
EE 427 Lab 1

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Part A:

```
% Part 1:

A = 2;
f = 4;
n = -25:24;
fs = 100;

x = A*cos(2*pi*(f/fs)*n);
figure();
stem(x);
title('Sampled signal, 2*cos(8*pi*t)');
ylabel('amplitude');
xlabel('sample index (n)');

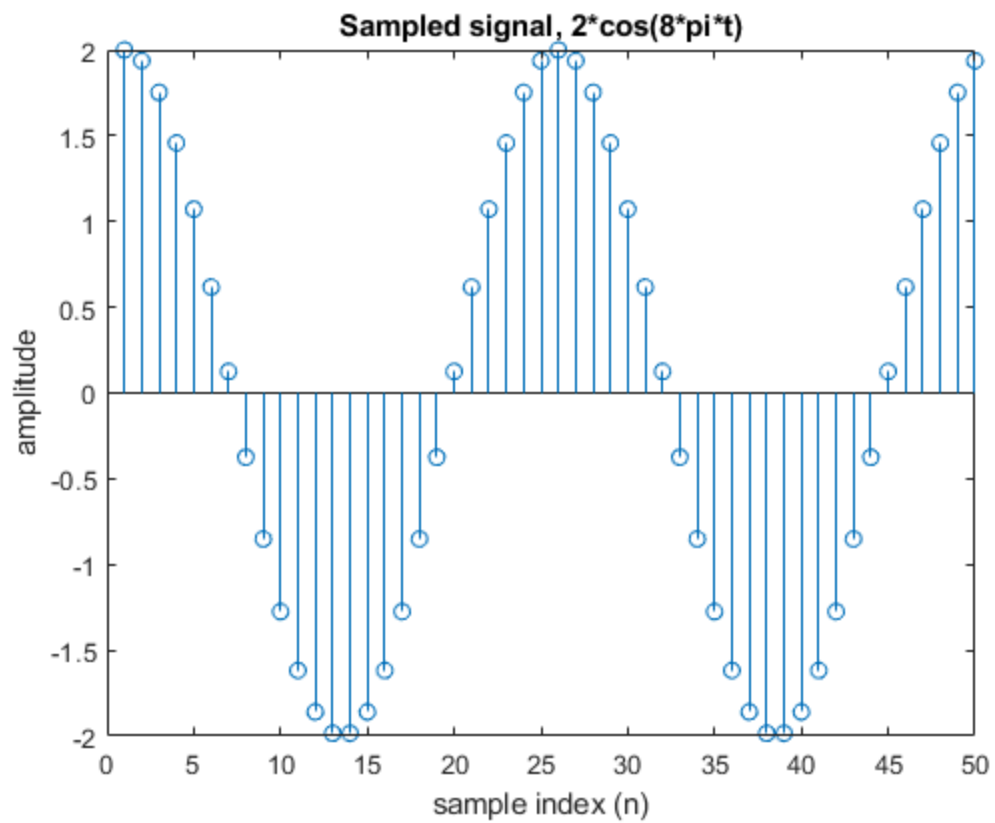
% Part 2:
pic = imread('EE427Project1A2.jpeg');
pic = imrotate(pic,-90);
figure();
imshow(pic);
title('Work done by hand');

% Part 3:
