

Advanced Digital Signal Processing with MATLAB® (15 + 25 hours (40 hours))

This course mainly deals with using MATLAB® Signal Processing toolbox for Digital signal processing, analysis, visualization, and algorithm development. The training covers various topics such as filter design, windowing techniques, transforms, multi-rate signal processing, statistical signal processing, parametric modeling etc.

COURSE CONTENT :

Introduction to DSP (2 hours)

- Introduction to DSP
- Sampled data systems
- Aliasing and antialiasing
- Reconstruction
- Practical limitations
- Frequency & amplitude resolution
- Quantization and timing errors
- Correlation and convolution
- Frequency analysis
- Fourier transforms
- Frequency 'leakage'
- Windowing
- Multi-rate signal processing

Transforms (2 hours)

- Fourier Transform
- Z – Transform
- DCT Transform
- Hilbert Transform
- Wavelet Transform

Filters (5 hours)

- FIR Filter - FIR digital Filters
- FIR filter basics
 - Analysis of FIR filters
 - Frequency & impulse responses
 - The window design method
 - Optimization design methods
 - Practical limitations of FIR filters
- IIR Filter -
- IIR filter basics

- Analysis of FIR filters
 - Frequency & impulse responses
 - IIR filter design
 - Poles, zeroes and filter response
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**Cepstral analysis
(1 hour)**

- Complex Cepsturm
 - Inverse complex cepstrum
 - Real cepstrum and minimum phase reconstruction
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**Statistical signal processing
(3 hours)**

- Introduction to statistical parameters
 - Autocorrelation matrix
 - Power spectral density (PSD)
 - Cross power spectral density
 - Finding PSD using various Methods (periodogram, modified periodogram, covariance, Eigen vector, burg, yule walker, Welch, MUSIC Algorithm, Root MUSIC Algorithm)
 - Spectrogram
 - Transfer function estimation
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**Parametric modeling
(2 hours)**

- Introduction to signal modelling
 - Study of Auto Regressive Moving Average Models (ARMA), AR Models and MA models
 - Estimation of model parameters using various methods like Yule-Walker, prony etc)
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**DSP with MATLAB®
(3 hours)**

- Introduction to DSP Toolbox
 - Signal processing functions in MATLAB® (conv, conv2, corrcoef, cov, cplxpair, deconv, fft, fft2, fftshift, filter2, freqspace, ifft, ifft2,unwrap)
 - Time domain analysis of a signal
 - Frequency domain analysis of a signal
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**Digital Filter Design in
MATLAB®
(2 hours)**

- Discrete-Time Filters (Direct form I, Direct form II, lattice filters)
 - 1_D Median filtering
 - Butterworth filter design
 - Chebyshev Type I filter design (pass band ripple)
 - Chebyshev Type II filter design (stop band ripple)
 - Raised cosine FIR filter design
 - Recursive digital filter design
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**Window Design
(2 hour)**

- Rectangular window
- Hamming window
- Hanning window

- Bartlett window
 - Kaiser window etc
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**Transforms
(2 hour)**

- Discrete fourier transform
 - Discrete cosine transform
 - Hilbert transform
 - Discrete wavelet transform
 - inverse transforms
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**Multi-rate Signal Processing
(2 hours)**

- Decimation
 - Interpolation
 - Up-Sampling
 - Down-Sampling
 - Re-Sampling
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**Linear Systems
(1 hour)**

- Stabilize polynomial
 - z-transform partial-fraction expansion
 - conversion of digital filter parameters to transfer function form/
pole-zero form etc
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**Cepstral analysis
(1 hour)**

- Complex cepstral analysis
 - Inverse complex cepstrum
 - Real cepstrum and minimum phase reconstruction
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**Statistical signal processing
(4 hours)**

- Cross Correlation
 - Covariance
 - Data matrix for autocorrelation matrix estimation
 - Power spectral density (PSD)
 - Cross power spectral density
 - Finding PSD using various Methods (periodogram, modified
periodogram, covariance, Eigen vector, burg, yule walker, Welch,
MUSIC Algorithm, Root MUSIC Algorithm)
 - Spectrogram
 - Transfer function estimation
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**Parametric Modeling
(4 hours)**

- Autoregressive (AR) all-pole model parameters estimated using
Burg method
- Estimate AR model parameters using covariance method
- Estimate AR model parameters using modified covariance
method
- Estimate autoregressive (AR) all-pole model using Yule-Walker
method
- Cross power spectral density

- Prony method for filter design
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**Waveform Generation
(30 min)**

- Swept-frequency cosine
 - periodic sinc function
 - Pulse train
 - Saw-tooth or triangle wave
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**GUI's
(1 hour)**

- Filter Design and Analysis Tool
 - GUI-based filter design
 - Open interactive digital signal processing tool
 - Open Filter Visualization Tool
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**Bi-level Waveform
Measurements
(30 min)**

- Duty cycle of pulse waveform
 - Fall time of negative going bi-level waveform transitions
 - Period of bi-level pulse
 - Separation between bilevel waveform pulses
 - Bilevel waveform pulse width
 - Slew rate of bilevel waveform
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