

BTECH
(SEM V) THEORY EXAMINATION 2022-23
DIGITAL SIGNAL PROCESSING

Time: 3 Hours

Total Marks: 100

Note: 1. Attempt all Sections. If require any missing data; then choose suitably.

SECTION A

- 1. Attempt all questions in brief. 2 x 10 = 20**
- a. Determine the linear convolution of the sequences
 $x_1(n)=\{1,2,3,4\}$ and $x_2(n)=\{1,1,2,2\}$
 - b. If $x(n)=\{4,-2,4,-6\}$ find and sketch its odd and even parts with $-2 \leq n \leq 1$.
 - c. Give the statement of Nyquist Sampling Theorem.
 - d. With the help of block diagram illustrate the process of analog to digital conversion.
 - e. Define the properties of convolution in an LTI system.
 - f. Illustrate Twiddle factor and its two properties.
 - g. Differentiate between FIR and IIR filters with example.
 - h. Define frequency warping in Bilinear Transformation method for IIR filter.
 - i. Illustrate the symmetry property and periodicity property of phase factor W_N used for FFT.
 - j. Compute the DFTs of sequence $x(n)=\cos(n\pi/2)$, where $N=4$, using DIF FFT algorithm.

SECTION B

- 2. Attempt any three of the following: 10 x 3 = 30**
- a. (i) Check whether the following discrete time system is static/dynamic, linear/Non-linear, Shift invariant/variant.
 $y(n)=e^{x(n)}$
(ii) Check the stability of filter for $H(Z) = \frac{Z^2 - Z + 1}{Z^2 - Z + \frac{1}{2}}$
 - b. Explain discrete time processing of continuous time signal with the help of block diagram.
 - c. Determine the impulse response for the system given by following difference equation.
 $y(n) = x(n) + 3x(n-1) - 4x(n-2) + 2x(n-3)$
 - d. Explain IIR filter design by bilinear transformation technique. Convert the analog filter into a digital filter whose system function is
$$H(s) = \frac{S + 0.2}{(S + 0.2)^2 + 9}$$
Use the impulse invariant technique. Assume $T=1$ Sec.
 - e. Differentiate between Wavelet Transform and Fourier Transform and also give the applications of Wavelet cosine transform.

SECTION C

3. Attempt any one part of the following: 10 x 1 = 10

a. (i) Consider a LTI system with unit sample response.

$$h(n) = \begin{cases} a^n & n \geq 0, \quad |a| < 1 \\ 0 & n < 0 \end{cases}$$

Find the response to an input of $x(n) = U(n) - U(n - N)$

(ii) Check whether the following system is linear & time invariant.

$$F[x(n)] = a[x(n)]^2 + bx(n)$$

b. Explain any two IIR filter realization methods with suitable example.

4. Attempt any one part of the following: 10 x 1 = 10

a. Derive the expression for sampling theorem and also explain Aliasing.

b. Explain multirate signal processing in detail.

5. Attempt any one part of the following: 10 x 1 = 10

a. Compute circular convolution of the following using graphical method and verify the result using DFT and IDFT.

$$x_1(n) = \begin{bmatrix} \uparrow \\ 1, 2, 3, 4 \end{bmatrix} \quad x_2(n) = \begin{bmatrix} \uparrow \\ 1, 1, 2, 2 \end{bmatrix}$$

b. Determine the magnitude & phase responses for the system characterized by the difference equation

$$y(n) + \frac{1}{2}y(n-1) = x(n) - x(n-1)$$

6. Attempt any one part of the following: 10 x 1 = 10

a. A low pass filter is to be designed with following desired frequency response.

$$h_d(e^{j\omega}) = \begin{cases} e^{-j2\omega}, & -\frac{\pi}{4} \leq \omega \leq \frac{\pi}{4} \\ 0 & \frac{\pi}{4} < |\omega| \leq \pi \end{cases}$$

Determine the filter coefficients $h_d(n)$ if the window function is defined as.

$$w(n) = \begin{cases} 1 & 0 \leq n \leq 4 \\ 0 & \text{otherwise} \end{cases}$$

Also determine the frequency response $H(e^{j\omega})$ of the designed filter.

b. Determine $H(z)$ for a Butterworth filter satisfying the following constraints

$$\begin{aligned} \sqrt{0.5} \leq |H(e^{j\omega})| \leq 1 & \quad 0 \leq \omega \leq \pi/2 \\ |H(e^{j\omega})| \leq 0.2 & \quad 3\pi/4 \leq \omega \leq \pi \end{aligned}$$

With $T=1$ sec. Apply impulse invariant transformation method.

7. Attempt any one part of the following: 10 x 1 = 10

a. Draw the flow graph for the implementation of 8-point DIT FFT of the following sequence

$$x(n) = \{1, 2, 3, 4, 4, 3, 2, 1\}$$

b. Explain radix-2 DIT-FFT algorithm. Compare it with DIF-FFT algorithm.