Digital Signal Processing [Lab-3]

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Objective:

In this lab we used deconvolution for finding (impulse response) h

%(called System identification) from x (input) & y(output) and for finding %input (called Input estimation) from (inpulse response) h & y (output) %We found the least mean square to find the error from the deconvolution %from the the original answer

Program:

```
clc;
clear all;
close all;
% * \mid *Class example for finding impulse response by input, output signal * \mid
x=([3, 2,1]);
y=([3,5,3,1]);
%Making of X matrix
[x row, x col]=size(x);%Size of x
[y_row, y_col]=size(y);%Size of y
h_size=y_col-x_col+1;%Finding the size of impulse response
h_col=h_size; % impulse response column size
size_x_col=x_col+h_col-1;%X column length or simply y_col
X=zeros(size_x_col,h_col); Making X matrix full of zeros
k=0;%variable for shifting in the matrix
for i=1:h col%Looping through column
   for j=1:x_col%Looping through the row
    X(j+k,i)=x(j);
```

```
end
   k=k+1;
end
xtx=transpose(X)*X;%Finding X*Xtranspose
det(xtx); % Finding the determinant of X*Xtranspose
h1=inv(xtx); %Finding the inverse of this xtx square matix
h2=h1*transpose(X); %Multiplyting the previous answer with Xtranspose
h_find=h2*transpose(y); %Multiplyting the previous answer with ytranspose
% * |*Finding impulse response by input, output signal from audio sample*|
%Functions used
% <include>X xy.m</include>
% <include>h finding.m</include>
[y, fs]=audioread('Signal_Processing_Audio.mp3');
y_n = y(:,1);
t=0:1/fs:5;%taking 5sec of samples
size_y = size(y_n);%Finding the size of size_y
[size_t_row, size_t_col] = size(t); %Finding the size of t matrix
y_fivesec=y_n(1:size_t_col); %Taking 5 sec of audio samples
zeros_to_add=mod( size(y_fivesec) , 512 ); % Finding the modulus of y_fivesec with 5
y_fivesec = vertcat(y_fivesec,zeros(171,1)); %171 zeros added to the input
no_interations=size(y_fivesec)/512;%no_iterations=431
vector_y=transpose(y_fivesec);%Making to horizontal matrix
isvector(vector_y);%Finding if vector_y is a vector
h=h1;
input_matrix=zeros(431,512); Input matrix for making input to groups of 512
k=1;
for i=1:431%Iterating over 431 rows
    input=vector y(k:k+511); %Selecting 512 elements
    input_matrix(i,:)=input;%Adding to the ith row
    k=k+512;
end
block_conv=zeros(431,572); %Making a matrix of 431x572 of zeros
for i=1:431
    block_conv(i,:)=conv(input_matrix(i,:),h); %y output stored row wise
end
h_matrix=zeros(61,431); % Making a matrix for keeping h of 431 inputs
for i=1:431
    x=input_matrix(i,:);%x taken from input matrix
    y=block conv(i,:); %y taken from output matrix
    X=X_xy(x,y); Converting to X matrix from x vector and y output
    h_block=h_finding(X,y);%finding the h
    h_matrix(:,i)=h_block; %keeping the impulse response in i column
end
%For finding the least square error of the impulse response for y(output)
```

```
h_leastsq_error=zeros(1,431); % Vector to store 431 least square numbers
for i=1:431
   h_samples=transpose(h_matrix(:,i));
   h error=h samples-h; Taking the difference between the h found from deconvolut
    sum=0;%for adding the least squares
    for j=1:61
       sum=sum+(h_error(:,j)*h_error(:,j));%Adding hte least squares to sum varia
    end
    h_leastsq_error(1,i)=sum; %ith sample's least sqyuare error in ith position of
end
%Finding h for noisy output
h noisy matrix=zeros(61,431);
for i=1:431
   x=input matrix(i,:);%x taken from input matrix
   y_inverted=mdfdNoiseAddBlockData(:,i); %y taken from noisy-output matrix
   y=transpose(y_inverted);%Transpose of y_inverted variable
   X=X_xy(x,y); Converting to X matrix from x vector and y output using this fund
   h block=h finding(X,y); % finding the h using the function h finding
    h_noisy_matrix(:,i)=h_block; %keeping the impulse response in i column
end
%For finding the least square error of the impulse response for noisy y
noisy h leastsq error=zeros(1,431); Making a matrix of 1x431
for i=1:431
   noisy_h_samples=transpose(h_noisy_matrix(:,i)); %Trnaspose of ith sample of noi
   noisy_h_error=noisy_h_samples-h;%Finding the error in the noisy h sample from
    sum=0;%sum variable used for adding the quare error
    for j=1:61
       sum=sum+(noisy h error(:,j)*noisy h error(:,j));%Adding the square error t
    end
    noisy_h_leastsq_error(1,i)=sum; %Least square error of ith sample added to the
end
       Error using conv (line 26)
       A and B must be vectors.
       Error in Lab3submit (line 79)
           · Class example for finding input signal from impulse response and output
 signal
```

```
h=([1,1]);
y=([3,5,3,1]);

%Making of h matrix
[h_row, h_col]=size(h);%Size of h
[y_row, y_col]=size(y);%Size of y

x_col=y_col-h_col+1;
H=zeros(y_col,x_col);
k=0;%variable for shifting in the matrix
for i=1:x_col%Looping through column
```

