Laboratory Test
DSP 2022-2023

Information

- The test is taken during the lab, and lasts for 1 hour

- General Matlab stuff you need to know is listed in the **Syllabus** section. This is not an exhaustive list. There may be things I forgot to put in the list, but the basics are there.

- Template subjects (i.e. exercises extracted from the labs) are in **Template Subjects** section

- The test will be roughly based on these templates, with modifications

Syllabus (not exhaustive)

- Define scalars, vectors, matrices

  - Generate constant vectors (zeros, ones, some other value)
  - Generate sin and cos signals of a certain length, with various amplitude, frequencies and initial phase
  - Generate random numbers, vectors and matrices, using `rand` and `randn`
  - Generate linearly spaces values between a start value and a stop value (e.g. `linspace()` or `start:step:stop`)

- Concatenate vectors and matrices. Build longer signals from several parts.

- Access elements from vectors/matrices. Select rows or columns from matrices.

- Do basic mathematical operations with scalars, vectors, matrices

- Use basic instructions (`if`, `for`, `while`)

- Make plots

- Plot a vector
- Plot a vector against another vector (e.g. put the vector `n` on the horizontal axis)
- Plot multiple signals on the same figure
- Use subplots

- Operate with audio files (`.wav`, `.mp3`)

  - load a file
  - load only a certain part of a file (e.g. the first 5 seconds)
  - extract a single channel
  - play with the data, e.g. swap channels

- Operate with images

  - Create a simple grayscale or color image and display it
  - Load and display an image file. Also do simple adjustments (convert to grayscale, divide to 255).
  - Apply a simple filter to an image (e.g. a $3 times 3$ matrix with coefficients)

- Create and display a video based on a simple animation of an image (e.g. scrolling the image, change luminosity)

- Create and use Matlab functions

- Implement a system in a Matlab function, based on the system equation

  - the function takes `x` as input and outputs the result vector `y`

- Do convolutions of vectors with `conv()`

- Compute the Discrete Fourier transform (DFT) of a vector, compute and display the modulus and the phase of the Fourier transform

- Implement a system in Simulink based on the equation

  - Also apply some input, visualize the output
  - Apply an impulse and visualize the impulse response

- Design a IIR or FIR low-pass/high-pass etc. filter with `fdatool` GUI, according to specifications.

  Export the coefficients to Matlab, or implement the filter in Simulink.

- Use the filter coefficients to filter a signal, or to filter an image by columns and rows.

- Design an oscillator with a prescribed frequency, and implement it in Simulink (Lab 13)

Template Subjects

Lab 2

1. Plot on the same figure the signals $sin(2\pi ft + \frac{\pi}{4})$ and $cos(2\pi ft + \frac{\pi}{8})$, with $f = 0.3$ and $t \in [0, 10]$.

2. Load the audio file 'Kalimba.mp3' in the Matlab workspace. Only load samples between 1 and 200000 (to avoid out of memory error)

   a. Play it through the computer's audio device
   b. Change the sampling frequency to half the correct value, and play again. How will the sound be changed?
   c. Amplify the sound by multiplying the data by 4. Play the sound and observe the difference.
   d. Swap the left and right channels (it's a stereo file) and play the sound again.

Lab 3 version 1

3. Create a color image representing the Romanian flag (3 stripes of blue, yellow, red). Create the image using the following steps: Create three matrices for the R, G, B components of the image Concatenate the three matrices across third dimension, into a 3D tensor

Variant: Make another simple flag or figure, color or grayscale, and show it

Lab 3 version 2

1. Load the `Lena` image (use `imread()`), convert it to double, convert it to grayscale, scale the values to the [0, 1] range, and display the image (use `imshow()`).

2. Construct a new image based on the `Lena`, but in which each pixel value is set as a linear combination of the original pixels around it, as in the following equation:

$$y[i, j] = \frac{1}{9}x[i - 1, j - 1] + \frac{1}{9}x[i - 1, j] + \frac{1}{9}x[i - 1, j + 1]$$
$$+ \frac{1}{9}x[i, j - 1] + \frac{1}{9}x[i, j] + \frac{1}{9}x[i, j + 1]$$
$$+ \frac{1}{9}x[i + 1, j - 1] + \frac{1}{9}x[i + 1, j] + \frac{1}{9}x[i + 1, j + 1]$$

Ignore the first and last row/column, if needed.