L = length(x);
FFTLength = 2^nextpow2(L);
xPrime = fft(x,FFTLength)/L;
xPrime = fftshift(xPrime);

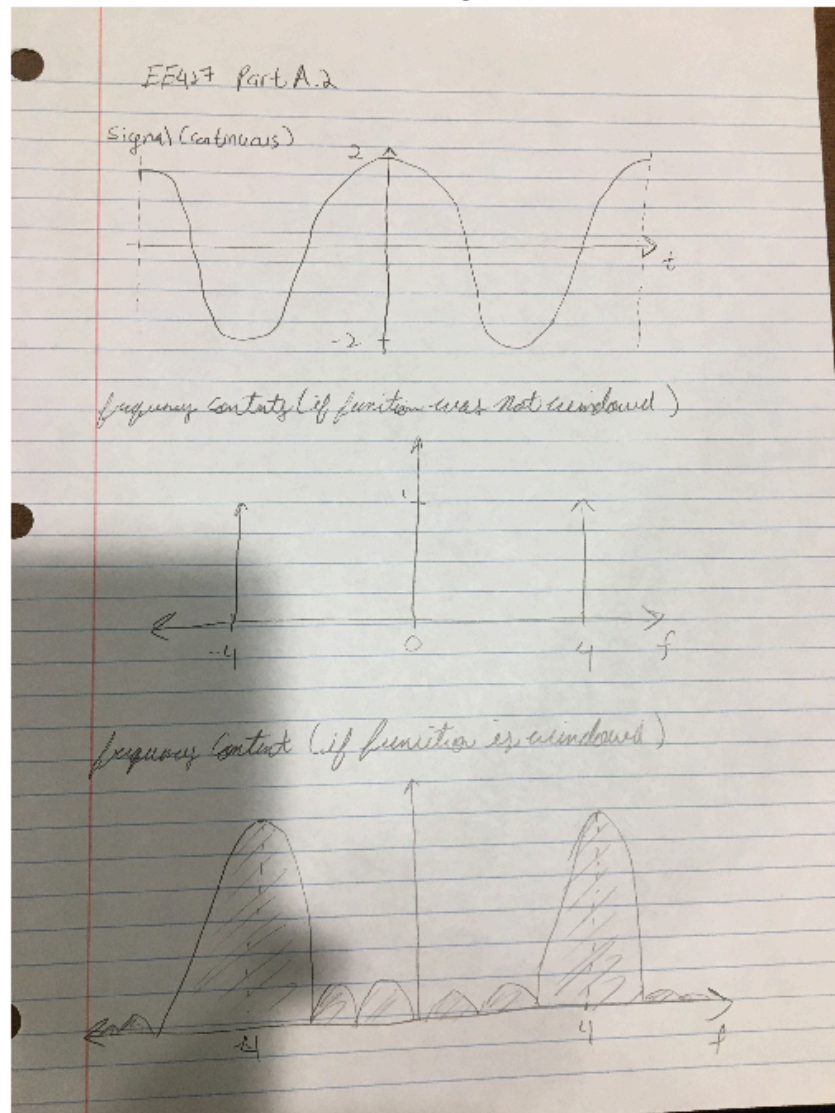
figure();
stem(linspace(-33, 32, length(xPrime)),abs(xPrime));
title('Fourier Transform of Sampled Cosine');
xlabel('frequency');
ylabel('Amplitude');

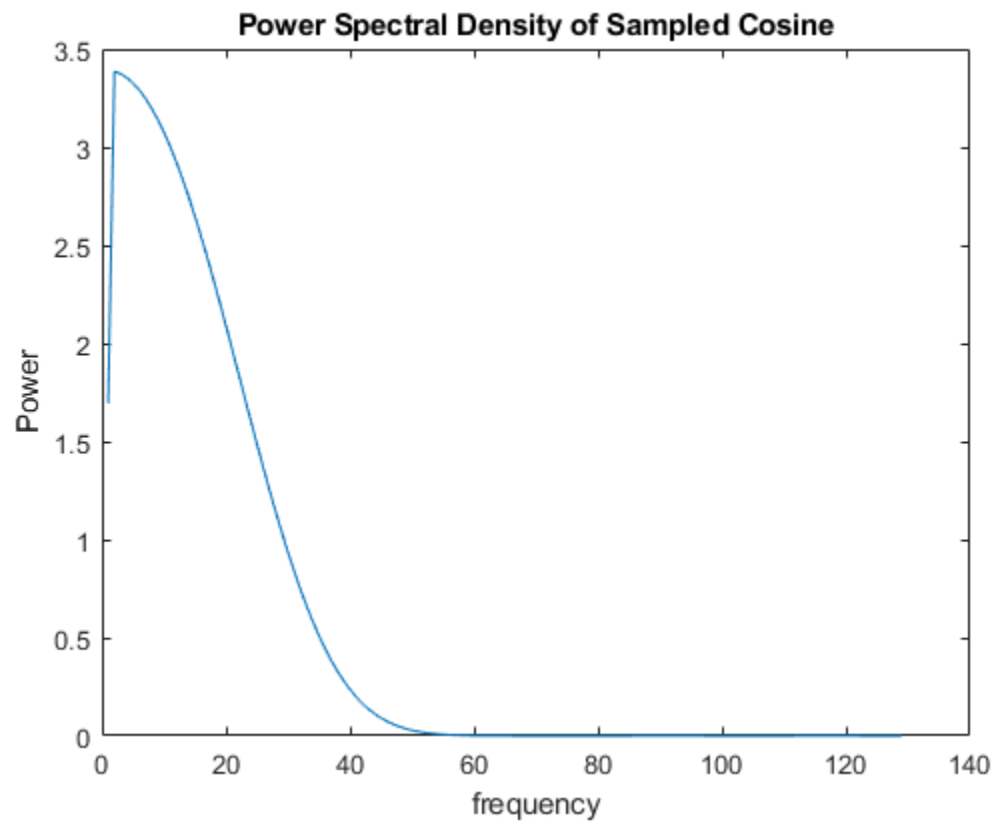
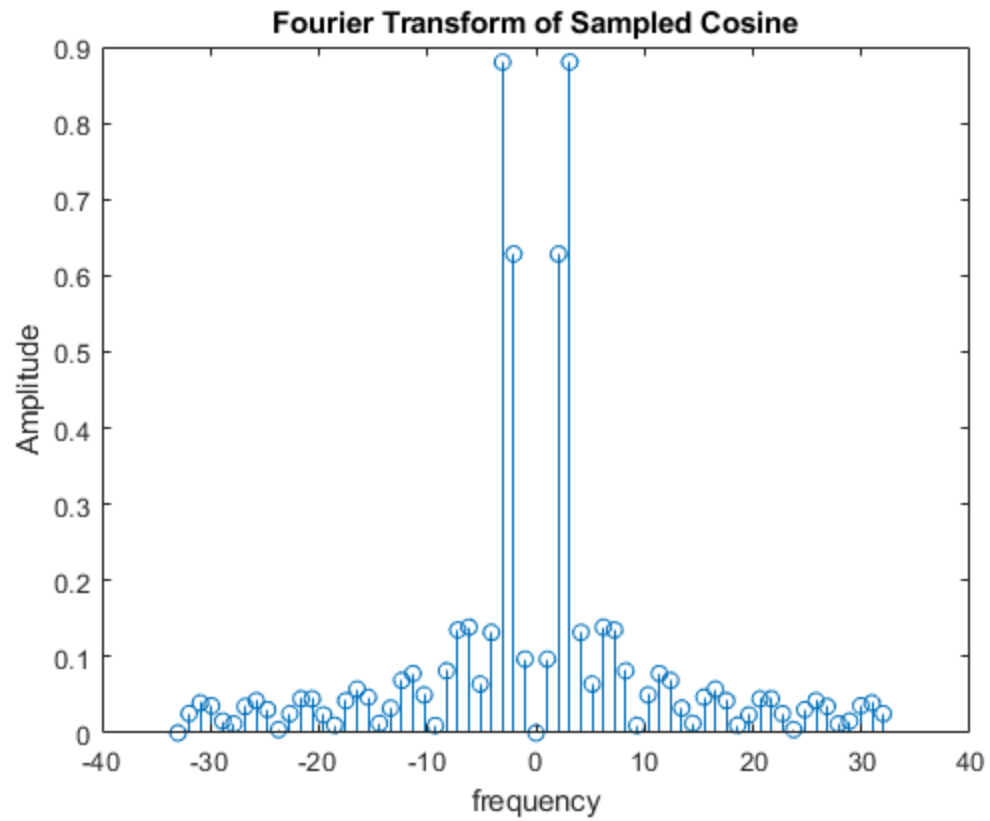
% Part 4:
figure();
XPSD = pwelch(x);
plot(XPSD);
title('Power Spectral Density of Sampled Cosine');
```

```
xlabel('frequency');  
ylabel('Power');  
  
% Part 5:  
% The FFT shows the magnitude at which each frequency is present in  
% the  
% signal being analyzed. The FFT function creates a double-sided  
% spectrum  
% of the magnitude of each frequency. The PSD, however, shows the  
% power  
% each frequency contributes to the signal. The pwelch function  
% creates a  
% single-sided distribution of the power.
