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# Digital Signal Processing

## MATLAB HW - q1

### Table of Contents

Clear recent data .....	1
Convolution in time doamain (part c) .....	1
Convolution in frequency doamain (part c) .....	2
Convolution in time doamain (part d) .....	3
Convolution in frequency doamain (part d) .....	4
Part e .....	5
M point DTFT (part a) .....	6
Convolution with time index (part b) .....	6

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## Clear recent data

```
clear; close all; clc;
```

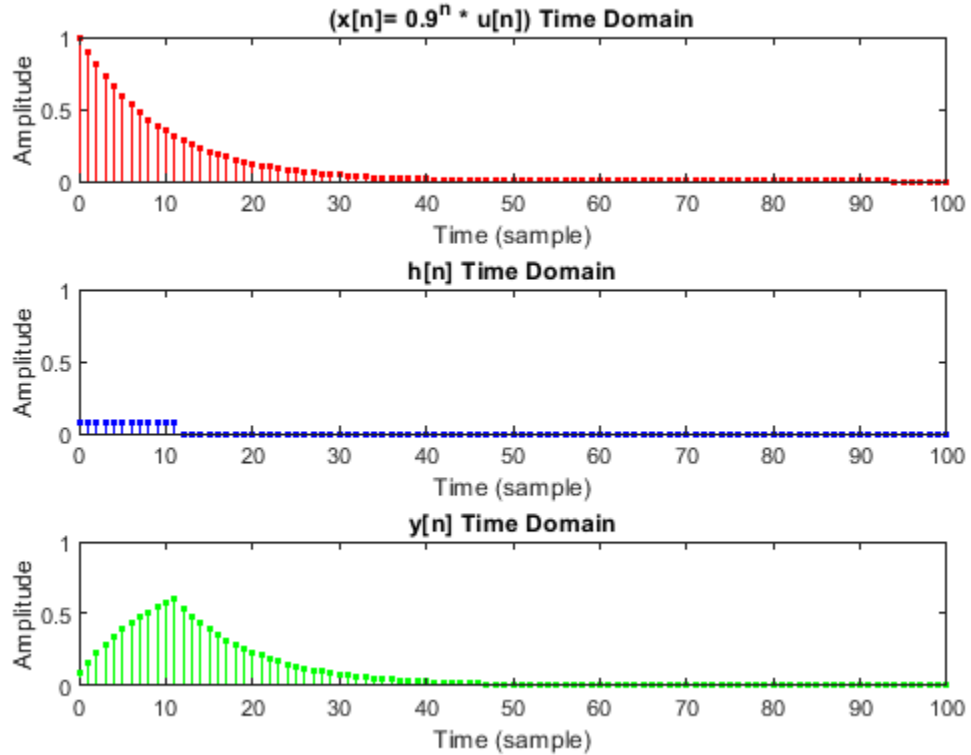
## Convolution in time doamain (part c)

```
% define parameters
NoS1 = 100;
n1 = 0 : NoS1;
h = zeros(length(n1), 1);
h(1 : 12) = 1 / 12;
x1 = (0.9 .^ n1);

% calculate convolution
[y1, s1] = convolution(h, n1, x1, n1);

% plot results in time domain
figure();
subplot(3, 1, 1);
stem(n1, x1, 'r.');
axis([0 NoS1 0 1]);
title('(x[n]= 0.9^n * u[n]) Time Domain');
xlabel('Time (sample)');
ylabel('Amplitude');
subplot(3, 1, 2);
stem(n1, h, 'b.');
axis([0 NoS1 0 1]);
title('h[n] Time Domain');
xlabel('Time (sample)');
ylabel('Amplitude');
subplot(3, 1, 3);
stem(s1, y1, 'g.');
```

