# **Signals Systems and Communication Lab**

Laboratory report submitted for the partial fulfillment of the requirements for the degree of

Bachelor of Technology in Electronics and Communication Engineering

by

Mohit Akhouri - 19ucc023

Course Coordinator
Dr. Navneet Upadhyay



Department of Electronics and Communication Engineering The LNM Institute of Information Technology, Jaipur

February 2021

Copyright © The LNMIIT 2021 All Rights Reserved

## **Contents**

Chapter				Page	
3	Expe	eriment -	.3	iv	
	3.1	Aim of	the experiment	iv	
	3.2	Softwa	re Used	iv	
	3.3	Theory	,	iv	
		3.3.1	About Discrete Fourier transform (DFT):	iv	
		3.3.2	About Inverse Discrete Fourier transform (IDFT):	iv	
		3.3.3	About Correlation:	V	
			3.3.3.1 Auto Correlation function (ACF):	V	
			3.3.3.2 Cross Correlation function (CCF):	V	
	3.4	Code a	nd Results	vi	
		3.4.1	Exercise $1(a)$ : Creating myDFT $(x,N)$ function to compute DFT of signal $x(t)$ :	vi	
		3.4.2	Exercise 1(b): Creating myIDFT (X,N) to calculate IDFT of signal $x_1[n] = [1 \ 1]$	1]: ix	
		3.4.3	Exercise 2: Plotting the magnitude and phase spectra of speech signal (dove.wav)	: xii	
		3.4.4	Exercise 3: MATLAB program to calculate the CCF of signals $x[n]$ and $y[n]$ : . $x$	- civ	
	3.5	Conclu	sion	vii	
	•				

## Chapter 3

## **Experiment - 3**

## 3.1 Aim of the experiment

- 1. Implementation of discrete Fourier transform (DFT) and inverse DFT (IDFT) algorithm
- 2. Implementation of autocorrelation and cross correlation algorithm

## 3.2 Software Used

**MATLAB** 

## 3.3 Theory

### 3.3.1 About Discrete Fourier transform (DFT):

In mathematics, the discrete Fourier transform (DFT) converts a finite sequence of equally-spaced samples of a function into a same-length sequence of equally-spaced samples of the discrete-time Fourier transform (DTFT), which is a complex-valued function of frequency. The interval at which the DTFT is sampled is the reciprocal of the duration of the input sequence. It can be said to convert the sampled function from its original domain to the frequency domain. The sequence of N complex numbers  $x_0, x_1, x_2, \ldots, x_{N-1}$  is transformed into an N-periodic sequence of complex numbers:

$$X[k] = \sum_{n=0}^{N-1} x[n]e^{-j2\pi kn/N}, k \in \mathbb{Z}$$
(3.1)

#### 3.3.2 About Inverse Discrete Fourier transform (IDFT):

An inverse DFT is a Fourier series, using the DTFT samples as coefficients of complex sinusoids at the corresponding DTFT frequencies. It has the same sample-values as the original input sequence. The inverse Fourier transform maps the signal back from the frequency domain into the time domain.

3.3. THEORY v

The perfect invertibility of the Fourier transform is an important property for building filters which remove noise or particular components of a signals spectrum. The IDFT is given by:

$$X[n] = \frac{1}{N} \sum_{k=0}^{N-1} x[k] e^{j2\pi kn/N}, n \in \mathbb{Z}$$
 (3.2)

#### 3.3.3 About Correlation:

Correlation is a measure of similarity between two signals. The relationship between two signals indicates whether one depends on the other, both depend on some common phenomenon, or they are independent. Correlation function indicates how correlated two signals are as a function of how much one of them is shifted in time. There are two types of correlation:

- 1. Auto correlation
- 2. Cross correlation

#### 3.3.3.1 Auto Correlation function (ACF):

It is defined as correlation of a signal with itself. Auto correlation function is a measure of similarity between a signal its time delayed version. Consider a random process x(t) (continuous time) ,Its autocorrelation is computed as :

$$R_{xx}(\tau) = \lim_{T \to \infty} \frac{1}{2T} \int_{-T}^{T} x(t)x(t+\tau)dt$$
 (3.3)

For sampled signal with N samples, ACF is defined as:

$$R_{xx}(m) = \frac{1}{N} \sum_{n=1}^{N-m+1} x[n]x[n+m-1], m = 1, 2, \dots, N+1$$
(3.4)