```



Work done by hand





Part B:

```
POneWOne= audiorecorder(8000,16,1); POneWTwo= audiorecorder(8000,16,1); PTwoWOne= audiorecorder(8000,16,1); PTwoWTwo= audiorecorder(8000,16,1);

speaking part disp('start speaking'); recordblocking(POneWOne,2); % first person speaking first word
disp('end'); pause(1);

disp('start speaking'); recordblocking(POneWTwo,2); % first person speaking second word disp('end');
pause(1);

disp('start speaking'); recordblocking(PTwoWOne,2); % second person speaking first word disp('end');
pause(1);

disp('start speaking'); recordblocking(PTwoWTwo,2); % second person speaking second word disp('end');

% This part was commented out for the ease of publishing, since we
% didn't
% know when to start and stop speaking for the words, we pre-recorded
% them
% before publishing. The code was, however, used to capture each
% recording

wordOne = getaudiodata(POneWOne);
wordTwo = getaudiodata(POneWTwo);
wordThree = getaudiodata(PTwoWOne);
wordFour = getaudiodata(PTwoWTwo);

figure();
subplot(2,2,1);
plot(linspace(0,2,length(wordOne)),wordOne);
hold on
ylabel('intensity');
xlabel('time (s)');
title('First Person Word 1');
hold off
subplot(2,2,2);
plot(linspace(0,2,length(wordTwo)),wordTwo);
hold on
ylabel('intensity');
xlabel('time(s)');
title('First Person Word 2');
hold off
subplot(2,2,3);
plot(linspace(0,2,length(wordThree)),wordThree);
hold on
ylabel('intensity');
xlabel('time (s)');
title('Second Person Word 1');
hold off
subplot(2,2,4);
plot(linspace(0,2,length(wordFour)), wordFour);
hold on
```

```
ylabel('intensity');
xlabel('time (s)');
title('Second Person Word 2');
hold off

PSD1 = pwelch(wordOne);
PSD2 = pwelch(wordTwo);
PSD3 = pwelch(wordThree);
PSD4 = pwelch(wordFour);

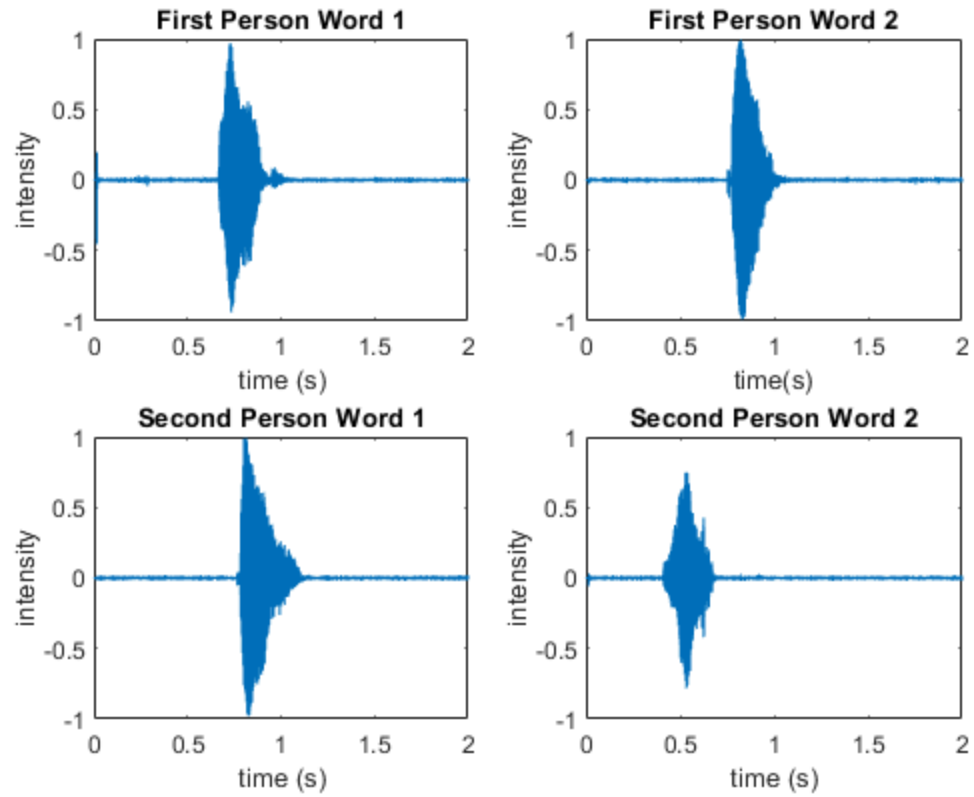
figure();
subplot(2,2,1);
plot(linspace(0,4000,length(PSD1)),PSD1);
hold on
ylabel('Power');
xlabel('frequency (f)');
title('First Person Word 1');
hold off
subplot(2,2,2);
plot(linspace(0,4000,length(PSD2)),PSD2);
hold on
