# FOURIER SERIES REPRESENTATION OF CONTINUOUS TIME SIGNALS LAB # 10



#### **CSE301L Signals & Systems Lab**

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Class Section: **B** 

"On my honor, as a student of University of Engineering and Technology, I have neither given nor received unauthorized assistance on this academic work."

Stud	ent S	gnature:
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Submitted to: Engr. Durr-e-Nayab

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## Lab Objectives:

Objectives of this lab are as follows:

- Fourier Series Representation of Continuous Time Period Signals
- Convergence of CT Fourier Series
- Properties of CT Fourier Series
  - o Linearity
  - Time Shifting
  - o Frequency Shifting
  - o Time Reversal
  - Time Scaling

#### **Task # 1:**

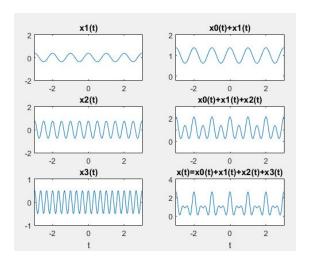
In the above example, ak's are chosen to be symmetric about the index k=0, i.e. ak=a-k. Select new ak's on your own to alter this symmetry and form the new signal. What do you observe? Is x(t) a real signal when coefficients are not symmetric?

#### **Observation:**

x(t) is a real signal when coefficients are not symmetric.

```
t = -3:0.01:3; % duration of signal
% dc component for k=0
x0 = 1;
% first harmonic components for k=-1 and k=1
x1 = (1/4)*exp(j*(-1)*2*pi*t)+(1/8)*exp(j*(1)*2*pi*t);
y1 = x0 + x1; % sum of dc component and first harmonic
% second harmonic components for k=-2 and k=2
x2 = (1/2)*exp(j*(-2)*2*pi*t)+(1/4)*exp(j*(2)*2*pi*t);
y2 = y1 + x2; % sum of all components until second harmonic
% third harmonic components for k=-3 and k=3
x3 = (1/3)*exp(j*(-3)*2*pi*t)+(1/6)*exp(j*(3)*2*pi*t);
x = x0 + x1 + x2 + x3; % sum of all components until third harmonic
figure;
subplot(3,2,1);
plot(t,x1);
axis([-3 3 -2 2]);
title('x1(t)');
subplot(3,2,2);
plot(t,y1);
axis([-3 3 -0.2 2]);
title('x0(t)+x1(t)');
subplot(3,2,3);
plot(t,x2);
axis([-3 3 -2 2]);
title('x2(t)');
```

```
subplot(3,2,4);
plot(t,y2);
axis([-3 3 -1 3]);
title('xO(t)+x1(t)+x2(t)');
subplot(3,2,5);
plot(t,x3);
xlabel('t');
axis([-3 \ 3 \ -1 \ 1]);
title('x3(t)');
subplot(3,2,6);
plot(t,x);
xlabel('t');
axis([-3 \ 3 \ -1 \ 4]);
title('x(t)=x0(t)+x1(t)+x2(t)+x3(t)');
```



#### **Task # 2:**

A discrete-time periodic signal x[n] is real valued and has a fundamental period of N = 5. The non-zero Fourier series coefficients for x[n] are

$$a_0 = 1, a_2 = a_{-2} = e^{j \text{ pi/4}}, a_4 = a_{-4} = 2e^{j \text{ pi/3}}$$

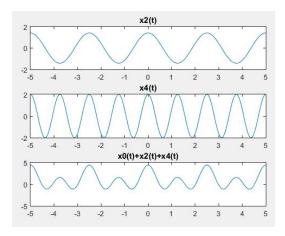
 $a_0=1 \ , a_2=a_{.2}=e^{j\ pi/4}, \ a_4=a_{.4}=2e^{j\ pi/3}$  Express x[n] as a linear combination of given coefficients.

