

Code No: **RT42023B**
R13
Set No. 1
IV B.Tech II Semester Regular/Supplementary Examinations, April - 2018
DIGITAL SIGNAL PROCESSING
(Electrical and Electronics Engineering)
Time: 3 hours
Max. Marks: 70
Question paper consists of Part-A and Part-B
Answer ALL sub questions from Part-A
Answer any THREE questions from Part-B

PART-A (22 Marks)

1. a) Determine the range of values of “P” and “q” for the stability of LTI system with impulse response $h(n)=p^n; n<0$
 $=q^n; n\geq 0$ [4]
- b) Discuss the periodicity and linearity property of DFT. [4]
- c) Explain the need for going to FFT rather than DFT. [3]
- d) What is the drawback in FIR filter design using windows and frequency sampling method? How is it overcome? [4]
- e) Consider the discrete time signal $x(n)=\{1,2,3,4\}$ determine the up sampled version of the signals for the sampling rate multiplication factor i) $I=2$ ii) $I=3$. [4]
- f) Explain the concept of pipelining and mention its importance in DSP processors. [3]

PART-B (3x16 = 48 Marks)

2. a) Test the following system for linearity

$$y(n) = \sum_{m=0}^M b_m x(n-m) - \sum_{m=1}^N c_m y(n-m)$$
 [6]
- b) Find the system response of the following difference equation $y(n) - \frac{7}{12}y(n-1) + \frac{1}{12}y(n-2) = 2$ for $n \geq 0$, when signal $x(n)=2u(n)$ assume initial conditions are $y(-1)=2$ and $y(-2)=3$. [10]
3. a) Explain the need of zero padding in DFT Sequence. Compute the 8-point of DFT of the following sequence $x(n)=\{1,1,1,1,0,0,0,0\}$. [8]
- b) Compute circular convolution of the following two sequences using DFT.
 $x_1(n)=\{0,1,0,1\}$ and $x_2(n)=\{1,2,1,2\}$. [8]
4. a) Find the inverse FFT of the given $X(k)=\{1,2,3,4\}$ using DIF algorithm. [8]
- b) Compute the 8-point DFT of the sequence $x(n)=\cos(n\pi/2)$ using the DIT-FFT algorithm. Show all intermediate results. [8]
5. a) Show that decimator and interpolator are time-invariant systems. [8]
- b) Discuss the sampling rate conversion by a rational factor I/D . [8]

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6. a) Design a linear phase FIR low pass filter using rectangular window by taking 7 samples of window sequence and with cutoff frequency $\omega_c=0.2\pi$ rad/sample. [8]
- b) Derive the relation between analog and digital filter poles in impulse invariant transformation method. [8]
7. a) Explain with neat sketch the Architecture of TMS 320C5X. [10]
- b) Write any four special instructions of TMS320C5X processors that are suitable for signal processing applications and explain. [6]