ylabel('Power');
xlabel('frequency (f)');
title('First Person Word 2');
hold off
subplot(2,2,3);
plot(linspace(0,4000,length(PSD3)),PSD3);
hold on
ylabel('Power');
xlabel('frequency (f)');
title('Second Person Word 1');
hold off
subplot(2,2,4);
plot(linspace(0,4000,length(PSD4)), PSD4);
hold on
ylabel('Power');
xlabel('frequency (f)');
title('Second Person Word 2');
hold off

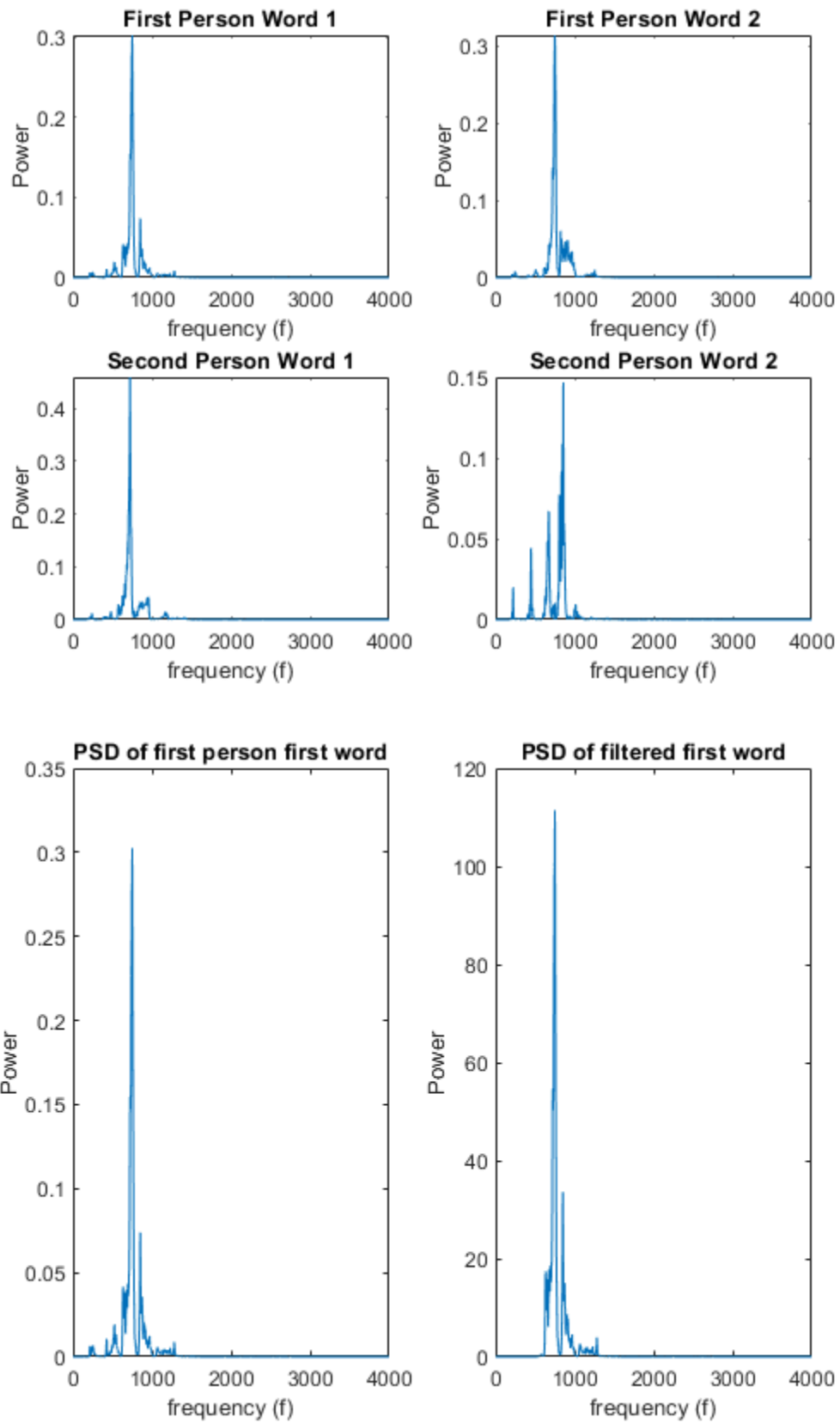
filtWordOne = 20*filter(filt,wordOne);

% The filter used was a high-pass filter with a transition band from
% 500 Hz
% to 600 Hz. Applying the high pass filter allowed us to take out the
% lower
% frequencies, which gave the voice a brighter tone.

figure();
subplot(1,2,1)
plot(linspace(0,4000,length(PSD1)),PSD1);
ylabel('Power');
xlabel('frequency (f)');
title('PSD of first person first word')
```

```
subplot(1,2,2)
plot(linspace(0,4000,length(pwelch(filtWordOne))),pwelch(filtWordOne));
ylabel('Power');
xlabel('frequency (f)');
title('PSD of filtered first word');
```





Part C