Finding input signal from impulse response and output signal using audio signal

```
%Functions used
% <include>H_hy.m</include>
% <include>x finding.m</include>
[y, fs]=audioread('Signal_Processing_Audio.mp3');
y_n = y(:,1);
t=0:1/fs:5;%taking 5sec of samples
size_y = size(y_n);%Finding the size of size_y
[size_t_row, size_t_col] = size(t);%Finding the size of t matrix
y fivesec=y n(1:size t col); Taking 5 sec of audio samples
zeros to add=mod( size(y fivesec) , 512 );
y_fivesec = vertcat(y_fivesec,zeros(171,1)); %171 zeros added to the input
no_interations=size(y_fivesec)/512;%no_iterations=431
vector y=transpose(y fivesec); %Making to horizontal matrix
isvector(vector_y); % checking if vector_y is a vector
input_matrix=zeros(431,512); Input matrix for making input to groups of 512
k=1;
for i=1:431%Iterating over 431 rows
    input=vector y(k:k+511); % Selecting the 512 blocks of elements from the input
    input_matrix(i,:)=input;%Adding to the ith row
    k=k+512;
end
block_conv=zeros(431,572);
for i=1:431
    block_conv(i,:)=conv(input_matrix(i,:),h); %y output stored row wise
end
x_matrix=zeros(512,431); %Making a matrix for keeping x of 431 inputs
k=1;
for i=1:431
    y=block_conv(i,:); %y taken from output matrix
    H=H_hy(h,y); Finding H matrix from h, y from function H_hy
    x block=x finding(H,y); %Finding x from H matrix and y from the function x find
    x_found(k:k+511)=transpose(x_block);%Finding the input x from block inputs and
    k=k+512;
```

end

```
Finding the least square error of the input got from output
        x error=x found-vector y; Finding the error between the x found from deconvolution
        x_leastsq=0;%variable for finding the least square error
        for i=1:220672
            x_leastsq=x_leastsq + x_error(:,i)*x_error(:,i); %least square error getting ad
        end
        %Finding x for noisy output
        x_noisy_matrix=zeros(512,431);
        k=1;
        for i=1:431
            y inverted=mdfdNoiseAddBlockData(:,i); %y taken from noisy-output matrix
            y=transpose(y_inverted);
            H=H_hy(h,y);%H matrix found from h,y by function H_hy
            x_block=x_finding(H,y); %x input found from H matrix and y by function x_findin
            x noisy found(k:k+511)=transpose(x block); %Finding x from block inputs and app
            k=k+512;
        end
        %Finding the least square error of the input got from noisy output
        x noisy error=x noisy found-vector y; %Finding error difference in the input
        x noisy leastsq=0; %varible for finding the least square error
        for i=1:220672
            x_noisy_leastsq= x_noisy_leastsq + (x_noisy_error(:,i)*x_noisy_error(:,i));%Su
        end
Results:
        * * |*Result for class problem for finding impulse response from y,x*|
        h_find
        · Plot for the least square error value for impulse response
        figure;plot(h_leastsq_error);
        title('Least Square Error of Speech Signal'); xlabel('Block of input'); ylabel('Ampl
        Plot for the least square error value of impulse response for noisy
        output
        figure;plot(noisy_h_leastsq_error);
        title('Least Square Error of Speech Signal for noisy output'); xlabel('Block of inp
        %*|*Result for class problem for finding input from h,y*|
        x find
        %* | *Result for finding input from h,y* |
        x_leastsq
        %*|*Result for finding input from h and noisy output*|
        x_noisy_leastsq
```

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