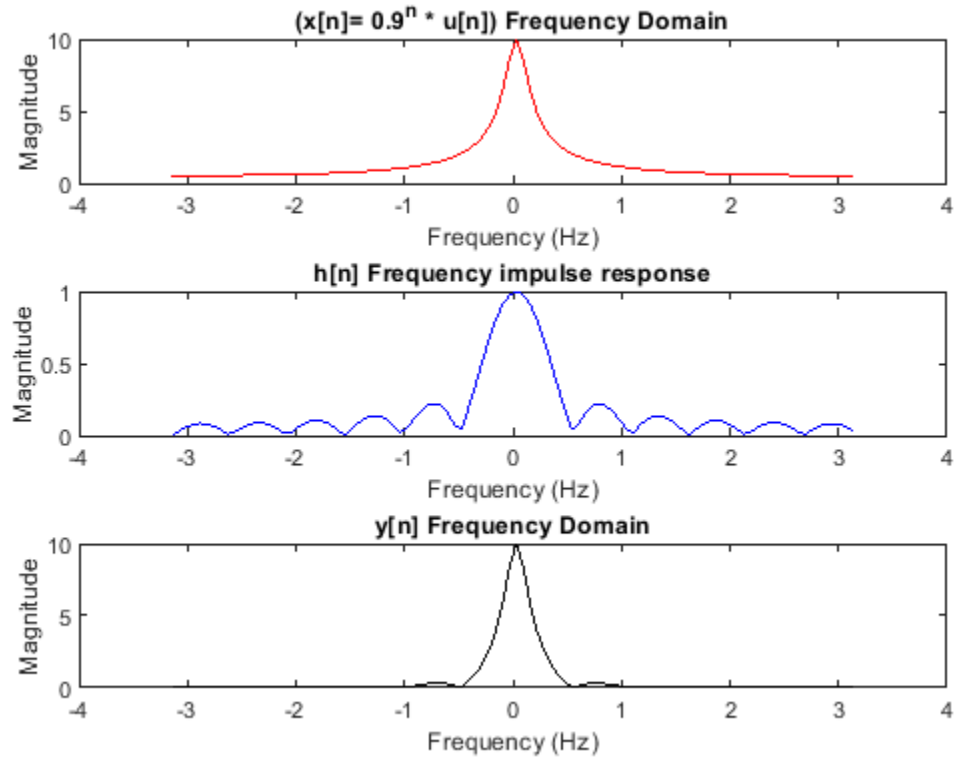
```
axis([0 NoS1 0 1]);  
title('y[n] Time Domain');  
xlabel('Time (sample)');  
ylabel('Amplitude');
```



## Convolution in frequency doamain (part c)

```
% calculate DTFT of signals  
[f_x1, fft_x1] = DTFT(x1, NoS1);  
[f_h, fft_h] = DTFT(h, NoS1);  
[f_y1, fft_y1] = DTFT(y1, NoS1);  
  
% show results in frequency domain  
figure();  
subplot(3, 1, 1);  
plot(f_x1, fft_x1, 'r');  
title('(x[n]= 0.9^n * u[n]) Frequency Domain');  
xlabel('Frequency (Hz)');  
ylabel('Magnititude');  
subplot(3, 1, 2);  
plot(f_h, fft_h, 'b');  
title('h[n] Frequency impulse response');  
xlabel('Frequency (Hz)');  
ylabel('Magnititude');  
subplot(3, 1, 3);  
plot(f_y1, fft_y1, 'k');
```

```
title('y[n] Frequency Domain');  
xlabel('Frequency (Hz)');  
ylabel('Magnitude');
```



## Convolution in time doamain (part d)

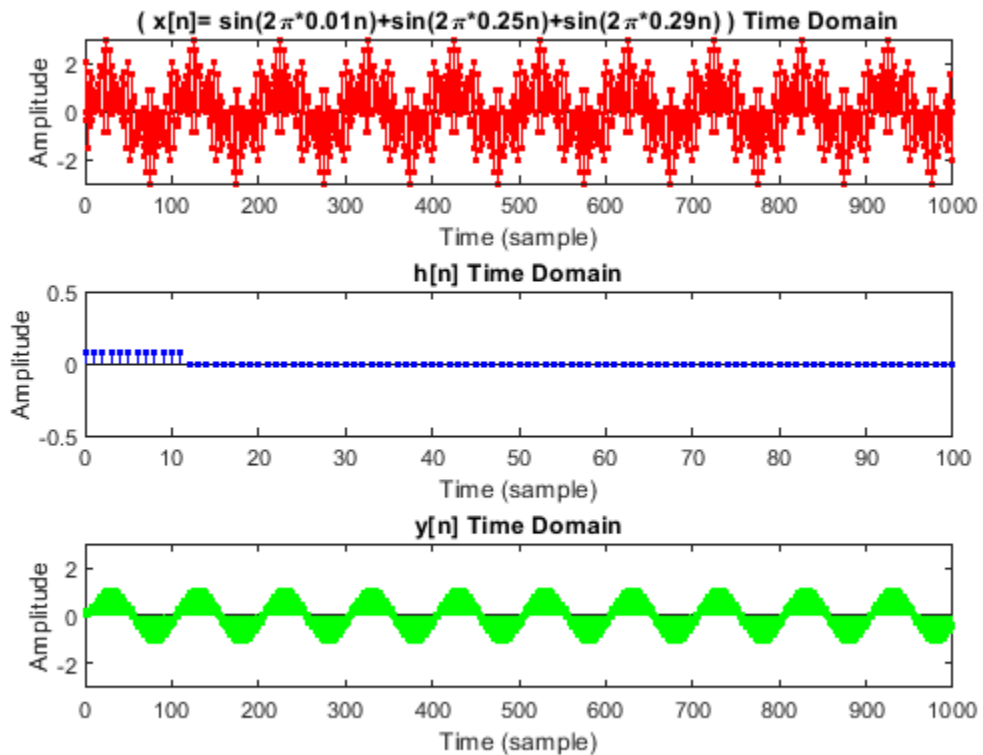
```
% define parameters  
NoS2 = 1000;  
n2 = 0 : 1000;  
x2 = sin(2 * pi * 0.01 * n2) + sin(2 * pi * 0.25 * n2) + sin(2 * pi *  
    0.29 * n2);  
  
% calculate convolution  
[y2, s2] = convolution(h, n1, x2, n2);  
  
% show results in time domain  
figure();  
subplot(3, 1, 1);  
stem(n2, x2, 'r.');
```