Display the resulting image in a new window

Lab 4

1. Load the `Lena` image (use `imread()`), convert it to a grayscale image, convert it to `double` type, adapt the values to the $[0, 1]$ range, and display it (use `imshow()`).

2. Create a video sequence by scrolling the Lena image circularly to the right, by 3 pixels at every frame. Display the video at 25fps.

Code template for creating a video sequence in Matlab:

```
height = ...; % desired height
width  = ...; % desired width
NoF    = ...; % desired number of frames
% an array of size height x width x 1 x NoF:
video  = zeros(height, width, 1, NoF);
for i = 1:NoF
    video(:,:,:,i) = ... the frame number i ... ;
end

% Play the sequence
implay(video);
```

Variant: do another thing instead of scrolling, like change luminosity etc.

Lab 6 & 7

1. Create a Simulink model to implement the following system H_1:

$$y[n] = H_1\{x[n]\} = \frac{1}{4}(x[n] + x[n-1] + x[n-2] + x[n-3])$$

- the system should be implemented as a Subsystem block with one input and one output signal

2. Visualize the impulse response of the system

- add a unit impulse as the input (hint: can be created from two unit ramp blocks, delayed)
- add a Scope at the output to visualize the data

Variant 3. Filter some audio signal, or some other input signal.

Lab 8

1. Load an audio signal and extract a 10 seconds long sequence of it.

   a. Convolve the sequence with the impulse response $\{1/6, 1/6, 1/6, 1/6, 1/6, 1/6\}$. Play the resulting sequence.
   b. Load another impulse response from the file "Scala Milan Opera Hall.wav" (use `audioread()`). Call the resulting vector **h**. Convolve the original audio signal with **h** and play the result.
   c. Convolve the result from b) with another impulse response from the pack. Play the resulting signal.
   d. Compute and display the equivalent impulse response of the complete system in points b) and c).

Lab 9

1. Generate a 100 samples long signal **x** defined as $x[n] = 0.3\cos(2\pi f_1 n) + 0.8\sin(2\pi f_2 n)$, with $f_1 = 0.05$ and $f_1 = 0.1$.

   a. Plot the signal in the top third of a figure (use `subplot()`).

   b. Compute the Fourier series coefficients with `fft()` and plot their magnitude in the middle third, and their phase in the lower third.

   c. Repeat the plot but do the FFT in N=1000 points (use `fft(x, N)`)

Lab 11

1. Use the Filter Design tool in Matlab (`fdatool`) to design a IIR high-pass filter with order 3, with cutoff frequency 0.07.

   Export its coefficients to Matlab Workspace.

2. Generate a 300-long periodic square signal, composed of 30 values of 1 followed by 30 values of 0, repeated for 5 times.

   Filter the signal with the filter obtained in Exercise 1, and display the input and the output signal.

Lab 12

1. Use the Filter Design tool in Matlab (`fdatool`) to design a FIR high-pass filter with order ..., with cutoff frequency 0.07, etc.

   Export its coefficients to Matlab Workspace.

2. Generate a 300-long periodic square signal, composed of 30 values of 1 followed by 30 values of 0, repeated for 5 times.

   Do a zero-phase filtering of the signal with the filter obtained in Exercise 1, and display the input and the output signal.

or:

1. Load the `Lena512.bmp` image and filter the image with this filter, as follows:

   - filter very row in the image, separately
   - then filter every column in the resulting matrix.

Lab 13

1. Use the Filter Design tool in Matlab (`fdatool`) to design an oscillator with frequency 0.05. Implement it in Simulink, visualize & play the output signal.

   Use the following steps to design the oscillator:

   1. design a system of order 2 with 2 conjugate poles placed **on the unit circle** at the correct frequency, and 2 zeros at low & high frequencies
   2. implement the system in Simulink, **omitting the input signal** (not necessary)
   3. set a non-zero initial condition in the system, to start-up the oscillator