

**COURSE TITLE** : **DIGITAL SIGNAL PROCESSING**  
**COURSE CODE** : **6044**  
**COURSE CATEGORY** : **E**  
**PERIODS/WEEK** : **4**  
**PERIODS/SEMESTER** : **60**  
**CREDITS** : **4**

**TIME SCHEDULE**

| MODULE       | TOPICS                   | PERIODS   |
|--------------|--------------------------|-----------|
| 1            | Signals and Systems      | 15        |
| 2            | Fourier and Z Transform  | 15        |
| 3            | Fast Fourier Transform   | 15        |
| 4            | Digital Signal Processor | 15        |
| <b>TOTAL</b> |                          | <b>60</b> |

**Course General Outcome:**

| MODULE | G.O. | ON COMPLETION OF THE STUDY OF THIS COURSE THE STUDENTS WILL BE ABLE: |
|--------|------|--|
| 1      | 1    | To understand different types of signals and signal operations       |
|        | 2    | To know different classification of systems                          |
| 2      | 3    | To understand Fourier and Z Transform                                |
| 3      | 4    | To understand Fast Fourier Transform                                 |
| 4      | 5    | To understand Digital signal Processor                               |

G.O - General Outcome

**On the completion of the study the student will be able:**

**MODULE I SIGNALS AND SYSTEMS**

**1.1.0 To understand different types of signals and signal operators**

- 1.1.1 To describe Discrete time signals
- 1.1.2 To define unit impulse, unit step , ramp and exponential signals
- 1.1.3 To describe Sinusoidal signals, periodic and aperiodic signals
- 1.1.4 To describe Even and odd signal, causal and non causal signal
- 1.1.5 To explain Shifting, time reversal and time scaling operator
- 1.1.6 To explain Scalar multiplication and signal multiplication
- 1.1.7 To explain Addition operator

**1.2.0 To know different classification of systems**

- 1.2.1 To explain discrete time system, linear and non linear, causal and non causal system
- 1.2.2 To describe Time variant and time invariant system
- 1.2.3 To define LTI system
- 1.2.4 To explain the block diagram of a DSP system
- 1.2.5 To list the advantage of DSP system

## **MODULE II FOURIER AND Z TRANSFORM**

### **2.1.0 To understand Fourier and Z Transform**

- 2.1.1 To Describe discrete Fourier series
- 2.1.2 To describe discrete Fourier transform
- 2.1.3 To find DFT of unit impulse and other dc signal, sinusoidal signal
- 2.1.4 To list properties of DFT
- 2.1.5 To state Linearity and periodicity, circular shift and symmetry
- 2.1.6 To explain circular convolution
- 2.1.7 To define Z Transform
- 2.1.8 To discuss Z transform of unit step, unit impulse and sinusoidal signal
- 2.1.9 To discuss properties of Z transform; linearity, left shift ,right shift
- 2.1.10 To explain convolution, Multiplication by  $a^n u(n)$ , initial value theorem and final value theorem
- 2.1.11 To discuss inverse Z transform
- 2.1.12 To list the types inverse Z transform
- 2.1.13 To find inverse Z transform by partial fraction method – simple problems

## **MODULE III FAST FOURIER TRANSFORM**

### **3.1.0 To understand Fast Fourier Transform**

- 3.1.1 To explain Decimation in time
- 3.1.2 To discuss 4 point and 8 point FFT using radix 2 DIT block diagram
- 3.1.3 To draw 8 point FFT using radix 2 DIT butterfly diagram
- 3.1.4 To explain decimation in frequency
- 3.1.5 To discuss 4 point and 8 point FFT using radix 2 DIF block diagram
- 3.1.6 To draw 8 point FFT using radix 2 DIF butterfly diagram

## **MODULE IV DIGITAL SIGNAL PROCESSOR**

### **4.1.0 To understand Digital signal Processor**

- 4.1.1 To describe FIR filters
- 4.1.2 To explain about FIR filter coefficients
- 4.1.3 To describe about FIR windows
- 4.1.4 To describe IIR filters
- 4.1.5 To explain about IIR filter coefficients
- 4.1.6 To discuss the overview of Digital signal processors
- 4.1.7 To describe the selection of digital signal processor
- 4.1.8 To explain architectural features, execution speed, type of arithmetic, word length
- 4.1.9 To explain the Texas DS processor TMX320c50 DSP
- 4.1.10 To describe the addressing Modes of TMX320c50 DSP
- 4.1.11 To list the DSP applications

## CONTENTS

### MODULE I SIGNALS AND SYSTEMS

Classification of signals- discrete time signals- unit step –unit ramp-unit impulse, exponential sequence, sinusoidal signal - periodic and non periodic signals, even and odd signals - causal and non-causal signal, operation of signal –shifting, time reversal, time scaling, scalar multiplication, signal multiplier, addition, discrete time system – classification – linear and nonlinear system – causal and non-causal system – time variant and time invariant system, LTI system - block diagram of a DSP system – advantages of a DSP system

### MODULE II FOURIER AND Z TRANSFORM

Fourier series- fundamentals - Discrete Fourier transform- DFT of unit impulse, unit step signal, sinusoidal signal, properties of DFT- linearity- periodicity- circular shift – symmetry property- circular convolution of time domain signal- circular convolution of frequency domain, Z- transform- Z-transform of unit step- unit impulse – sinusoidal, property of z- transform- linearity – left shift of a signal- right shift of a signal- convolution – multiplication by  $a^n u(n)$  – initial value theorem – final value theorem, inverse z- transform

### MODULE III FAST FOURIER TRANSFORM

Decimation in Time(DIT) – 4 point and 8 point FFT using radix 2 DIT FFT- flow graph for 8 point DFT, Decimation in Frequency(DIF)- 4 point and 8 point FFT using radix 2 DIF FFT, comparison of DIT and DIF.

### MODULE IV DIGITAL SIGNAL PROCESSOR

Filters – types - FIR filter – coefficients - FIR windows, IIR filters – coefficients, Overview of a digital signal processor – selecting a digital signal processor- architecture of Texas Instruments- TMX320c50 DSP – CPU – Central ALU – Parallel Logic Unit – Auxiliary Register -Arithmetic unit- index Register- Aux. Reg Compare Reg. - Block move address register – status register- program controller- program counter – on chip memory – on chip peripherals, addressing modes, Applications of DSP

### TEXT BOOK

1. Digital Signal Processing – Salivahanan – TMH - 2<sup>nd</sup> Edition
2. Digital Signal Processing - P Ramesh Babu, Scitech - 4<sup>th</sup> Edition

### REFERENCE

1. Digital Signal Processing – Nagoor Kani – TMH
2. Digital Signal Processing – Alan V Oppenheim, Ronald W Schafer (Pearson)
3. A Text book of Digital Signal Processing – R S Kaler, M Kulkarni, Umesh Gupta (I K International publishing company, NewDelhi)