Name: Thong Thach

SID:862224662

UCR net ID: <u>tthac005@ucr.edu</u>

Session: 022, Thursday 2-4:50

1)

Assume that the system between the voice x[n] from your mouth and the acoustic signal y[n] picked up by a microphone is linear and time-invariant, and hence

$$y[n] = h[n] * x[n] = \sum_{l=1}^{L} h[l]x[n-l]. \tag{1}$$

To see the effect of the echo distortion, assume

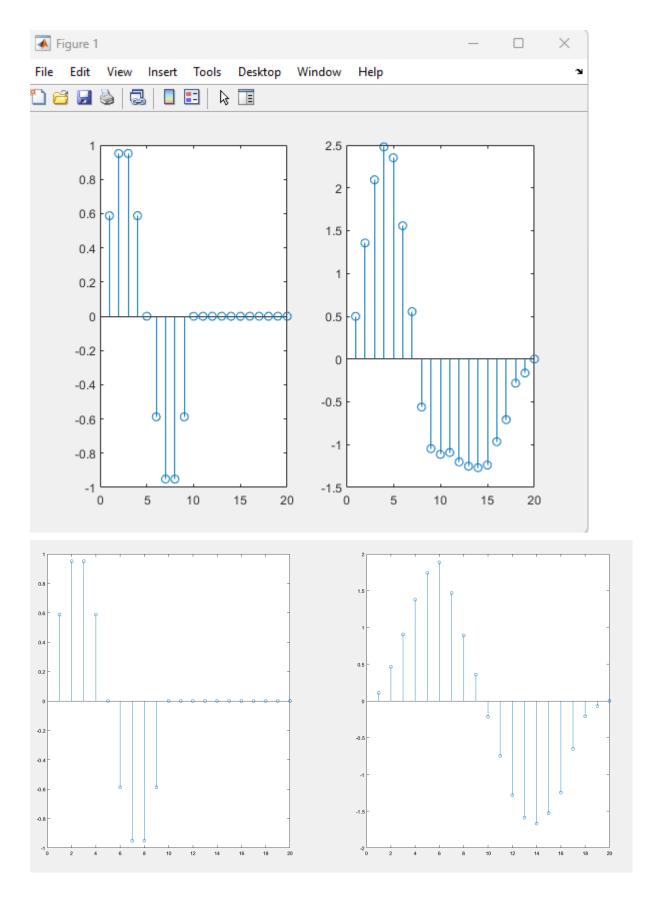
$$x[n] = \sin(\frac{\pi}{5}n)(u[n] - u[n-10]) \tag{2}$$

and also choose the echo coefficients h[l] for all  $0 \le l \le 10$  randomly. Here L = 10. Plot x[n] and y[n] for  $0 \le n \le 20$ . Discuss the impact of h[n] on y[n] in relation to x[n].

For the coding part:

```
n=1:10;
 1
          x=sin(pi/5 * n);
 2
 3
          x=[x,zeros(1,10)];
          h=rand(1,11);
4
 5
          y=conv(x,h);
          figure(1);
 6
          subplot(121);
7
          stem(x(1:20));
 8
          subplot(122);
9
          stem(y(1:20));
10
```

For the plot:



Discuss the impact of h[n] on y[n] in relation to x[n]:

The impact of h[n] on y[n] in relation to x[n] is h[n] created the noise on x[n] original signal pulse, which makes the y[n] signal pulses are the distortion from x[n] signal pulses. The distortion effects deform the signal of the original waveform. It first appears that the waveform shifted up after a few first signal pulses relative to the beginning of the start pulse; after that, the original position signal pulses have been dropped to -0.9, but the signal pulses after distortion have oscillated between the range -0.5 to -1.5. Finally, the original signal pulses have been shifted down from 0 to negative values on the y-axis. A lot of noises have interrupted the original signal pulses. That's distortion effects on h[n].