```
Original = audioread('EE427ProjectSoundClip.wav');

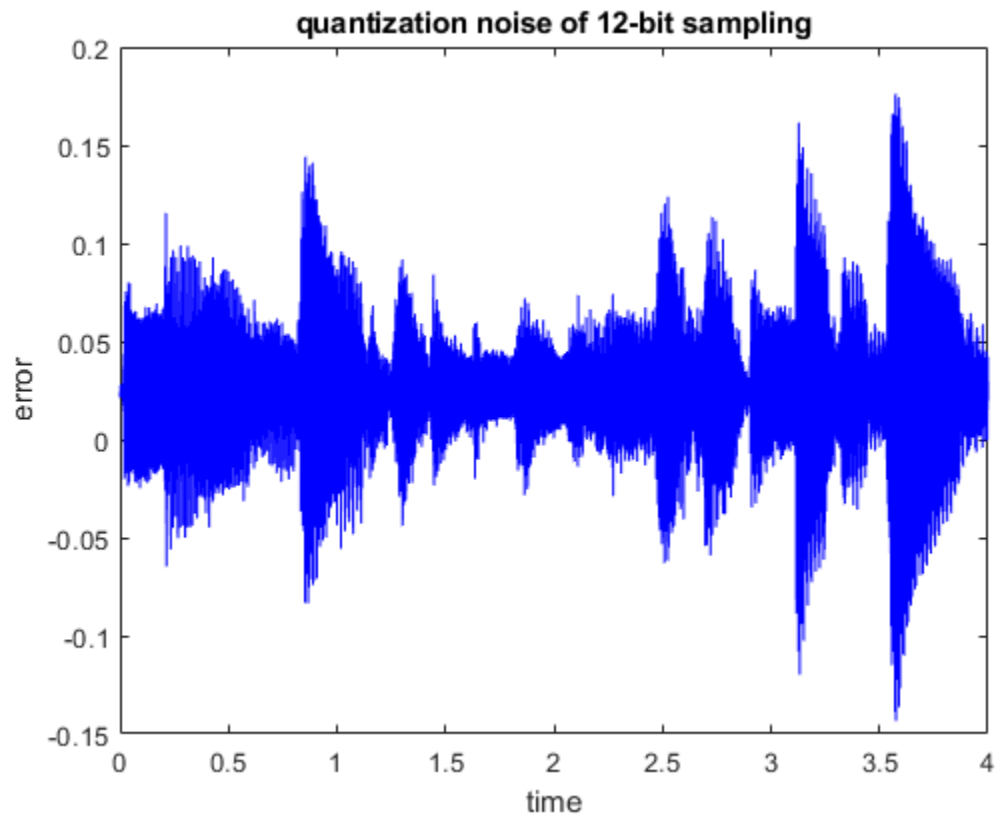
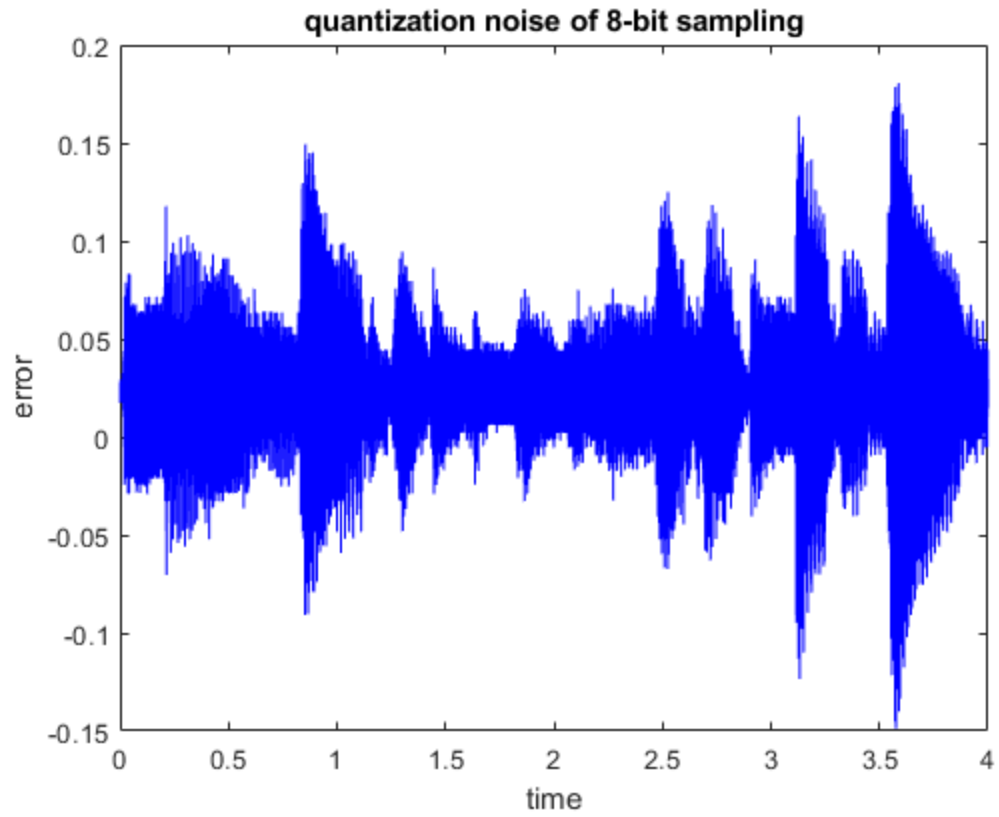
eightBit = double(uencode(Original,8));
eightBit = eightBit/max(eightBit);
eightBit = 2*(eightBit - ((1/2)*(max(eightBit)-
min(eightBit))+min(eightBit))));
d1 = Original(:,1)-eightBit;

figure();
plot(linspace(0,4,length(d1)),d1,'b');
title('quantization noise of 8-bit sampling');
xlabel('time');
ylabel('error');

twelveBit = double(uencode(Original,12));
twelveBit = twelveBit/max(twelveBit);
twelveBit = 2*(twelveBit - ((1/2)*(max(twelveBit)-
min(twelveBit))+min(twelveBit))));
d2 = Original(:,1)-twelveBit;

figure();
plot(linspace(0,4,length(d2)),d2,'b');
title('quantization noise of 12-bit sampling');
xlabel('time');
ylabel('error');

% when playing the audio back, it sounds similar to the original
% audio. However,
% the quantization adds a "grainy" noise to the background of the
% audio file.
```



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