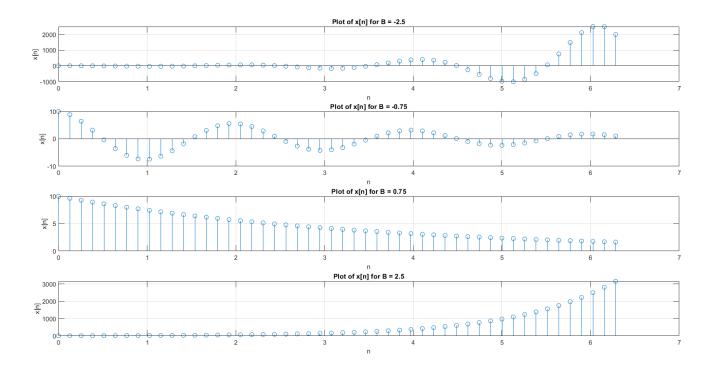
## EE387: Signal Processing Lab 2: Laboratory on Discrete Time Signals

Ranage R.D.P.R. - E/19/310

1). Understanding properties of Discrete Time Sinusoidal signals

```
a. Plot the discrete time real sinusoidal signal x[n] = 10\beta^n for positive C when,
     i. \beta < -1
     ii. -1 < \beta < 0
     iii. 0 < \beta < 1
     iv. \beta > 1
% Define values for n
n = linspace(0, 2*pi, 50)';
% Define different values of B
Bs = [-2.5, -0.75, 0.75, 2.5];
% Create a new figure for subplots
figure;
% Loop through each value of B and plot in a subplot
for i = 1:length(Bs)
    % Calculate result for current B
    result = xnB(n, Bs(i));
    % Create subplot for current result
    subplot(length(Bs), 1, i);
    stem(n, result);
    title(['Plot of x[n] for B = ', num2str(Bs(i))]);
    xlabel('n');
    ylabel('x[n]');
    grid on;
end
function result = xnB(n, B)
    % generate the result as given in the equation
    result = 10 * (B .^n);
    % Ensure that the result is real
    result = real(result);
end
```



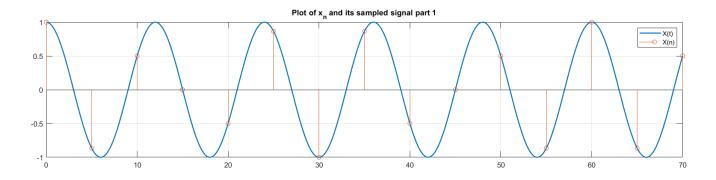
b. Plot x[n] and x(t) in the same plot for the following sinusoidal signals. Let n = kT where T = 5s and  $k \in \mathbb{Z}$ . That is x[n] is obtained by sampling x[t] at every 5 seconds. Determine the theoretical fundamental period of each signal.

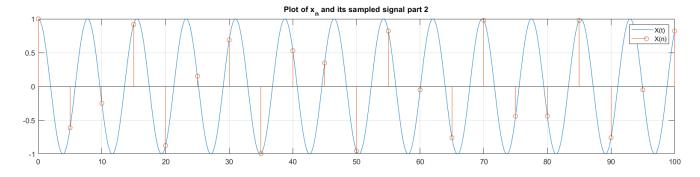
i). 
$$x[n] = cos\left(\frac{2\pi n}{12}\right)$$
,  $x[t] = cos\left(\frac{2\pi t}{12}\right)$ 

ii). 
$$x[n] = cos\left(\frac{8\pi n}{31}\right), x[t] = cos\left(\frac{8\pi t}{31}\right)$$

Is the observed period of the signal from the plot always equal to the theoretical period?

```
% sampling frequency
Ts = 5;
% t to plot continuous time signals
t1 = 0:0.01:70;
% n = KTs
n1 = 0:Ts:70;
xt1 = cos(t1*2*pi/12);
xn1 = cos(n1*2*pi/12);
% t to plot continuous time signals
t2 = 0:0.01:100;
% n = KTs
n2 = 0:Ts:100;
xt2 = cos(t2*8*pi/31);
xn2 = cos(n2*8*pi/31);
figure;
subplot(2,1,1);
plot(t1,xt1,'linewidth',1.5);
hold on;
stem(n1,xn1);
grid on;
title('Plot of x n and its sampled signal part 1')
legend('X(t)', 'X(n)');
subplot(2,1,2);
plot(t2,xt2);
hold on;
stem(n2,xn2);
grid on;
title('Plot of x n and its sampled signal part 2')
legend('X(t)', 'X(n)');
```





i). Theoretical fundamental period of x(t)

$$w_0 t = 2 * pi * t / 12$$

$$T = 2 * pi / w_0$$

$$T = 12s$$

Observed fundamental period of x[n] = 60

ii). Theoretical fundamental period of x(t)