# **Problem Analysis:**

Change the coefficients of  $a_0$  to 1 and  $a_2$ ,  $a_{-2}$  equal to  $e^{j \, pi/4}$  and  $a_4$ ,  $a_{-4}$  equal to  $2 e^{j \, pi/3}$ 

```
t = -5:0.01:5; % duration of signal
% dc component for k=0
x0 = 1;
% second harmonic components for k=?2 and k=2
x2 = (\exp(j*(pi/4)))*\exp(j*(-2)*(2*pi/5)*t) + (\exp(j*(pi/4)))*\exp(j*(2)*(2*pi/5)*t);
y2 = x0 + x2; % sum of all components until second harmonic
% fourth harmonic components for k=?4 and k=4
```

```
 \begin{array}{l} x4 = (2^*exp(j^*(pi/3)))^*exp(j^*(-4)^*(2^*pi/5)^*t) + (2^*exp(j^*(pi/3)))^*exp(j^*(4)^*(2^*pi/5)^*t); \\ x = x0 + x2 + x4; \% \ sum \ of \ all \ components \ until \ fourth \ harmonic \ figure; \\ subplot(3,1,1); \\ plot(t,x2); \\ title('x2(t)'); \\ subplot(3,1,2); \\ plot(t,x4); \\ title('x4(t)'); \\ subplot(3,1,3); \\ plot(t,x); \\ title('x0(t)+x2(t)+x4(t)'); \end{array}
```



#### **Task # 3:**

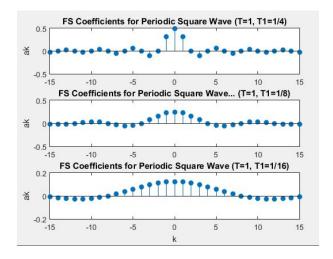
Considering the FS coefficients plot given below, what do you observe happens to the envelope of the coefficients when T1 is reduced from 1/4 to 1/16 with constant time period T?

#### **Observation:**

As T1 is changed from 1/4 to 1/16 and we observe that the frequency of the wave decreases and time period increases.

```
\begin{array}{l} k=-15:15; \ \% number \ of \ square \ wave \ coefficients \\ T=1; \ \% time \ period \ of \ square \ wave \\ T1=1/4; \ \% duty \ cycle \ of \ square \ wave \\ ak1=\sin(k^*2^*pi^*(T1/T))./(k^*pi); \ \% square \ wave \ Fourier \ series \ coefficients \\ \% \ Ignore \ the \ "divide \ by \ zero" \ warning \ that \ happens \\ \% \ because \ k \ in \ the \ denominator \ hits \ 0. \ We \ will \ now \ do \\ \% \ a \ manual \ correction \ for \ a0 \ ?> \ ak1(16) \\ ak1(16)=2^*T1/T; \\ subplot(3,1,1); \\ stem(k,ak1,'filled'); \\ ylabel('ak'); \end{array}
```

```
title('FS Coefficients for Periodic Square Wave (T=1, T1=1/4)');
T1 = 1/8;
ak2 = sin(k*2*pi*(T1/T))./(k*pi);
ak2(16) = 2*T1/T; % Manual correction for a0 ?> ak2(16)
subplot(3,1,2);
stem(k,ak2,'filled');
ylabel('ak');
title('FS Coefficients for Periodic Square Wave... (T=1, T1=1/8)');
T1 = 1/16;
ak3 = sin(k*2*pi*(T1/T))./(k*pi);
ak3(16) = 2*T1/T; % Manual correction for a0 ?> ak3(16)
subplot(3,1,3);
stem(k,ak3,'filled');
xlabel('k');
ylabel('ak');
title('FS Coefficients for Periodic Square Wave (T=1, T1=1/16)');
```



#### **Task # 4:**

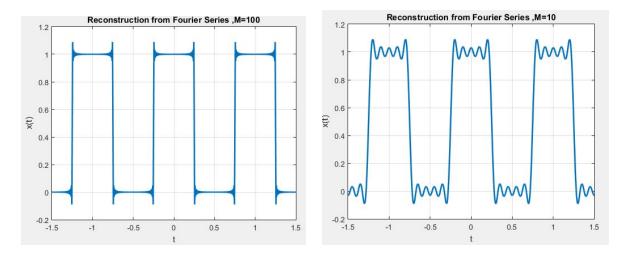
Considering the plots of square wave reconstructed using M = 10, 20, & 100 terms above, what do you observe about Gibb's phenomenon?