#### **3.3.3.2** Cross Correlation function (CCF):

Cross correlation is the measure of similarity between two different signals. For two wide sense stationary (WSS) processes x(t) and y(t), the CCF is defined as:

$$R_{xy}(\tau) = \lim_{T \to \infty} \frac{1}{2T} \int_{-T}^{T} x(t)y(t+\tau)dt$$
 (3.5)

For sampled signal with N samples, CCF is defined as:

$$R_{xy}(m) = \frac{1}{N} \sum_{n=1}^{N-m+1} x[n]y[n+m-1], m = 1, 2, \dots, N+1$$
(3.6)

#### 3.4 Code and Results

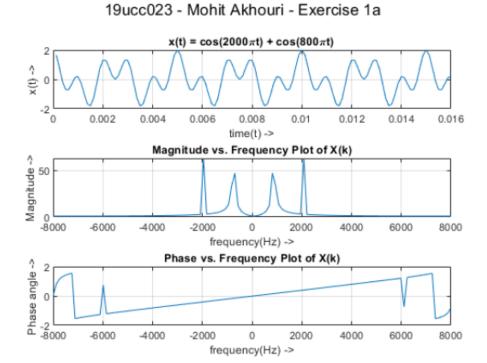
## 3.4.1 Exercise 1(a): Creating myDFT (x,N) function to compute DFT of signal x(t):

```
function [X] = myDFT(x,N)
% This function will calculate the DFT of function 'x' with total
samples N
% Name = Mohit Akhouri
% Roll no = 19UCC023
% SSC LAB Batch D1 - Monday ( 2-5 pm )
X=zeros(1,N); %initializing output X
% main loop algorithm for calculating DFT of 'x' starts here
for k=1:N
    sum=0;
    for n=1:N
        sum=sum+(x(n).*(exp(-1j*2*pi*(k-1)*(n-1)/N)));
    end
    X(k) = sum;
end
end
```

**Figure 3.1** Code for the function myDFT designed for finding DFT of signal x(t)

```
% Name = Mohit Akhouri
% Roll no = 19UCC023
% SSC LAB Batch D1 - Monday ( 2-5 pm )
% This is the code to perform DFT for
x(t) = cos(2000*pi*t) + cos(800*pi*t)
% by calling function myDFT(x,N)
% initializing constants fs,N,ts and to
fs = 8000;
N = 128;
ts = 1/fs;
to = N*ts;
t = ts:ts:to; % defining range from 'ts' to 'to'
f = linspace(-fs,fs,N); % defining range from '-fs' to 'fs'
x = cos(2000*pi*t) + cos(800*pi*t); % defining x(t)
subplot (3,1,1);
plot(t,x);
xlabel('time(t) ->');
ylabel('x(t) ->');
title('x(t) = cos(2000\pit) + cos(800\pit)');
grid on;
X = myDFT(x,N); % calculating DFT of 'x' using myDFT
mag = fftshift(abs(X));
ang = angle(X);
subplot (3,1,2);
plot(f, mag);
xlabel('frequency(Hz) ->');
ylabel('Magnitude ->');
title('Magnitude vs. Frequency Plot of X(k)');
grid on;
subplot (3,1,3);
plot(f, ang);
xlabel('frequency(Hz) ->');
ylabel('Phase angle ->');
title('Phase vs. Frequency Plot of X(k)');
sqtitle('19ucc023 - Mohit Akhouri - Exercise 1a');
```

**Figure 3.2** Code for calculating DFT of signal x(t) with N=128 and  $F_s = 8000$  Hz



**Figure 3.3** Graph of calculated DFT of signal x(t) with N=128 and  $F_s = 8000$  Hz

## 3.4.2 Exercise 1(b): Creating myIDFT (X,N) to calculate IDFT of signal $x_1[n] = [1\ 1\ 1\ 1]$ :

```
function [x] = myIDFT(X, N)
% This function will calculate the inverse discrete fourier transform
% (IDFT) of X with total samples N
% Name = Mohit Akhouri
% Roll no = 19UCC023
% SSC LAB Batch D1 - Monday ( 2-5 pm )
x=zeros(1,N); % initializing output
% main loop algorithm to calculate IDFT
for n=1:N
    sum=0;
    for k=1:N
        sum=sum+(X(k).*(exp(1j*2*pi*(k-1)*(n-1)/N)));
    sum=sum/N;
    x(n) = sum;
end
x=abs(x);
end
```