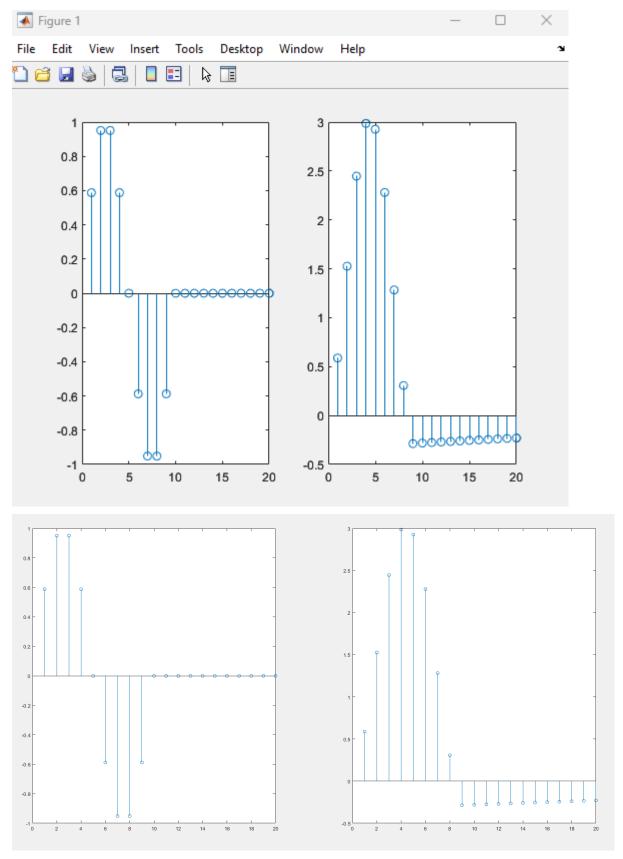
- Assume the same x[n] in (2) but  $h[n] = 0.98^n u[n]$  which is the impulse response of a first-order feedback system.
  - a) Compute and plot y[n] = x[n] \* h[n] for  $0 \le n \le 20$  (with a large L such that h[L] is negligible). Discuss the distortion effect by h[n].

# For the coding part:

```
n=1:10;
x=sin(pi/5 * n);
x=[x,zeros(1,10)];
n=1:1000;
h=0.98.^(n-1);

y=conv(x,h);
figure(1);
subplot(121);
stem(x(1:20));
subplot(122);
stem(y(1:20));
```

For the plot:



The distortion effects deform the signal of the original waveform. It first appears that the waveform shifted up after a few first signal pulses relative to the beginning of the start pulse;

after that, the original position signal pulses have been dropped to -0.9, but the signal pulses after distortion have been smoothed out to -0.2. Finally, the original signal pulses have been shifted down from 0 to -0.2. A lot of noises have interrupted the original signal pulses. That's distortion effects on h[n].

b)

b) Compute and plot v[n] = g[n] \* y[n] for  $0 \le n \le 20$  where  $g[n] = \delta[n-5] - 0.98\delta[n-6]$ . Discuss the quality of g[n] as an echo cancellation filter.

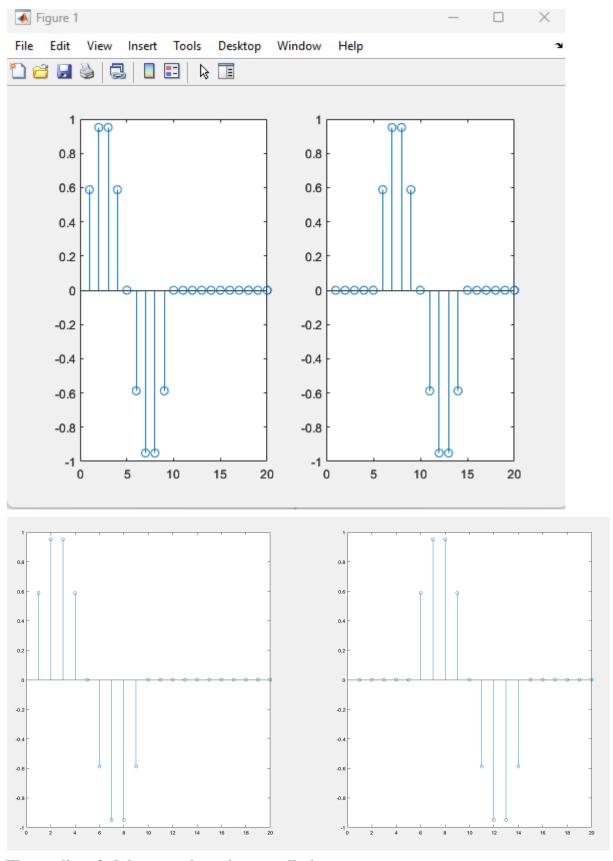
## For the coding part

```
n=1:10;
x=sin(pi/5 * n);
x=[x,zeros(1,10)];
n=1:1000;
h=0.98.^(n-1);

y=conv(x,h);

g=zeros(1,10);
g(6)=1;
g(7)=-0.98;
v=conv(y,g);
figure(1);
subplot(121);
stem(x(1:20));
subplot(122);
stem(v(1:20));
```

