bit)

25. (a) Explain with a neat diagram and mathematical equations, how the sampling rate will be reduced by a factor D . (10) 5 3

(OR)

(b) Explain how digital filters are used for Image processing. (10) 5 3

PART- C (1 x 10 = 10 Marks)

(Q.No.26 is compulsory)

Marks CO RBT LEVEL

- 26.** Compute the response of LTI filter for the following sequences, $x[n] = \{1,2,2\}$ and $h[n]=\{2,-1\}$ using DFT and IDFT method.

(10) 1 4