#### **Observation:**

As the number of signals to be added increases from 10 to 100 the square wave smoothens in form and is more defined.

```
t = -1.5:0.005:1.5; %square wave duration
T = 1; %time period of square wave
T1 = 1/4; %duty cycle of square wave
```

```
 w0 = 2*pi/T; % fundamental radian frequency of square wave M = 100; % number of coefficients k = -M:M; % 2M+1 total coefficients to construct square wave ak = <math>sin(k*2*pi*(T1/T))./(k*pi); ak(M+1) = 2*T1/T; % Manual correction for a0 ?> ak(M+1) x = zeros(1,length(t)); for k = -M:M x = x + ak(k+M+1)*exp(j*k*w0*t); end plot(t,x,'lineWidth',2); grid; xlabel('t'); ylabel('x(t')); title('Reconstruction from Fourier Series');
```



## **Task # 5:**

Given the following FS coefficients:

$$a_k = \begin{cases} 1, & k \text{ even} \\ 2, & k \text{ odd} \end{cases}$$

Plot the coefficients & reconstructed signal. Take the terms for reconstructed signal to be M = 10, 20, & 50. What effect do you see when M is varied?

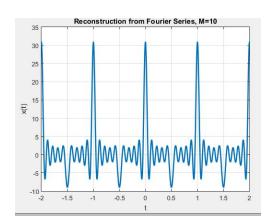
## **Observation:**

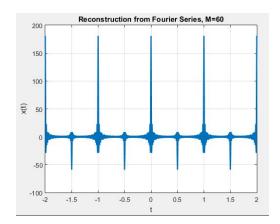
As the number of signals to be added increases from 10 to 60 the wave smoothens in form.

#### Code:

```
t = -2:0.005:2; %square wave duration
T = 1; %time period of square wave
T1 = 1/4; %duty cycle of square wave
w0 = 2*pi/T; %fundamental radian frequency of square wave
M = 60; %number of coefficients
x = zeros(1, length(t));
for k = -M:M; %2M+1 total coefficients to construct square wave
if(mod(k,2)==0)
ak = 1;
x = x + ak*exp(j*k*w0*t);
else
ak = 2;
x = x + ak*exp(j*k*w0*t);
\quad \text{end} \quad
end
plot(t,x,'lineWidth',2);
grid;
xlabel('t');
ylabel('x(t)');
title('Reconstruction from Fourier Series, M=60');
```

# **Output:**





## **Task # 6:**

Given the following FS coefficients:

$$a_{k} = \begin{cases} jk, & |k| < 3\\ 0, & otherwise \end{cases}$$

Plot the coefficients & reconstructed signal. Take 10 terms (M=10) for reconstructed signal.

## **Problem Analysis:**

Take M equal to 10 and plot the signal.

## **Code:**

```
t = -5:0.005:5; %square wave duration
T = 1; %time period of square wave
T1 = 1/4; %duty cycle of square wave
w0 = 2*pi/T; %fundamental radian frequency of square wave
M = 10; %number of coefficients
x = zeros(1, length(t));
for k = -M:M; %2M+1 total coefficients to construct square wave
if(abs(k) < 3)
ak = j*k;
x = x + ak*exp(j*k*w0*t);
else
ak = 0;
x = x + ak*exp(j*k*w0*t);
end
end
plot(t,x,'lineWidth',2);
grid;
xlabel('t');
ylabel('x(t)');
title('Reconstruction from Fourier Series, M=10');
```

# **Output:**

