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SIDDHARTH INSTITUTE OF ENGINEERING & TECHNOLOGY:: PUTTUR
(AUTONOMOUS)

B.Tech III Year I Semester Regular Examinations March-2023

DIGITAL SIGNAL PROCESSING

(Electronics and Communication Engineering)

Time: 3 hours

Max. Marks: 60

(Answer all Five Units 5 x 12 = 60 Marks)

UNIT-I

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|-----|--|-----|----|----|
| 1 a | Determine the 8 point DFT of the sequence $x(n) = \{1,1,1,1,1,1,1,0\}$. | CO1 | L3 | 8M |
| b | Find the IDFT of the sequence $X(K) = \{1,0,1,0\}$. | CO1 | L3 | 4M |

OR

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|-----|--|-----|----|-----|
| 2 a | List the methods for filtering of long duration sequences. | CO1 | L1 | 2M |
| b | Evaluate the output $y(n)$ of a filter whose impulse response is $h(n)=\{1,2\}$ and input signal $x(n)=\{1,2,-1,2,3,-2,-3,-1,1,1,2,-1\}$ using overlap and save method and overlap add method. | CO1 | L5 | 10M |

UNIT-II

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|-----|--|-----|----|-----|
| 3 a | Compare Analog and Digital filters. | CO2 | L2 | 2M |
| b | Design an analog Butterworth filter that has 2dB pass band attenuation at a frequency of 20 rad/sec and at least 10dB stop band attenuation at 30 rad/sec. | CO2 | L3 | 10M |

OR

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|-----|--|-----|----|----|
| 4 a | Explain the frequency transformation technique in analog domain for converting low pass to low pass filter and low pass to high pass filter with frequency response. | CO3 | L2 | 6M |
| b | Construct the cascade form structure of the system with difference equation
$y(n) = \frac{3}{4}y(n-1) - \frac{1}{8}y(n-2) + x(n) + \frac{1}{3}x(n-1)$ | CO2 | L3 | 6M |

UNIT-III

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|-----|--|-----|----|----|
| 5 a | Design an FIR digital filter to approximate an ideal Low pass filter with pass band gain of unity, cutoff frequency of 1 KHz, and working at a sampling frequency f_s 4 KHz. The length of the impulse response should be 11. Use Fourier series method. | CO3 | L3 | 6M |
| b | Give the equations for Rectangular, Hanning and Hamming window and explain its significance. | CO2 | L2 | 6M |

OR

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|-----|---|-----|----|----|
| 6 a | Write the design steps of FIR filter using Frequency sampling technique. | CO2 | L2 | 6M |
| b | Construct the Direct form realization of system function.
$H(Z) = 1 + 2Z^{-1} - 3Z^{-2} - 4Z^{-3} + 5Z^{-4}$ | CO3 | L3 | 6M |

UNIT-IV

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|-----|--|-----|----|----|
| 7 a | Compare fixed point and floating point arithmetic. | CO4 | L4 | 6M |
| b | Discuss in detail the errors resulting from rounding and truncation. | CO5 | L2 | 6M |

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|-----|--|-----|----|----|
| 8 a | What is meant by Overflow limit cycle oscillations? Explain with example. | CO4 | L2 | 6M |
| b | Explain Signal scaling for second order IIR filter with necessary mathematical | CO4 | L2 | 6M |

expressions.

UNIT-V

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|----|---|--|-----|----|----|
| 9 | a | What are the advantages of the DSP processors over conventional microprocessors? | CO6 | L1 | 4M |
| | b | Illustrate on the various phases of pipelining concept. | CO6 | L3 | 8M |
| OR | | | | | |
| 10 | a | Explain internal memory organization in TMS320C54x architecture. | CO6 | L2 | 6M |
| | b | What are the different buses of TMS320C54x and their functions? | CO6 | L1 | 6M |

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UNIT-I

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|----|---|---|-----|----|-----|
| 1 | a | Determine the DTFT of the sequence $x(n) = [1, 1, 1, 1, 1, 0]$. | CO1 | L3 | 8M |
| | b | Find the IDFT of the sequence $X(K) = [1, 0, 1, 1, 1, 1]$. | CO1 | L3 | 4M |
| OR | | | | | |
| 2 | a | List the methods for filtering of long duration sequences. | CO1 | L1 | 3M |
| | b | Evaluate the output $y(n)$ of a filter whose impulse response is $h(n) = [1, 2]$ and input signal $x(n] = [1, 2, 3, 4, 5, 6]$ using overlap and save method and overlap and method. | CO1 | L3 | 10M |

UNIT-II

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|----|---|--|-----|----|-----|
| 3 | a | Compare FIR and IIR filters. | CO2 | L3 | 3M |
| | b | Design an analog Butterworth filter that has 3dB pass band attenuation at a frequency of 20 rad/sec and at least 18dB stop band attenuation at 40 rad/sec. | CO2 | L3 | 10M |
| OR | | | | | |
| 4 | a | Explain the frequency transformation techniques to convert analog low pass to low pass filter and low pass to high pass filter with converting low pass to low pass filter and low pass to high pass filter with frequency transformation. | CO2 | L3 | 6M |
| | b | Obtain the overall zero structure of the system with difference equation $y(n] = 1/3x[n] - 2/3x[n-1] + 1/3x[n-2] - 1/3x[n-3]$. | CO2 | L3 | 6M |

UNIT-III

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|---|---|---|-----|----|----|
| 5 | a | Design an FIR digital filter to approximate an ideal low pass filter with pass band gain of unity, cutoff frequency of 1 kHz, and working at a sampling frequency of 4 kHz. The length of the impulse response should be 14. Use Fourier series method. | CO3 | L3 | 6M |
| | b | Give the equations for Rectangular, Hanning and Hamming window and explain its significance. | CO3 | L3 | 8M |

OR

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|---|---|--|-----|----|----|
| 6 | a | Write the design steps of FIR filter using frequency sampling technique. | CO3 | L3 | 6M |
| | b | Obtain the Direct form realization of system function $H(z) = 1 + 2z^{-1} - 3z^{-2} - 4z^{-3} + 5z^{-4}$. | CO3 | L3 | 6M |

UNIT-IV

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|----|---|---|-----|----|----|
| 7 | a | Compare fixed point and floating point arithmetic. | CO4 | L4 | 6M |
| | b | Discuss in detail the errors resulting from rounding and truncation. | CO4 | L3 | 6M |
| OR | | | | | |
| 8 | a | Write a program to convert a real number to a fixed point number with a given number of bits. | CO4 | L3 | 6M |
| | b | Explain signal scaling in fixed point DSP filter with necessary mathematical expressions. | CO4 | L3 | 6M |