

**GURU NANAK INSTITUTE OF TECHNOLOGY**  
**An Autonomous Institute under MAKAUT**  
**2021**  
**DIGITAL SIGNAL PROCESSING**  
**EE605A**

TIME ALLOTTED: 3HR

FULL MARKS:70

*The figures in the margin indicate full marks.**Candidates are required to give their answers in their own words as far as practicable***GROUP – A****(Multiple Choice Type Questions)**Answer any *ten* from the following, choosing the correct alternative of each question: **10×1=10**

		<b>Marks</b>	<b>CO No</b>
1. (i)	The digital system in $y(n) = n.x(n)$ is a) linear and causal. b) linear and non causal. c) non linear and causal. d) non linear and non causal.	1	CO2
(ii)	The Z-transform of $a^n u[-n-1]$ is a) $1/(1-aZ^{-1})$ b) $Z/(a-Z)$ c) $a1/[aZ(1-Z^{-1})]$ d) $(1+aZ^{-1})$	1	CO2
(iii)	The ROC of an infinite causal sequence is the a) interior of a circle. b) exterior of a circle. c) entire z plane except $z = 0$ . d) entire z plane except $z = \infty$ .	1	CO2
(iv)	The s plane and z plane are related as a) $z = e^{sT}$ b) $z = e^{2sT}$ c) $z = 2e^{sT}$ d) $z = e^{sT}/2$	1	CO2
(v)	Zero padding of a signal a) reduces aliasing. b) increase time resolution. c) increase frequency resolution. d) has no effect.	1	CO4
(vi)	DFT is applied to a) Infinite sequences b) Continuous finite sequences c) Finite discrete sequences d) Periodic continuous time signals	1	CO1
(vii)	FIR filter is a) recursive and linear. b) non recursive and linear. c) recursive and non linear. d) none of these.	1	CO5

(viii)	For rectangular window used for designing FIR filters, the peak amplitude of the side lobe is a) - 41 dB b) - 3 dB c) 0 dB d) - 13 dB	1	CO5
(ix)	Overlap save method is used to find a) circular convolution. b) linear convolution c) DFT d) Z- transform.	1	CO4
(x)	The speech signal is obtained after a) Analog to digital conversion b) Digital to analog conversion c) Modulation d) Quantization	1	CO5
(xi)	Poles of Butterworth filter lie on a) Circle b) Ellipse c) Circle and Ellipse d) none of these.	1	CO3
(xii)	Infinite memory system is also known as a) FIR system b) IIR system c) Digital system d) Analog system	1	CO3

**GROUP – B**

**(Short Answer Type Questions)**

Answer any *three* from the following: **3×5=15**

		<b>Marks</b>	<b>CO No</b>
2.	Find the DFT of the sequence $x(n) = \{1, 1, 0, 0\}$	5	CO1
3.	For the following system determine whether it is linear, causal and time invariant $y(n) = x(n) + 3u(n-1)$	5	CO2
4.	What do you mean by Quantization and how it affects on digital filter?	5	CO4
5.	Find the circular convolution using concentric circle method of the two finite duration sequence $x_1(n) = \{1, -1, -2, 3, -1\}$ and $x_2(n) = \{1, 2, 3\}$ .	5	CO4
6.	A DTLTI system with impulse response $h[n] = \{1, 1, 1\}$ is excited by a sequence $x[n] = \{4, 3, 2, 1\}$ . Determine the output $y[n]$ of the system.	5	CO2

**GROUP – C**

**(Long Answer Type Questions)**

Answer any *three* from the following: **3×15=45**

		<b>Marks</b>	<b>CO No.</b>
7.	(a) Prove that the energy of a real valued energy signal is equal to the sum of the energies of the	4	CO1

		even and odd components i.e. $E_s = E_e + E_o$		
	(b)	Find the inverse transform of $X(z) = \frac{z}{3z^2 - 4z + 1},$ , where the ROC is (a) $ z  > 1$ , (b) $ z  < (1/3)$ , (c) $(1/3) <  z  < 1$	6	CO2
	(c)	If $x(n) = \{1, -2, 1\}$ and $h(n) = \{1, 1, 1, 1, 1\}$ , then find $x(n)*h(n)$ using Z-transform method.	5	CO2
8.	(a)	$x(n) = \{1, 2, -1, 2, 3, -2, -3, -1, 1, 1, 2, -1\}$ , $h(n) = \{1, 2\}$ . i. Using linear convolution find $y(n)$ . ii. Compare the result by solving the problem using overlap add method.	8	CO4
	(b)	Determine the inverse Z-transform of $X(z) = \frac{z+2}{2z^2 - 7z + 3}.$ If the ROCs are a. $ z  > 3$ , b. $ z  < 1/2$ , c. $1/2 <  z  < 3$ .	7	CO2
9.	(a)	Using linear convolution find $y(n) = x(n)*h(n)$ for the sequence $x(n) = (1, 2, -1, 2, 3, -2, -3, -1, 1, 1, 2, -1)$ . and the impulse response of the system is $h(n) = (1, 2)$ . Compare the result by solving the problem using (a) Overlap -save method (b) Overlap-add method.	10	CO3
	(b)	Write short notes on the following : ( Any one ) i. Recursive and non-recursive systems. ii. Aliasing and sampling rate in signal processing. iii. TMS320C 5416/6713 processor	5	CO2, CO1
10.	(a)	Design a Butterworth Filter using Impulse invariant method for the following specifications. Given $T = 1$ sec. $\begin{array}{ll} 0.8 \leq  H(e^{j\omega})  \leq 1 & 0 \leq \omega \leq 0.2\pi \\  H(e^{j\omega})  \leq 0.2 & 0.6\pi \leq \omega \leq \pi \end{array}$	10	CO5
	(b)	Find the order of the Butterworth Filter that has a -2 dB passband attenuation at a frequency of 20 rad/sec and -10 dB stopband attenuation at 30 rad/sec.	5	CO3
11.	(a)	A lowpass filter is to be designed with the following desired frequency response $H_d(e^{j\omega}) = e^{-j2\omega} \quad -\pi/4 \leq \omega \leq \pi/4$ $= 0 \quad \pi/4 < \omega \leq \pi$ Determine the filter coefficient $h_d(n)$ if the window function is defined as $w(n) = 1 \quad 0 \leq n \leq 4$	10	CO5

$$= 0 \quad \text{otherwise}$$

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

- (b) Given sequences  $x_1(n) = \{1, 2, 3, 1\}$  and  $x_2(n) = \{4, 3, 2, 2\}$ , find  $x_3(n)$  such that  $X_3(k) = X_1(k) X_2(k)$  5 CO1, CO4