

Lecture “Digital Signal Processing”

Prof. Dr. D. Klakow, summer term 2019

Tutorial 4

Submission deadline: 27.05.2019, 10:15

Submission Instructions:

You have one week to solve the tutorials.

The code should be well structured and documented. Do not use any Matlab-Toolbox or Python external libraries if not mentioned that you could use it.

- You are allowed to hand in your practical solutions in groups of two students.
- The theoretical part should be submitted before the lecture.
- For the practical tasks please submit files via the email address
(Mon)**Tutorial 1: dsp.tutorial1@gmail.com.**
(Thus)**Tutorial 2: dsp.tutorial2@gmail.com.**
- The subject of the letter should be [DSP TUTORIAL 4].
- Rename and pack the main directory:
Ex04_matriculationnumber1_matriculationnumber2.zip.

The directory that you pack and submit should contain the following files:

- code files (contain *main.m* for matlab);
- file “answers.pdf” which contains answers to the questions appearing in the exercise sheet;
- file “README” that contains an information on all team members:
name
matriculation number
email address.

1 (6P)Exercise

In this exercise we will discuss (two types of) cross-correlation, and how it can be used to determine the time delay between two sensors.

1.1 (2P) Subtask

Given a discrete, finite signal $x[n]$, $n = 0, \dots, N-1$, the discrete Fourier transform is defined as

$$DFT\{x\}[k] = X[k] = \sum_{n=0}^{N-1} x[n]e^{-j2\pi nk/N}, \quad k = 0, \dots, N-1 \quad (1)$$

and the inverse discrete Fourier transform, i.e. the original signal is

$$DFT^{-1}\{x\}[n] = x[n] = \frac{1}{N} \sum_{k=0}^{N-1} X[k]e^{+j2\pi nk/N}, \quad n = 0, \dots, N-1 \quad (2)$$

The (circular) cross-correlation of two signals x, y is defined as

$$(x \star y)[n] = \sum_{m=0}^{N-1} x^*[m]y[(m+n) \bmod N], \quad n = 0, \dots, N-1 \quad (3)$$

where $*$ denotes the complex conjugate

What is the intuitive meaning of the cross-correlation as a function of n ?

What is the asymptotic running time of the “naive” algorithm for calculating the cross-correlation vector?

Prove that

$$(x \star y)[n] = DFT^{-1}\{DFT^*\{x\} \cdot DFT\{y\}\}[n] \quad (4)$$

How can we use this fact to calculate the cross-correlation more quickly? What is the asymptotic running time of the improved algorithm?

1.2 (2P) Subtask

There are two microphone sensors recording the same source of audio.

Write a program that reads in the given audio file *Sensor1.wav* and *Sensor2.wav*.

Use the fact shown in the subtask 1.1 to calculate the cross-correlation of the two signals.

Plot the cross-correlation against the time in seconds.

What does the maximum value of cross-correlation function mean? Corresponding value on X-axes.

What is the delay time between these two signals? which sensor did the audio arrive first?

Hint: you may use built-in functions for the transforms, but not for the cross-correlation (i.e. do not use `xcorr()`)

1.3 (2P) Subtask

Another type of cross-correlation is GCC-PATH cross-correlation

$$\hat{G}_{PATH}(f) = \frac{X_i(f)X_j^*(f)}{|X_i(f)X_j^*(f)|} \quad (5)$$

where $X_i(f), X_j(f)$ are Fourier transforms of the two signals $x_i(n)$ and $x_j(n)$. $*$ denotes complex conjugate.

Please read GCC-PATH Note and write a summary of it.

Plot GCC-PATH for two signals from subtask 1.2. What do you observe? Compare it with the result of 1.2.

Include two plots (for both types of cross-correlation) into your report.

2 (4P)Exercise

In this part we will discuss linear microphone arrays. Please see *microphonearray.pdf* for information.

2.1 (2P) Subtask

Giving sampling rate $f_s = 45kHz$ and the distance between the microphones $\Delta x = 0.25m$, $c = 343 m/s$.

Plot the angle of incidence φ (in degrees) for time delay $\Delta t = m \cdot T_s$ where $m = -24, \dots, 24$ and T_s is period of sampling.

Change Δx to $0.5m$, what do you observe? Include two plots into your report.

With the frequency of sound signal up to $900Hz$ and $\Delta x = 0.25m$, what is the maximum angle where the sound can be located?

2.2 (2P) Subtask

Here is a simplified 2-D speaker-located system. Supposed there is only one speaker in an ideal environment (ignore the effect of environment such as temperature, winds and etc.) and no sound reflection. Read the audio files *Sensor1.wav*, *Sensor2.wav* and *Sensor3.wav* recorded by three different sensors.

Please use the function that you implement in 1.3 to locate the speaker and draw it in the graph ($c = 343m/s$).

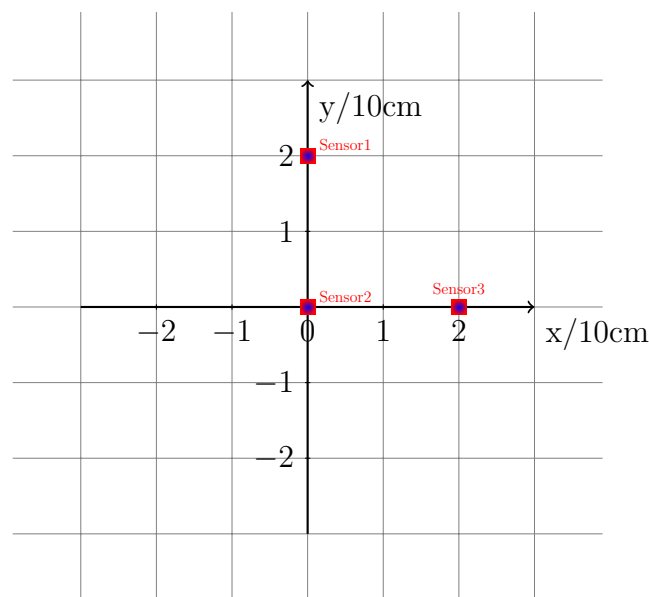


Figure 1

Is the speaker location in 3-D unique? Justify your answer.

2.3 (0P) Just thinking

The system will not be accurate since there is too much interference in the environment. Even a small error in φ could lead different location. Is there any method could improve the estimation of the location?