axis([0 NoS2 -3 3]);  
title('( x[n]= sin(2\pi\*0.01n)+sin(2\pi\*0.25n)+sin(2\pi\*0.29n) ) Time  
Domain');

```
xlabel('Time (sample)');  
ylabel('Amplitude');
```

subplot(3, 1, 2);  
stem(n1, h, 'b.');

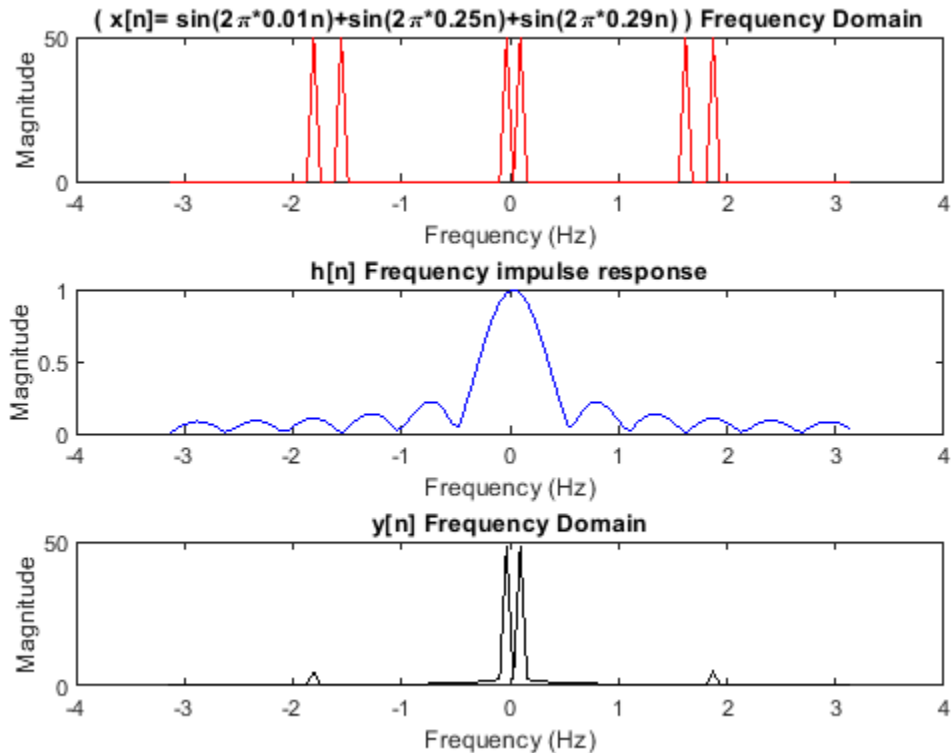
```
axis([0 NoS1 -0.5 0.5]);  
title('h[n] Time Domain');  
xlabel('Time (sample)');  
ylabel('Amplitude');  
subplot(3, 1, 3);  
stem(s2, y2, 'g.');
```



## Convolution in frequency doamain (part d)

```
% calculate DTFT of signals  
[f_x2, fft_x2] = DTFT(x2, NoS1);  
[f_y2, fft_y2] = DTFT(y2, NoS1);  
  
% show results in frequency domain  
figure();  
subplot(3, 1, 1);  
plot(f_x2, fft_x2, 'r');  
title('( x[n]= sin(2\pi*0.01n)+sin(2\pi*0.25n)+sin(2\pi*0.29n) )  
Frequency Domain');  
xlabel('Frequency (Hz)');  
ylabel('Magnititude');  
subplot(3, 1, 2);
```

