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Digital Signal Processing Lab 1: Signal generation &

Lab 1: Signal generation & analysis in Matlab

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Digital Signal Processing: Lab Sessions

- Session 1: Signal generation & analysis in Matlab
- Session 2: Embedded DSP implementation of energy-based voice activity detector
- Session 3: Filter analysis & implementation in Matlab
- Session 4: Embedded DSP implementation of FIR filter
- Session 5: NLMS adaptive filtering in Matlab
- Session 6: Embedded DSP implementation of NLMS adaptive filter
- Session 7: Embedded DSP implementation of acoustic echo canceller





- In this session, we'll use these **built-in Matlab functions**:
 - basis arithmetic: +, -, *, /, sin, cos, abs, ...
 - random signal generation: rand, randn, ...
 - frequency analysis: fft, ifft
 - (FFT implementation of discrete Fourier transform)
 - time-frequency analysis: spectrogram
 (FFT implementation of short-time Fourier transform)
 - audio I/O: audioread, audiowrite, soundsc, ...
 - visualization: figure, plot, ...
- Remember: Matlab Help is your best friend
 - >> help
 - >> doc

• Exercise 1.1: Generation & analysis of sinusoids

- Generate a signal of length 10 s sampled at 16 kHz, containing a sum of sines of 50, 100, 200, 500, 1000, 2000, 4000 and 6000 Hz. Plot the signal as a function of time.
- Compute and plot the frequency magnitude spectrum. Think about what you observe.
- Compute and plot the spectrogram. How to interpret this?
 Play around with the spectrogram parameters (window size, FFT size, ...) and see how the figure changes.
- Play back the signal through your PC loudspeaker. Make sure clipping is avoided!

• Exercise 1.2: Generation & analysis of sinusoids

- Multiply the signal from Exercise 1.1 with a gain factor that linearly decreases from value 1 at time = 0 s to value 0 at time = 10 s.
- Plot again the time-domain signal, the frequency magnitude spectrum, and the spectrogram, and listen to the audio playback of the signal.
- What do these results tell you about the frequency spectrum of a non-stationary signal?

- Exercise 1.3: Generation & analysis of random noise
 - Generate a Gaussian white (pseudo-)random noise signal of length 10 s sampled at 16 kHz. Plot the signal as a function of time and listen to the audio playback of the signal.
 - Compute and plot the frequency magnitude spectrum as well as the spectrogram and interpret the results.
 - Square the frequency magnitude spectrum values and transform the result back to the time domain. Plot the resulting time-domain signal. What does this represent? Why does it look like this?





- Exercise 1.4: Recording & analysis of speech
 - Record about 10 s of your own voice using your favorite audio recording/editing software.
 - Save the recording in a WAV file using a sampling frequency of 16 kHz and 16-bit quantization.
 - Open the recorded WAV file in Matlab. Plot the signal as a function of time and listen to the audio playback of the signal.
 - Compute and plot the frequency magnitude spectrum as well as the spectrogram and interpret the results.
 - Repeat the last 3 steps using an 8 kHz sampling frequency.
 What do you observe?