$$w_0 t = 8 * pi * t / 31$$

$$T = 2 * pi / w_0$$

$$T = 31/4s$$

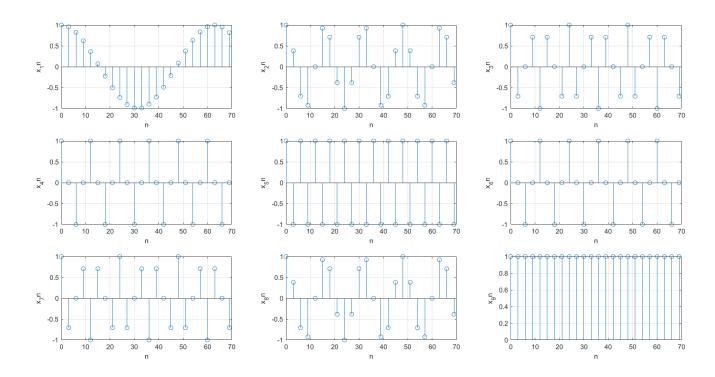
$$T = 7.5s$$

Observed fundamental period of x[n] is not detectable, (> 100)

### c. Plot the following nine discrete time signals in the same graph (use subplot command)

```
% sampling frequency
Ts = 3;
% n = KTs
n = (0:Ts:70)';
x1n = cos(n*0.1);
subplot(3,3,1);
stem(n, x1n);
grid on;
xlabel('n');
ylabel('x 1n');
x2n = cos(n*pi/8);
subplot(3,3,2);
stem(n, x2n);
grid on;
xlabel('n');
ylabel('x 2n');
x3n = cos(n*pi/4);
subplot(3,3,3);
stem(n, x3n);
grid on;
xlabel('n');
ylabel('x 3n');
x4n = cos(n*pi/2);
subplot(3,3,4);
stem(n, x4n);
grid on;
xlabel('n');
ylabel('x 4n');
x5n = cos(n*pi);
subplot(3,3,5);
stem(n, x5n);
grid on;
xlabel('n');
ylabel('x 5n');
x6n = \cos(n*pi*3/2);
subplot(3,3,6);
stem(n, x6n);
grid on;
```

```
xlabel('n');
ylabel('x 6n');
x7n = \cos(n*pi*7/4);
subplot(3,3,7);
stem(n, x7n);
grid on;
xlabel('n');
ylabel('x 7n');
x8n = cos(n*pi*15/8);
subplot(3,3,8);
stem(n, x8n);
grid on;
xlabel('n');
ylabel('x_8n');
x9n = cos(n*pi*2);
subplot(3,3,9);
stem(n, x9n);
grid on;
xlabel('n');
ylabel('x_9n');
```



## d. By observing the plots you have obtained in question 1.c, what can you tell about the shape of the signal as discrete frequency is varied?

When the sampling frequency is decreased, fewer data points are captured per unit of time, resulting in information loss. This leads to the original form of the signal being altered, as some data is no longer represented. Consequently, the signal's fidelity is compromised, and distortions may arise, especially at lower frequencies, due to inadequate data representation.

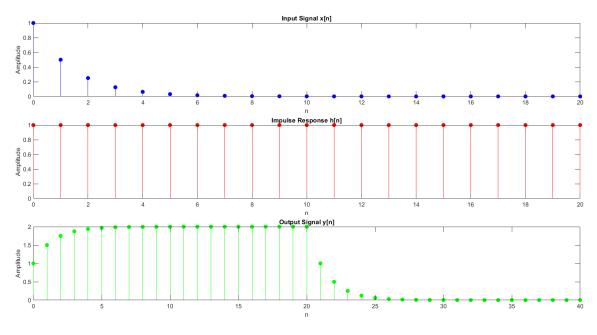
### 2). Discrete convolution

a. Write a matlab function to implement discrete convolution for n > 0. Note that y[n] = x[n] \* h[n] is given by the convolution summation

```
function y = conv disc(xn, hn)
    % Determine the lengths of xn and hn
    sizeX = length(xn);
    sizeH = length(hn);
     % Pad xn and hn with zeros to make their lengths equal to
    the sum of their original lengths minus one
    xn = [xn, zeros(1, sizeH - 1)];
    hn = [hn, zeros(1, sizeX - 1)];
    % Determine the size of the output
    out size = sizeH + sizeX - 1;
    % Pre-allocate memory for the output array for efficiency
    y = zeros(1, out size);
    % Perform convolution using nested loops
    for n = 1:out size
        kmin = max(1, n - sizeH + 1);
        kmax = min(n, sizeX);
        for k = kmin:kmax
            y(n) = y(n) + xn(k) * hn(n - k + 1);
        end
    end
end
```

