



SCHOOL OF ELECTRONICS ENGINEERING
CONTINUOUS ASSESSMENT TEST - 2
FALL SEMESTER 2025-2026

SLOT: G1+TG1

Programme Name & Branch : B.Tech. Electronics Engineering
(VLSI Design and Technology)
Course Code and Course Name : BEVD203L SIGNAL PROCESSING
Faculty Name(s) : Dr S KALAIVANI and Dr SHWETA THOMAS
Class Number(s) : VL2025260102586,84
Date of Examination : 11th OCTOBER, 2025 (AN)
Exam Duration : 90 minutes Maximum Marks: 50

General instruction(s):

- Answer All Questions
- M - Max mark; CO - Course Outcome; BL - Blooms Taxonomy Level (1 - Remember, 2 - Understand, 3 - Apply, 4 - Analyse, 5 - Evaluate, 6 - Create)
- Course Outcomes
CO.3. Apply the Fourier concepts to DT signals and Z transform to analyze the DT systems
CO.4. Comprehend the various analog filter design techniques and be able to design digital filters.

Q. No	Question	M	CO	BL
1.	(a). Find the 4-point DFT of the following sequence and also plot the magnitude and phase spectrum. $x[n] = \begin{cases} 1, & 0 \leq n \leq 2 \\ 0, & \text{otherwise} \end{cases}$ (b). If $x(n)=[1, 2, 0, 3, -2, 4, 7, 5]$ evaluate the following. (i). $X(0)$, (ii). $X(4)$, (iii). $\sum_{k=0}^7 X(k)$, (iv). $\sum_{k=0}^7 X(k) ^2$		3	3
2.	Given a sequence $x(n) = n$, for $0 \leq n \leq 7$, determine $X(k)$ using DIT FFT algorithm. $X(k) = [2, -4 + j9.656, -4 + j4, -4 + j1.656, -4, -4 - j1.656, -4 - j4, -4 - j9.656]$	10	3	3
3.	(a). A digital filter with a 3 dB bandwidth of 0.4 is to be designed from the analog filter whose system response is: $H(s) = \frac{\Omega_c}{s + 2\Omega_c}$ Use the bilinear transformation and obtain $H(z)$. (b). The normalized transfer function of an analog filter is given by, $H(s_n) = \frac{1}{s_n^2 + 1.6s_n + 1}$ Convert the analog filter to a digital low pass filter with a cutoff frequency of 0.6π rad/sample, using the bilinear transformation	10	4	3
4.	Find the order and 3-dB cut off frequency of the digital IIR low-pass Butterworth filter with passband edge at 1000 Hz and stopband edge at 1500 Hz for a sampling frequency of 5000 Hz. The filter is to have a passband ripple of 0.5 dB and a stopband ripple below 30 dB using the bilinear transformation.	10	4	3
5.	Determine the system function $H(z)$ of the lowest order Chebyshev IIR digital filter with the following specifications: 3 dB ripple in passband $0 \leq \omega \leq 0.2\pi$ 25 dB attenuation in stopband $0.45\pi \leq \omega \leq \pi$	10	4	3

$$H(s) = \left(\frac{1.01}{s + 1.01} \right) \left(\frac{3.223}{s^2 + 1.015s + 3.232} \right)$$

$$H(z) = \frac{3.25(1+z^{-1})^3}{7.423 - 1.554z^{-1} + 7.023z^{-2}}$$