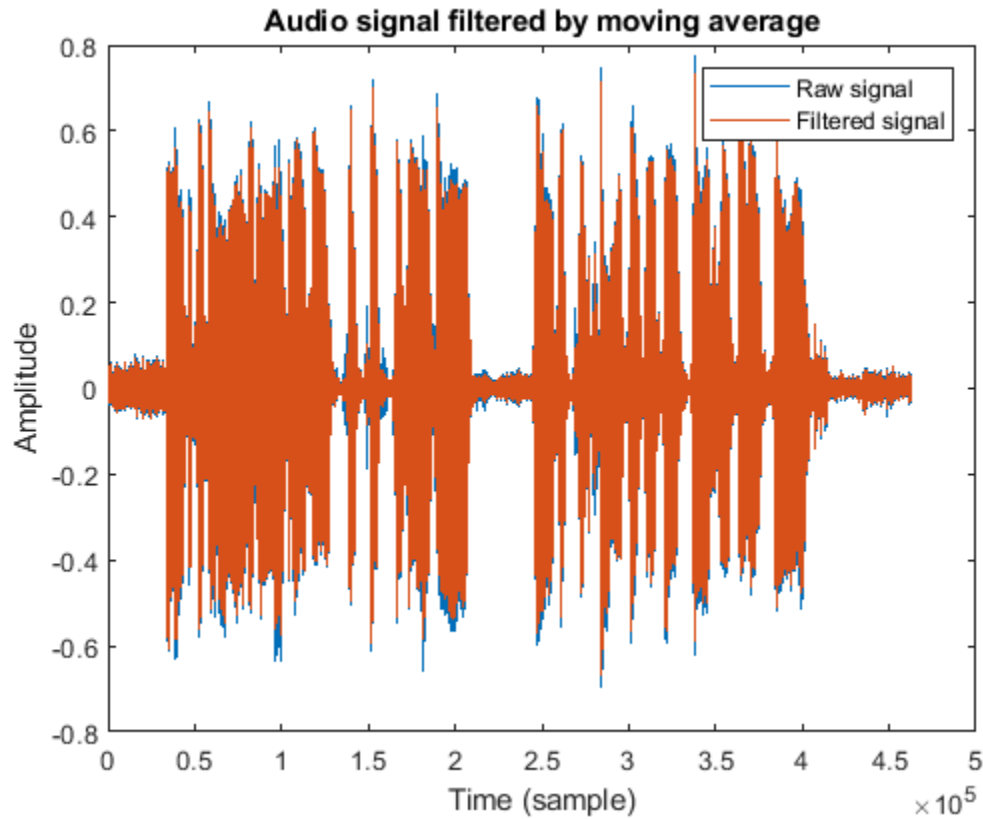
```
plot(f_h, fft_h, 'b');  
title('h[n] Frequency impulse response');  
xlabel('Frequency (Hz)');  
ylabel('Magnitude');  
subplot(3, 1, 3);  
plot(f_y2, fft_y2, 'k');  
title('y[n] Frequency Domain');  
xlabel('Frequency (Hz)');  
ylabel('Magnitude');
```



## Part e

```
% read audio file  
[sig, Fs] = audioread('voice.wav');  
% sound(sig, Fs);  
  
% calculate convolution  
oSig = convolution(h, n1, sig, 0:length(sig));  
  
% audio results  
figure();  
plot(sig);  
hold on;  
plot(oSig);  
title('Audio signal filtered by moving average');  
xlabel('Time (sample)');
```

```
ylabel('Amplitude');  
legend('Raw signal', 'Filtered signal')  
  
% generate output sound and write audio  
audiowrite('voice_q1_output.wav', oSig, Fs);  
% sound(oSig, Fs);
```



## M point DTFT (part a)

```
function [f, fft_sig] = DTFT(signal, M)  
    if (length(signal) < M)  
        error('M point is not larger than signal length!!');  
    end  
    fft_sig = abs(fftshift(fft(signal, M)));  
    f = linspace(-pi, pi, M);  
end
```

## Convolution with time index (part b)

```
function [y, s] = convolution(u, n, v, m)  
    y = conv(u, v);  
    s = (m(1) + n(1)) : (m(end) + n(end));  
end
```

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