- 1. Plot on the same figure the signals  $sin(2\pi ft)$  and  $cos(2\pi ft)$ , 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

- 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. How did it change?

3. Repeat Exercise 2 but change the values of the coefficients to

$$\begin{bmatrix} 0 & -1 & 0 \\ -1 & 5 & -1 \\ 0 & -1 & 0 \end{bmatrix}$$

- 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);
```

#### Lab 5

1. Create a function mysys1() that implements the following system  $H_1$ :

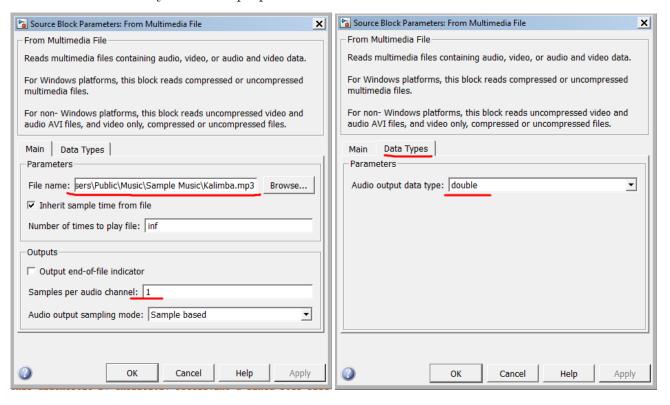
$$y[n] = H_1\{x[n]\} = \frac{1}{4}x[n] - \frac{1}{2}x[n-1] + \frac{1}{4}x[n-2]$$

- the function takes 1 input argument x and outputs 1 result vector y
- 2. In a separate script, test the linearity of this system in the following way:
  - generate two random vectors  $\mathbf{x}$  and  $\mathbf{y}$  and two random numbers  $\mathbf{a}$  and  $\mathbf{b}$
  - apply the function mysys1() to a\*x, b\*y, and a\*x + b\*y, and check if the results verify the linearity equation
  - the check shall be repeated for 5 times, with 5 different randomly generated data
  - is the system linear?

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. Apply the system to the audio data (mp3 file) loaded with FromMultimediaFile block and play the resulting output using Buffer and ToAudioDevice blocks.
  - make sure you set the properties of the FromMultimediaFile block as shown below:



#### Lab 7

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

$$y[n] = H_1\{x[n]\} = 0.8y[n-1] + 0.25x[n] + 0.1x[n-1]$$

- the system should be implemented as a Subsystem block with one input and one output signal
- 2. Test linearity of this system as follows:
  - create three copies of the system inside the model (copy/paste)
  - use two random input vectors **x** and **y** (use two Random blocks)
  - apply input signals x, y and x+y to the three copies of the system
  - add the outputs of the systems which have x and y as inputs, then subtract the output of the system which has x + y as input
  - show the resulting signal. Is the system linear?

- 1. Generate a 100 samples long signal x defined as  $x[n] = 0.7\cos(2\pi f_1 n) + 1.2\sin(2\pi f_2 n)$ , with  $f_1 = 0.05$  and  $f_1 = 0.1$ .
  - a. Plot the signal in the top half of a figure (use subplot()).
  - b. Compute the Fourier series coefficients with fft() and plot their magnitude in the lower half of the figure.
- 2. Take the Fourier series coefficients of the above signal x, and keep only the coefficients of the DC + first two sinusoidal components. Generate the signal from the Fourier coefficients with ifft() and plot it. What is the resulting signal?

# **Lab 12**

- 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.
  - 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
  - implement the system in Simulink. You can **omit the input signal** (not necessary, an oscillator has no input)
  - set a non-zero initial condition in the system, to start-up the oscillator
  - play the resulting output using Buffer and ToAudioDevice blocks. What frequency do you hear when running the simulation?