



Regulation R18 *Subject code: 2P5DC*
TKR COLLEGE OF ENGINEERING AND TECHNOLOGY
(Autonomous, Accredited by NAAC with 'A' Grade)
B.Tech V Semester Supplementary Examinations, January 2023
DIGITAL SIGNAL PROCESSING
(Electronics and Communication Engineering)

Maximum Marks: 70

Duration: 3 Hours

Part-A

All the following questions carry equal marks

(10x2M=20 Marks)

- 1 What is region of convergence (ROC).
- 2 Test the time invariance for the given signal $y(n)=nx(n)$
- 3 What is meant by zero padding?
- 4 Draw the basic butterfly diagram of DIT algorithm
- 5 Give any two properties of Butterworth lowpass filter
- 6 Write the bilinear transform equation between s-plane and z-plane.
- 7 Define Phase delay
- 8 Comment when FIR filter gives linear phase
- 9 Define interpolation process.
- 10 What is overflow oscillations.

Part-B

Answer all the question

(10MX 5=50Marks)

- 11 Define: a) Linear and Non-linear systems b) Causal systems and non-causal systems. [5+5]
OR
- 12 a) Define the LTI system and its need. [4]
b) Explain DSP block diagram. [6]
- 13 Explain the DFT and Find 4-Point DFT of the given time domain sequence $x(n) = \{1, 0, 3, 4\}$.
[10]

OR

- 14 Develop a 8 point DIT FFT algorithm. Draw the signal flow graph for the following sequence $x(n) = \{1, 1, 1, 1, 0, 0, 0, 0\}$ using the signal flow graph. Show all the intermediate results on the signal flow graph.
[10]
- 15 a) Distinguish between Butterworth and Chebyshev filters.
b) Design Chebyshev low pass filter with the given specifications $\alpha_p=2.5\text{dB}$,
 $\alpha_s=30\text{dB}$ at $\Omega_p=20\text{ rad/sec}$, $\Omega_s=50\text{ rad/sec}$.
[4+6]M

OR

- 16 For the given specifications design an analog Butterworth filter

$$0.707 \leq |H(j\Omega)| \leq 1 \quad \text{for } 0 \leq \Omega \leq \pi/2$$

$$|H(j\Omega)| \leq 0.2 \quad \text{for } 3\pi/4 \leq \Omega \leq \pi \text{ with } T=1\text{sec}$$

Using Bilinear transformation technique.

[10]

- 17 a) Distinguish between FIR and IIR filters

- b) Explain briefly method of designing FIR filter using Fourier series method. [4+6]M

OR

- 18 Design an ideal low pass filter whose frequency response

$$H_d(e^{j\omega}) = 1 \quad \text{for } -\pi/2 \leq \omega \leq \pi/2$$

$$= 0 \quad \text{for } \pi/2 \leq \omega \leq \pi$$

Find the values of $h(n)$ for $N=11$. Find $H(z)$ using Hamming Window. [10M]

- 19 a) Write about Dead band effects.

- b) Why rounding is preferred to truncation in realizing digital filters. [5+5]M

OR

- 20 Outline the process of Down sampling and derive its spectrum [10M]