

**JAYPEE UNIVERSITY OF INFORMATION TECHNOLOGY, WAKNAGHAT**

Supplementary Examination- 2026

B.Tech-VII Semester (CSE/IT)

COURSE CODE(CREDITS): 18B1WEC636 (2)

MAX. MARKS: 75

COURSE NAME: Fundamentals of Digital Signal Processing & Applications

COURSE INSTRUCTORS: Dr. Vikas Baghel

MAX. TIME: 2 Hours

**Note:** (a) All questions are compulsory.

(b) The candidate is allowed to make Suitable numeric assumptions wherever required for solving problems

(c) Use of a standard scientific calculator is allowed.

Q.No	Question	CO	Marks
Q1	a) State the Nyquist Sampling Theorem. Explain the concept of 'Aliasing' and how it can be prevented during the reconstruction of band-limited signals.	CO1	[6]
	b) Determine whether the system $y(n) = x(n) + n x(n-1)$ is:  i. Linear or Non-linear  ii. Time-invariant or Time-variant  iii. Causal or Non-causal		[6]
	c) A discrete-time system is characterized by the impulse response $h(n) = \{1, 2, 1, -1\}$ and an input sequence $x(n) = \{1, 2, 3, 1\}$ .  • Compute the linear convolution $y(n) = x(n) * h(n)$ . • Check if the system is stable and causal.		[6]
Q2	a) Find the Z-transform and the Region of Convergence (ROC) for the sequence:  $x(n) = \left(\frac{1}{2}\right)^n u(n) + \left(-\frac{1}{3}\right)^n u(n)$  • Plot the pole-zero pattern. • Determine the stability of the system based on the ROC.	CO2	[8]

	<p>b) Compute the 4-point DFT of the sequence <math>x(n) = \{1, 1, 0, 0\}</math>. Show the magnitude and phase for each frequency bin <math>X(k)</math>.</p> <p>c) Use the <b>Radix-2 Decimation-In-Time (DIT) FFT</b> algorithm to compute the 8-point DFT of the sequence <math>x(n) = \{1, 2, 1, 2, 0, 0, 0, 0\}</math>. Draw the butterfly signal flow graph clearly showing all twiddle factors <math>W_N^k</math>.</p>		<p>[7]</p> <p>[10]</p>
Q3	<p>a) Compare FIR and IIR filters based on their stability, phase response, and computational complexity.</p> <p>b) Design a digital Low Pass Filter (LPF) using the <b>Bilinear Transformation</b> method to meet the following specifications:</p> <ul style="list-style-type: none"> <li>• Passband edge: 200 Hz</li> <li>• Sampling frequency: 1000 Hz</li> <li>• Use a first-order Butterworth analog prototype <math>H(s) = \frac{\Omega_c}{s + \Omega_c}</math></li> </ul> <p>c) A system is described by the difference equation:</p> $y(n) = 0.5y(n-1) - 0.25y(n-2) + x(n) + 2x(n-1) + x(n-2)$ <ul style="list-style-type: none"> <li>• Obtain the Transfer Function <math>H(z)</math>.</li> <li>• Realize the system using <b>Direct Form II</b> and <b>Cascade Form</b> structures.</li> </ul>	CO3	<p>[4]</p> <p>[8]</p> <p>[10]</p>
Q4	<p>a) Explain the role of Digital Signal Processing in RADAR systems, specifically focusing on how DSP is used for object detection and distance measurement.</p> <p>b) Discuss the application of DSP in Biomedical Engineering, highlighting its use in processing ECG (Electrocardiogram) signals for noise removal.</p>	CO4	<p>[5]</p> <p>[5]</p>