



# **Summary sheet**

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#### **Objective**

- To be familiar with impulse response and convolution



#### Part I

(6)

Include all useful codes of Part I in (e)

Let A = the 5<sup>th</sup> digit of your student ID

Let the input sequence x[n] be the same as your student ID.

$$x[n] = [2, 0, 3, 4, 5, 6, 7, 8]$$

n = A to A+7

Write **Matlab code** to define the input x and the index nx of the input.

The impulse responses of two LTI systems are shown below.

System 1 : 
$$h1[n] = \delta[n] + \delta[n-2]$$

System 2: 
$$h2[n] = u[n - A] - u[n - A - 4]$$

System 1 and System 2 are connected in series to form an overall system h[n].

The input sequence x[n] is applied to the overall system h[n] to give the output y[n].

a) Write down the mathematical expression of the overall impulse response h[n] in terms of h1[n] and h2[n].

## (1) h[n] = h1[n] \* h2[n]

Write **Matlab code** to define h1. Write **Matlab code** to define h2 using "ones". **DO NOT use "dirac" and "heaviside".** Write **Matlab code** to obtain the overall impulse response h according to the answer provided in (a). Write **Matlab code** to define the index nh of the overall impulse response.

b) Fill in the following table after running your Matlab code.

## (1) Only show non-zero value of h and corresponding nh

nh	5	6	7	8	9	10			
h	1	1	2	2	1	1			

Write **Matlab code** to obtain the output y and define the index ny of the output.

c) Fill in the following table after running your Matlab code.

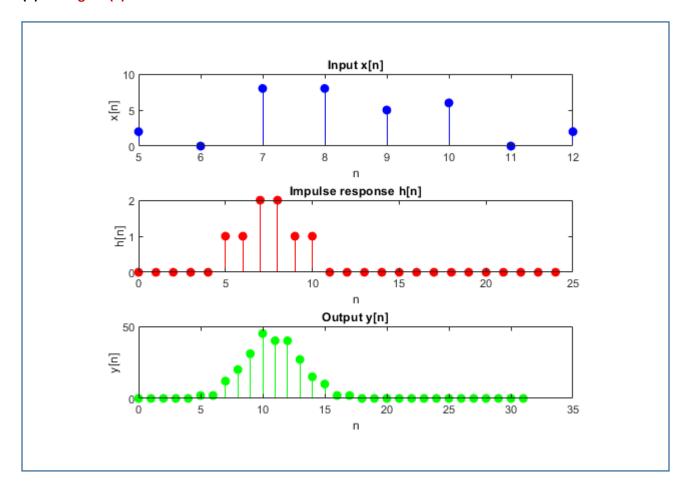
## (1) Only show non-zero value of y and corresponding ny

ny	5	6	7	8	9	10	11	12	13	14	15	16
У	2	2	12	20	31	45	40	40	27	15	10	2

ny	17						
У	2						

d) Write **Matlab code** to plot the input in **BLUE**, the overall impulse response in **RED** and the output in **GREEN** using "subplot" and "stem". Show the plot in figure (1).

## (1) figure (1)



## e) (2) Screenshot of Matlab code for Part I

Write Matlab code to read the audio file (song2a.wav).

a) What is the sampling frequency (fs) in Hz?

#### (1) The sampling frequency = 96000

The impulse responses of two CT LTI systems are given below.

System 1 : 
$$h1(t) = \delta(t)$$
 System 2 :  $h2(t) = 0.9 \delta(t - 0.3)$ 

These two CT LTI systems are converted into DT representation using the sampling frequency (fs). b) Write Matlab code to calculate the required number of zeros for representing a time delay of 0.3 seconds.

- (1) Number of zeros = 0.3 \* fs = 28800
- c) Write down the DT mathematical expressions of System 1 and System 2.

(1) 
$$h1[n] = \delta[n]$$
  $h2[n] = 0.9 \delta[n - 28800]$ 

These two DT LTI systems are connected in parallel and added together to form a new DT LTI system h[n]. d) Write down the DT mathematical expression of the impulse response h[n].

(1) 
$$h[n] = \delta[n] + 0.9 \delta[n - 28800]$$

Write Matlab code to define h1 and h2.

Write **Matlab code** to generate the impulse response h using h1 and h2.

