

**GIET UNIVERSITY, GUNUPUR – 765022**

B. Tech (Fifth Semester Regular) Examinations, December – 2023

21BECPC35003 – Digital Signal Processing

(ECE)

Time: 3 hrs

Maximum: 70 Marks

(The figures in the right hand margin indicate marks)

PART – A**(2 x 5 = 10 Marks)**Q.1. Answer **ALL** questions

	CO #	Blooms Level
a. Find the value of $x((n-753))_4$ if $x(n) = \{1, 2, -2, 4\}$.	CO1	K2
b. What is Convolution property?	CO1	K1
c. How many multiplications and additions are required for 32-point DFT?	CO2	K1
d. Why linear phase FIR filters are also called as constant delay filters?	CO3	K1
e. Draw the basic structure of 1 st order digital IIR filter.	CO4	K2

PART – B**(15 x 4 = 60 Marks)**Answer **ALL** questions

	Marks	CO #	Blooms Level
2. a. The DFT of $x(n)$ is described as $X(K) = \{0, -1-i, 6, -1+i\}$. Find the DFT of $x^2(n)$	15	CO1	K3
(OR)			
b. Calculate the linear convolution using Overlap Save method if $x(n) = \{1, 2, 3, 4, 4, 3, 2, 1, 0, 5\}$ & $h(n) = \{1, -3, 4\}$	15	CO1	K3
3.a. Find the 8-point DFT using DIF-FFT algorithm if $x(n) = \{1, -2, 2, 3, -1, 4\}$.	15	CO2	K3
(OR)			
b. Find the DFT of the given signals using 4-point DFT if $g(n) = \{1, 2, 0, 1\}$ and $h(n) = \{2, 2, 1, 1\}$.	15	CO2	K3
4.a. Design a FIR filter using frequency sampling of length 7 if the desired frequency response is	15	CO3	K3
$H_d(\omega) = e^{-j2\omega}$ for $-\pi/2 \leq \omega \leq \pi/2$			
$= 0$ Otherwise			
(OR)			
b. Determine the coefficient of linear phase FIR filter length $M=15$, which has a symmetric unit sample response and frequency response that satisfies the condition	15	CO3	K3

$$\begin{aligned} \text{Hr}\left(\frac{2\pi K}{15}\right) &= 1 & K &= 0, 1, 2, 3 \\ &= 0.4 & K &= 4 \\ &= 0 & K &= 5, 6, 7 \end{aligned}$$

- 5.a. If $H(z) = \frac{3+3.6z^{-1}+0.6z^{-2}}{1+.01z^{-1}-0.2z^{-2}}$ then draw the Parallel form structure using 1st order and 2nd order transfer function 15 CO4 K3

(OR)

- b. Design a digital IIR filter using impulse invariant method and the system function is 7 CO4 K3

$$H(s) = \frac{3}{(s+1)(s-5)}$$

- c. Convert the analog filter with system function

$$H(s) = \frac{s+0.1}{(s+0.1)^2+9}$$

- into a digital IIR filter using impulse invariance transformation method. The digital filter is to have resonant frequency of $\pi/2$. 8 CO4 K3

--- End of Paper ---