**Figure 3.4** Code for the function myIDFT designed for finding IDFT of signal  $x_1[n]$ 

```
% Name = Mohit Akhouri
% Roll no = 19UCC023
% SSC LAB Batch D1 - Monday ( 2-5 pm )
% This is the code to perform IDFT for x1[n]={1,1,1,1} with N=4
% by calling function myIDFT(X,N)
%defining constants N
N=4;
x1=[1 1 1 1]; %defining x1[n]
n=0:N-1; %defining range for 'n'
subplot (2,2,1);
stem(n,x1,'Linewidth',1);
xlabel('Total samples (N) ->');
ylabel('x_{1}(n) ->');
title('Plot of x_{1}(n) = [1 \ 1 \ 1]');
grid on;
X1=myDFT(x1,N); % calculating DFT of x1[n] using myDFT
mag=fftshift(abs(X1));
ang=angle(X1);
subplot (2,2,2);
stem(n, mag, 'Linewidth', 1);
xlabel('Total samples (N) ->');
ylabel('Magnitude ->');
title('Magnitude vs. Samples for X {1}(k)');
grid on;
subplot (2,2,3);
stem(n, ang, 'Linewidth', 1);
xlabel('Total samples (N) ->');
ylabel('Phase angle ->');
title('Phase vs. Samples for X {1}(k)');
grid on;
y1=myIDFT(X1,N); % calculating IDFT of X1[k]
subplot (2,2,4);
stem(n,abs(y1),'Linewidth',1);
xlabel('Total samples (N) ->');
ylabel('y {1}(n) ->');
title('y_{1}(n) obtained after IDFT of X_{1}(k)');
grid on;
sgtitle('19ucc023 - Mohit Akhouri - Exercise 1b');
```

**Figure 3.5** Code for calculating IDFT of signal  $x_1[n]$  with N=4

Plot of x<sub>1</sub>(n) = [1 1 1 1] Magnitude vs. Samples for  $X_1(k)$ Magnitude -> 0 0 2 0 1 2 3 3 Total samples (N) -> Total samples (N) -> y<sub>1</sub>(n) obtained after IDFT of X<sub>1</sub>(k) Phase vs. Samples for X<sub>1</sub>(k) -0.5 -1.5 -2.1.5 ^- (u)<sup>1</sup> x -2 0 3 0 3 Total samples (N) -> Total samples (N) ->

19ucc023 - Mohit Akhouri - Exercise 1b

**Figure 3.6** Graph of calculated IDFT of signal  $x_1[n]$  with N=4

#### 3.4.3 Exercise 2: Plotting the magnitude and phase spectra of speech signal (dove.wav):

```
% Name = Mohit Akhouri
% Roll no = 19UCC023
% SSC LAB Batch D1 - Monday ( 2-5 pm )
% This is the code to calculate and plot the magnitude and phase
% spectra of speech signal 'dove.wav'
[x,fs] = audioread('dove.wav'); % reading audio signal
ts=1/fs;
N=length(x);
to=N*ts;
f=linspace(-fs,fs,N); % defining rangde of 'f'
t=ts:ts:to; % defining range of 't'
subplot (3,1,1);
plot(t,x);
xlabel('time(t) ->');
ylabel('x(t) ->');
title('x(t) = dove.wav');
grid on;
X=myDFT(x,N); % calculating DFT of audio signal
mag=fftshift(abs(X));
ang=angle(X);
subplot(3,1,2);
plot(f, mag);
xlabel('frequency(Hz) ->');
ylabel('Magnitude ->');
title('Magnitude vs. Frequency Plot of X(k)');
grid on;
subplot (3,1,3);
plot(f,ang);
xlabel('frequency(Hz) ->');
ylabel('Phase angle ->');
title('Phase vs. Frequency Plot of X(k)');
sgtitle('19ucc023 - Mohit Akhouri - Exercise 2');
```

Figure 3.7 Code for calculating DFT of speech signal (dove.wav) using myDFT function

## 19ucc023 - Mohit Akhouri - Exercise 2

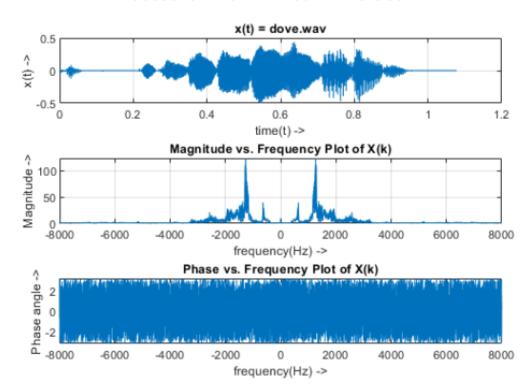


Figure 3.8 Graph of Magnitude and phase spectra of speech signal (dove.wav)