# For the plotting part:



The quality of g[n] as an echo noise cancellation:

The quality of g[n] is trying to smooth out the signal pulses from (a) in order to form the signal back to the original signal pulses. Using g[n] noise filter, it tries to cancel all the distortion signal pulses from (a) back to the original signal pulses quality. That's the quality of g[n] noise filter. As a comparison, g[n] has brought the distorted signal pulses from -0.2 back to -0.9 by canceling all the noises from (a)

3)

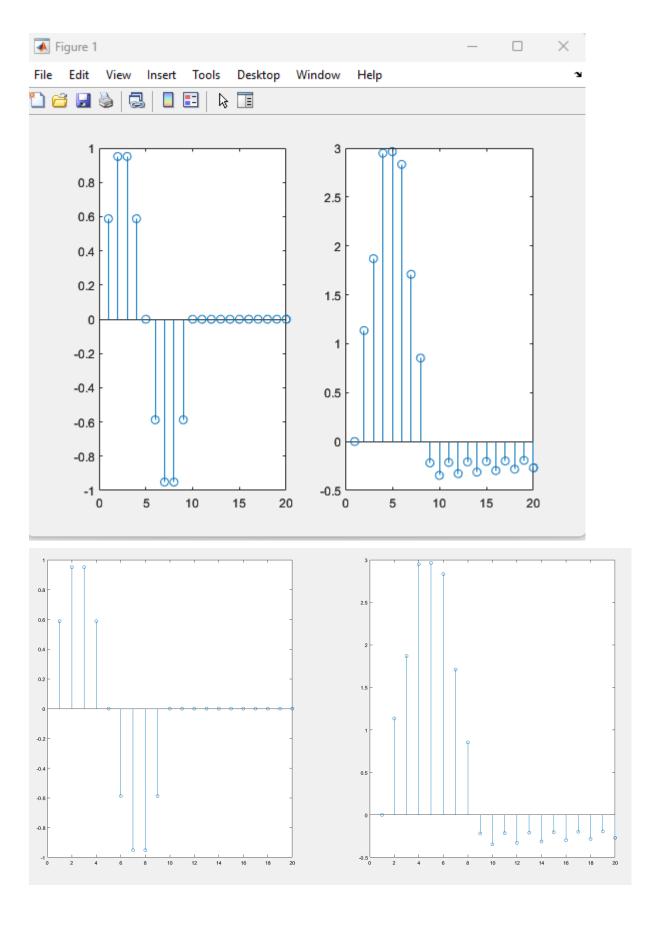
- 3) If  $h[n]=a^nu[n]-b^nu[n]$  (which is the impulse response of a second-order feedback system), then its DTFT is  $H(f)=\frac{1}{1-ae^{-j2\pi f}}-\frac{1}{1-be^{-j2\pi f}}=\frac{(a-b)e^{-j2\pi f}}{1-(a+b)e^{-j2\pi f}+abe^{-j4\pi f}}$ . A good inverse filter of H(f) has the frequency response  $G(f)=1-(a+b)e^{-j2\pi f}+abe^{-j4\pi f}$ . Assume the same x[n] in (2) but  $h[n]=0.98^nu[n]-(-0.95)^nu[n]$ .
  - a) Compute and plot y[n] = x[n] \* h[n] for  $0 \le n \le 20$  (with a large L such that h[L] is negligible). Discuss the distortion effect by h[n].