Write Matlab code to generate the output y using "conv" if the audio file is applied to the new DT LTI system h. Write **Matlab code** to hear the original audio file and the output y using "sound".

- e) What does the new DT LTI system h do?
- (1) It delays the original signal, and the sound is quieter, less hearable.
- f) (2) Screenshot of Matlab code for Part II

```
lab2_1.m × lab2_2.m × +
          [x, fs] = audioread("song2a.wav");
 1
 2
          x = x';
 3
          delay = 0.3*fs;
 4
 5
          h = [1 zeros(1,delay) 0.9];
 6
 7
          y = conv(x, fs);
          sound(x, fs);
          sound(y, fs);
 9
10
```

# Part III (7) Include all useful codes of Part III in (d)

Write **Matlab code** to read the image file (**image2a.jpg**) and show the image file in figure (2). No need to place figure (2) on the summary sheet.

The image file is applied to a 2-D filter 
$$h1[n]=\begin{bmatrix} -1 & -1 & -1 \\ -1 & 8 & -1 \\ -1 & -1 & -1 \end{bmatrix}$$
.

Write Matlab code to define h1.

Write Matlab code to obtain the output y1 using "convn".

Write Matlab code to show y1 in figure (3) using "uint8" and "imshow".

## (1) figure (3)



a) What does the 2-D filter h1[n] do ?

(1) Detects edges.

The image file is applied to a 2-D filter 
$$h2[n] = \begin{bmatrix} 0 & 0 & 0 \\ 0 & 1 & 0 \\ 0 & 0 & 0 \end{bmatrix}$$
.

Write Matlab code to define h2.

Write Matlab code to obtain the output y2 using "convn".

Write Matlab code to show y2 in figure (4) using "uint8" and "imshow". No need to place figure (4) below.

b) What does the 2-D filter h2[n] do ?

## (1) Nothing.

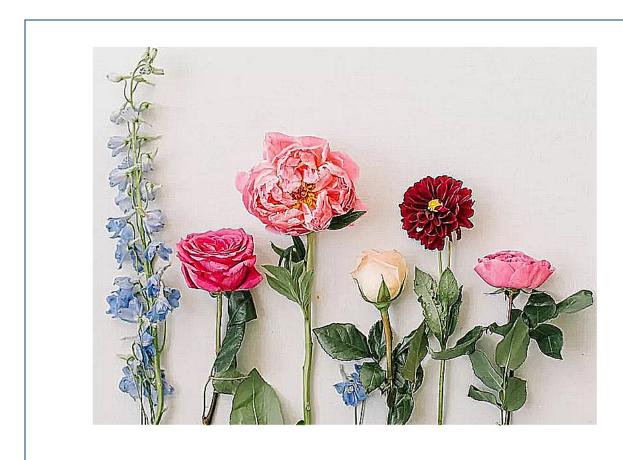
The 2-D filters h1[n] and h2[n] are connected in parallel and added together to form a new 2-D filter h3[n].

Write Matlab code to generate h3 using the defined h1 and h2.

Write **Matlab code** to obtain the output y3 using "convn" if the image file is applied to h3.

Write Matlab code to show y3 in figure (5) using "uint8" and "imshow".

## (1) figure (5)



- c) What does the new 2-D filter h3[n] do ?
- (1) Makes colors more expressive, sharper.

```
Editor - C:\Users\fkriuk\Desktop\lab2_3.m
  lab2_1.m × lab2_2.m × lab2_3.m × +
  1
           i1 = imread('image2a.jpg');
  2
           figure(2);
           imshow(i1);
  3
  4
  5
           h1 = [-1 -1 -1; -1 8 -1; -1 -1 -1];
  6
           y1 = convn(i1, h1, 'same');
  7
           figure(3);
  8
           imshow(uint8(y1));
  9
 10
           h2 = [0 \ 0 \ 0; \ 0 \ 1 \ 0; \ 0 \ 0 \ 0];
           y2 = convn(i1, h2, 'same');
 11
           figure(4);
 12
13
           imshow(uint8(y2));
 14
15
           h3 = h1 + h2;
           y3 = convn(i1, h3, 'same');
 16
           figure(5);
 17
           imshow(uint8(y3));
 18
 19
```