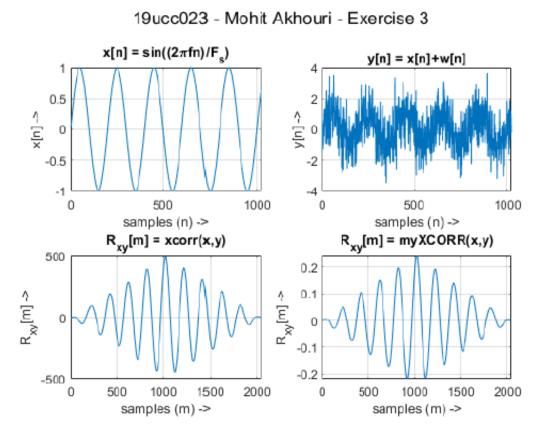
#### 3.4.4 Exercise 3: MATLAB program to calculate the CCF of signals x[n] and y[n]:

```
function [Rxy] = myXCORR(x, y)
% This function will calculate the cross correlation of two signals
% 'x' and 'y'
% Name = Mohit Akhouri
% Roll no = 19UCC023
% SSC LAB Batch D1 - Monday ( 2-5 pm )
% defining constants nx,ny and N
nx=length(x);
ny=length(y);
N=nx+ny-1;
Rxy=zeros(1,N); % initializing Rxy
flip(y); % flipping y
x=[x zeros(1,N-nx)]; % padding with right zeros
y=[zeros(1,N-ny) y]; % padding with left zeros
% main loop algorithm to calculate cross-correlation of 'x' and 'y'
for m=1:N+1
    sum=0;
    for n=1:N-m+1
        sum=sum+(x(n)*y(n+m-1));
    end
    sum=sum/N;
    Rxy(m) = sum;
end
end
```

**Figure 3.9** Code for function myXCORR(x,y) to perform the CCF of signals x[n] and y[n]

```
% Name = Mohit Akhouri
% Roll no = 19UCC023
% SSC LAB Batch D1 - Monday ( 2-5 pm )
% This code will calculate the cross correlation of two discrete
signals
variance
% of the Gaussian random process
% defining constants f,fs,N and n
f=1;
fs=200;
N=1024;
n=1:N;
x=sin((2*pi*f*n)/fs); % defining x[n]
subplot (2,2,1);
plot(n,x);
xlabel('samples (n) ->');
ylabel('x[n] ->');
title('x[n] = sin((2 \pi)/F_{s})');
grid on;
w=randn(1,N); % defining Gaussian random process
y=x+w; % defining y[n]
subplot (2,2,2);
plot(n,y);
xlabel('samples (n) ->');
ylabel('y[n] ->');
title('y[n] = x[n]+w[n]');
grid on;
Rxy inbuilt=xcorr(x,y); % calculating cross correlation of 'x' and 'y'
using inbuilt function
subplot (2,2,3);
plot(Rxy_inbuilt);
xlabel('samples (m) ->');
ylabel('R_{xy}[m] ->');
title('R_{xy}[m] = xcorr(x,y)');
grid on;
Rxy_myfunc=myXCORR(x,y); % calculating cross correlation of 'x' and
'y' using myXCORR function
subplot (2,2,4);
plot(Rxy_myfunc);
xlabel('samples (m) ->');
ylabel('R_{xy}[m] ->');
title('R_{xy}[m] = myXCORR(x,y)');
grid on;
sgtitle('19ucc023 - Mohit Akhouri - Exercise 3');
```

**Figure 3.10** Code for performing CCF of x[n] and y[n] using myXCORR(x,y)



**Figure 3.11** Graph for performed CCF of x[n] and y[n] using myXCORR(x,y)

3.5. CONCLUSION xvii

#### 3.5 Conclusion

In this experiment, we learnt about the transforms DFT for converting a signal from time domain to frequency domain and IDFT for the reverse process of converting signal from frequency domain to time domain. We also learnt about the Correlation algorithms (Cross Correlation and Auto Correlation functions) to calculate the similarity between two signals. We also created myDFT(x,N) and myIDFT(x,N) to perform DFT and IDFT respectively. We also learnt how to process an audio signal in MATLAB using audioread() function and how to plot its magnitude and phase spectra. Finally we created myXCORR(x,y) function to calculate cross-correlation of two signals x[n] and y[n].