## For the coding part:

```
n=1:10;
x=sin(pi/5 * n);
x=[x,zeros(1,10)];
n=1:1000;
h=0.98.^(n-1) - (-0.95) .^(n-1);

y=conv(x,h);
figure(1);
subplot(121);
stem(x(1:20));
subplot(122);
stem(y(1:20));
```

#### For the plotting part:



The distortion effects deform the signal of the original waveform. It first appears that the waveform shifted up after a few first signal pulses relative to the beginning of the start pulse; after that, the original position signal pulses have been dropped to -0.9, but the signal pulses after distortion have been smoothed out to -0.2. Finally, the original signal pulses have been shifted down from 0 to -0.2. A lot of noises have interrupted the original signal pulses. That's distortion effects on h[n].

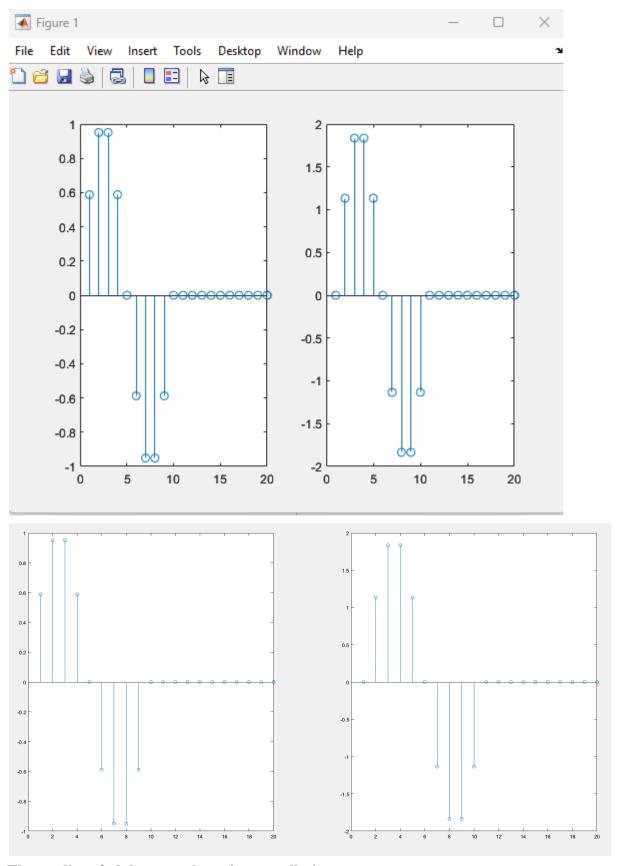
b)

b) Compute and plot v[n] = g[n] \* y[n] for  $0 \le n \le 20$  with  $g[n] = \delta[n] - (a+b)\delta[n-1] + ab\delta[n-2]$ , a=0.98 and b=-0.95. Discuss the quality of g[n] as an echo cancellation filter.

## For the coding part:

```
n=1:1000;
h=0.98.^(n-1) - (-0.95).^(n-1);
y=conv(x,h);
g=zeros(1,5);
a=0.98;
b=-0.95;
g(1)=1;
g(2)=-(a+b);
g(3)=a.*b;
v=conv(y,g);
figure(1);
subplot(121);
stem(x(1:20));
subplot(122);
stem(v(1:20));
```

For the plotting part:



The quality of g[n] as an echo noise cancellation:

The quality of g[n] is trying to smooth out the signal pulses from (a) to form the signal back to the original signal pulses. Using g[n] noise filter, it tries to cancel all the distortion signal pulses from (a) back to the original signal pulses quality. That's the quality of g[n] noise filter. As a comparison, g[n] has brought the distorted signal pulses from -0.2 back to -0.9 by canceling all the noises from (a)