4/4 B.Tech - SEVENTH SEMESTER

EC7L2

Digital Signal Processing Lab

Credits: 2

Lecture	:	Internal assessment: 25 marks
Lab :	3 periods/week	Semester end examination: 50 marks

Course Objectives:

- To perform DSP algorithms like convolution, DFT & FFT in software using a computer language such as C with TMS320C6713 floating point Processor.
- To design the digital filter types like IIR-Butterworth, Chebyshev, Bilinear, Impulse invariant filters and FIR window-design methods using MATLAB.
- To gain a working knowledge of the design, implementation, and debugging of real time DSP algorithms written in C language or MATLAB for an industry-standard DSP processor.

Learning Outcomes:

The students undergoing this course will be able to

- Design & implement the digital active/passive filter in C and MATLAB programming environments
- Program a DSP chip with a variety of real-time signal processing algorithms such as filtering for noise reduction or digital audio effects
- compute and analyze signal spectrum of digital system using DFT/FFT algorithms in MATLAB
- generate waveforms using digital filter(s)
- develop & implement a real-time DSP project

List of Experiments:

Part – A: (Using MATLAB)

- 1. Sum of Sinusoidal signals.
- 2. Frequency response of a system given in difference equation form.
- 3. Fast Fourier Transform.
- 4. Determination of Power Spectrum.
- 5. IIR Low pass filter design.
- 6. IIR High pass filter design.
- 7. FIR Low pass filter design.
- 8. FIR High pass filter design.
- 9. Implementation of Interpolation and Decimation.

Part – B: (Using Code Composer Studio)

- 10. Linear Convolution.
- 11. Circular Convolution.
- 12. Generation of Sine wave & Square wave.
- 13. Generation of Real time Sine wave.
- 14. FIR filter design.
- 15. IIR filter design.