# b. Using the function written in section a, convolve $x[n] = 0.5^n u(n)$ with h[n] = u[n]. Plot the output signal along with the two input signals.

```
% Define the input signals x[n] and h[n]
n = 0:20; % Define the range of n
x = 0.5.^n.^* (n \ge 0); % x[n] = 0.5^n * u[n]
h = (n >= 0); % h[n] = u[n]
y = conv disc(x, h);
subplot(3,1,1);
stem(n, x, 'b', 'filled'); % Plot x[n]
xlabel('n');
ylabel('Amplitude');
title('Input Signal x[n]');
subplot(3,1,2);
stem(n, h, 'r', 'filled'); % Plot h[n]
xlabel('n');
ylabel('Amplitude');
title('Impulse Response h[n]');
ny = 0:length(y)-1;
subplot(3,1,3);
stem(ny, y, 'g', 'filled'); % Plot the output signal
xlabel('n');
ylabel('Amplitude');
title('Output Signal y[n]');
```



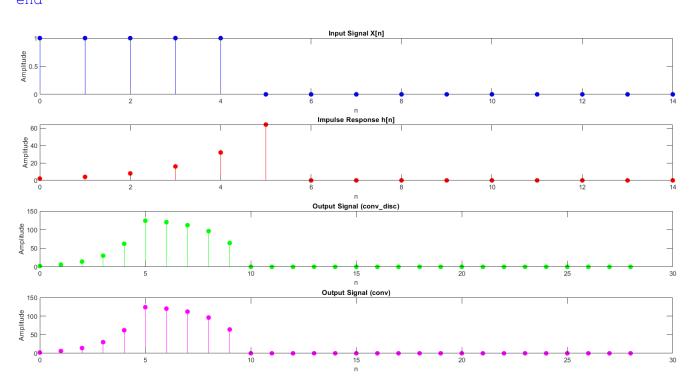
- c. Consider the following two signals
- i. X[n] = [1111110000000000]
- ii. h[n]= [2 4 8 16 32 64 0 0 0 0 0 0 0 0 0]
- iii. Convolve the two signals using the function written in part a. Use matlab conv command to verify your answer.

```
% Define the input signals X[n] and h[n]
h = [2 \ 4 \ 8 \ 16 \ 32 \ 64 \ 0 \ 0 \ 0 \ 0 \ 0 \ 0 \ 0];
y conv disc = conv disc(X, h);
y conv matlab = conv(X, h);
n x = 0:length(X)-1;
n h = 0:length(h)-1;
n y = 0:length(y conv disc)-1;
subplot(4,1,1);
stem(n x, X, 'b', 'filled'); % Plot X[n]
xlabel('n');
ylabel('Amplitude');
title('Input Signal X[n]');
subplot(4,1,2);
stem(n h, h, 'r', 'filled'); % Plot h[n]
xlabel('n');
ylabel('Amplitude');
title('Impulse Response h[n]');
subplot(4,1,3);
stem(n y, y conv disc, 'g', 'filled'); % Plot output signal from
conv disc
xlabel('n');
ylabel('Amplitude');
title('Output Signal (conv\ disc)');
subplot(4,1,4);
stem(n y, y conv matlab, 'm', 'filled'); % Plot output signal from
MATLAB's conv
xlabel('n');
ylabel('Amplitude');
title('Output Signal (conv)');
function y = conv disc(xn, hn)
    % Determine the lengths of xn and hn
    sizeX = length(xn);
    sizeH = length(hn);
    xn = [xn, zeros(1, sizeH - 1)];
    hn = [hn, zeros(1, sizeX - 1)];
```

```
% Determine the size of the output
out_size = sizeH + sizeX - 1;

% Pre-allocate memory for the output array for efficiency
y = zeros(1, out_size);

% Perform convolution using nested loops
for n = 1:out_size
    kmin = max(1, n - sizeH + 1);
    kmax = min(n, sizeX);
    for k = kmin:kmax
        y(n) = y(n) + xn(k) * hn(n - k + 1);
    end
end
end
```



iv. Considering the shape of the signal h[n] and the output signal, what sort of a transformation has been applied through the convolution operation?

There has been a linear transformation happened for h[n] [Y n] = 2 \* h[n] – 2 for x[n] = 1;

### 3). LTI Systems

- a. Consider the following processes. Identify intput x[n] and the output y[n] for each case. Implement a matlab function to implement the given system.
  - i. An investor is maintaining a bank account. The bank pays him a monthly interest of 1%. It is

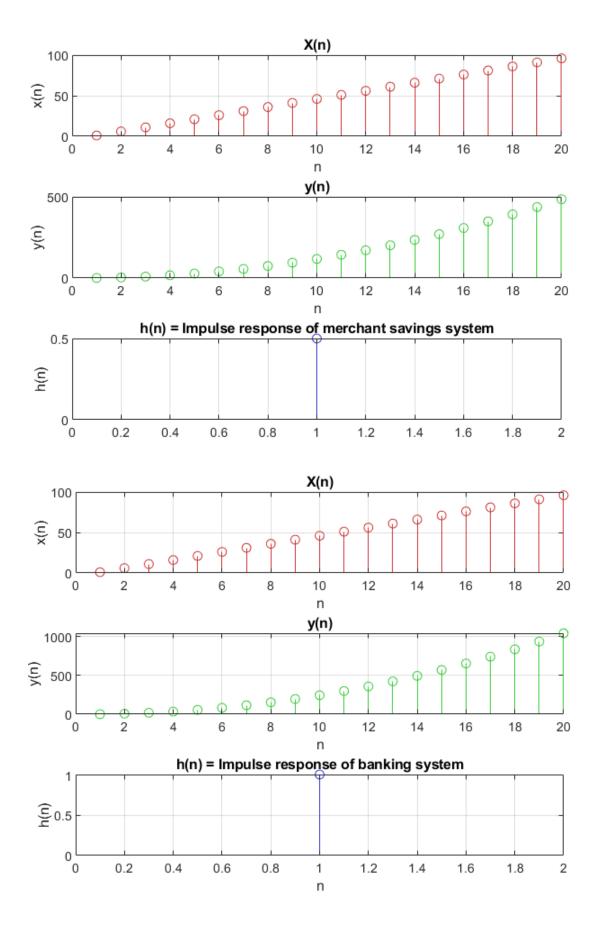
given that the net savings he makes is P. Write a function to calculate his current bank balance B in terms of B and P.

ii. A merchant earns M amount of money monthly. He spends half of it and retains the rest of it as savings. Write a function to calculate the amount of money he has as savings.

```
function y = bank balance(initial balance)
  interest rate = 0.01; % interest rate
  temp balance = initial balance(1); % store initial balance in
a temporary variable
 y(1) = temp balance * (1 + interest rate); % first month's
balance
  for i = 2:length(initial balance)
   temp balance = y(i - 1) + initial balance(i); % update
temporary balance
    y(i) = temp balance * (1 + interest rate); % generate
balance for each month
 end
end
% M will be an array something like [100,10,0,0]
function y = merchant savings(M)
  y(1) = M(1)/2; % first month's savings
  current savings = y(1); % store the current savings
  % calculate next savings based on savings given and add the
previous savings
  for i = 2:length(M)
    current savings = current savings + (M(i)/2); % update
current savings
    y(i) = current savings;
 end
end
```

## b. Find the impulse response of the above two LTI systems. Hint: you may use convolution function to obtain the impulse response

```
xn = 1:5:100;
yn = bank balance(xn); % get yn
% deconv to get hn - impulse response
hn = deconv(yn,xn);
display("Impulse response of banking system");
display(hn);
figure; % New figure for banking system
subplot(3,1,1);
stem(xn, 'Color', [0.8 0.2 0.2]); % Red color for X(n)
grid on;
title('X(n)'); xlabel('n'); ylabel('X(n)');
subplot(3,1,2);
stem(yn, 'Color', [0.2 0.8 0.2]); % Green color for y(n)
title('y(n)'); xlabel('n'); ylabel('y(n)');
grid on;
subplot(3,1,3);
stem(hn, 'Color', [0.2 0.2 0.8]); % Blue color for h(n)
title('h(n) = Impulse response of banking
system');xlabel('n');ylabel('h(n)');
grid on;
xn = 1:5:100;
yn = merchant savings(xn); % get yn
hn = deconv(yn, xn);
display("Impulse response of merchant savings system");
display(hn);
figure;
subplot(3,1,1);
stem(xn, 'Color', [0.8 0.2 0.2]); % Red color for X(n)
title('X(n)'); xlabel('n'); ylabel('X(n)');
subplot(3,1,2);
stem(yn, 'Color', [0.2 0.8 0.2]); % Green color for y(n)
title('y(n)'); xlabel('n'); ylabel('y(n)');
grid on;
subplot(3,1,3);
stem(hn, 'Color', [0.2 0.2 0.8]); % Blue color for h(n)
title('h(n) = Impulse response of merchant savings
system');xlabel('n');ylabel('h(n)');
grid on;
```



### c. Based on the results obtained at part b, classify the two LTI systems into IIR or FIR.

The system's two outputs are reliant on its earlier outputs. As a result, the output will not eventually zero out. Consequently, both systems fall within the category of infinite impulse response systems.