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ECE 203 Lab 4: Synthesis of Sinusoidal Signals -- Music Synthesis 2/17/2017

Section 2.2(a)

```
fsamp = 11025; %sampling frequency
x1 = syn_sin(800, 100, fsamp, (0.5 - pi/3), -pi/3);
tt = 0:1/fsamp:0.5; %creates time axis with sinusoid above
soundsc(real(x1)) %plays the created sinusoid
```

Section 2.2(b)

```
x2 = syn_sin(1200, 80, fsamp, (0.8 + pi/4), pi/4);
soundsc(real(x2))

% The sinusoid from part a sounds lower in pitch than the
% sinusoid from part b. This is probably due to the higher
% frequency in x2 than x1.
```

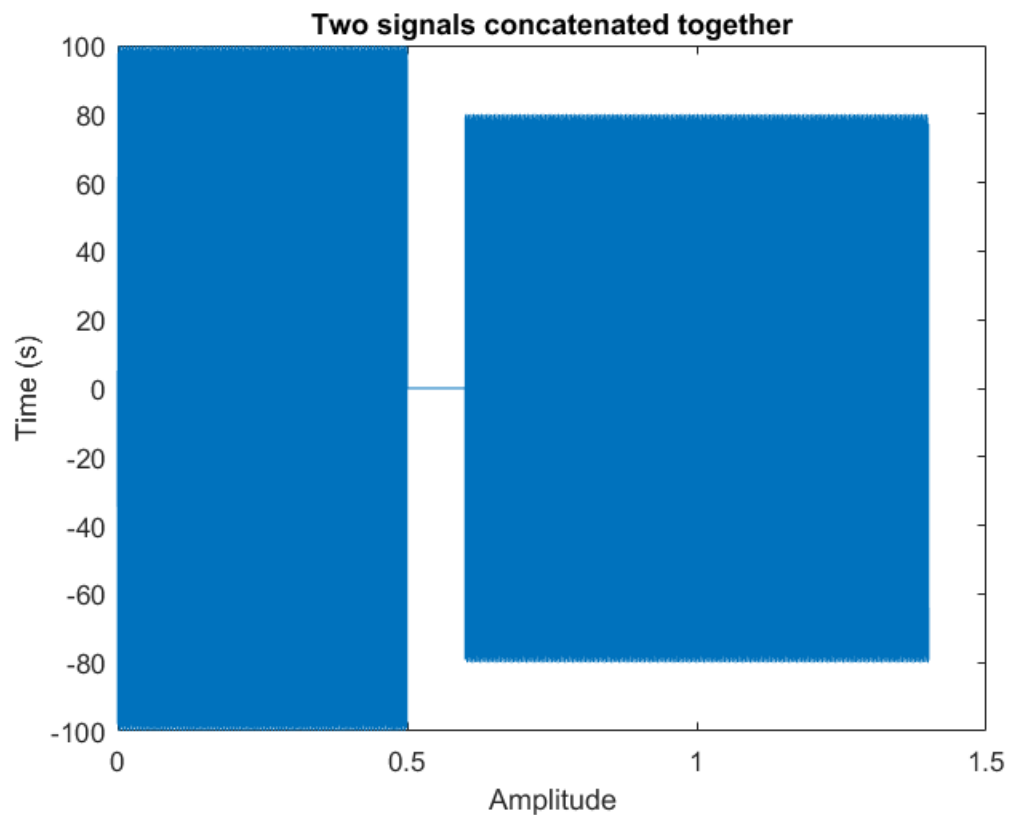
Section 2.2(c)

```
N = 1102; %One tenth of the sample rate
xx = [ x1, zeros(1,N), x2 ];
soundsc(real(xx)) %plays previous two signals in order with 0.1 second
gap of silence.
```

Section 2.2(d)

```
% plot of the two signals concatenated together.
tt = (1/fsamp)*(1:length(xx)); plot(tt,xx);
title('Two signals concatenated together');
xlabel('Amplitude');
ylabel('Time (s)');
```

Warning: Imaginary parts of complex X and/or Y arguments ignored



Section 2.2(e)

```
fs = 22050;
soundsc(real(xx), fs)
```

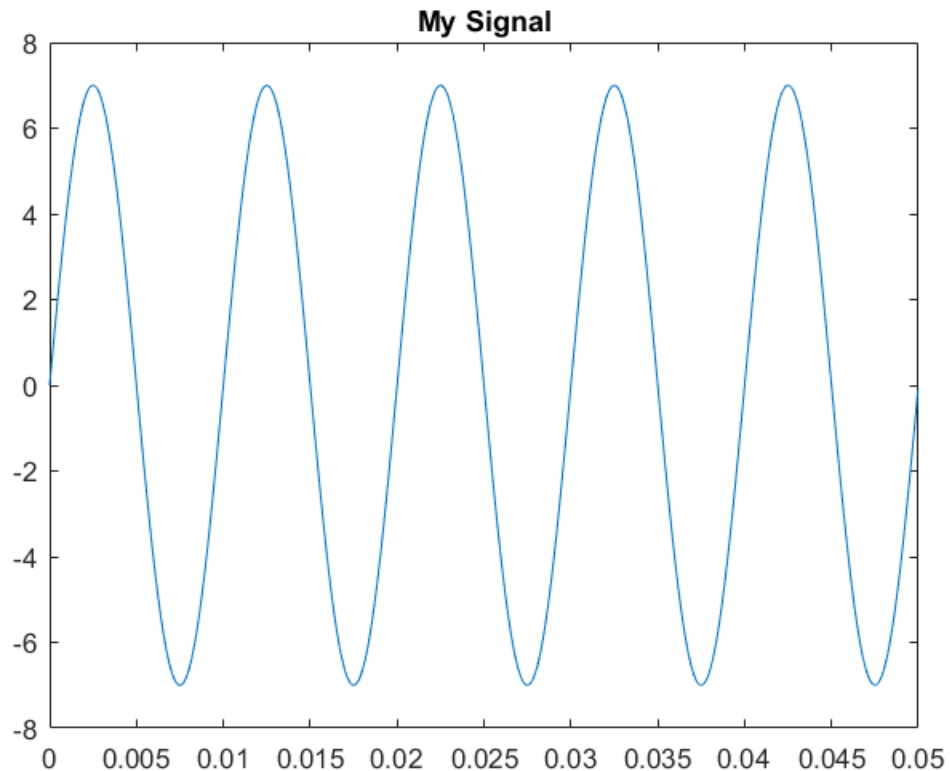
```
% Because we doubled the sampling period for the D-A converter
% the duration of the sound was much shorter, probably around half
% the time because the sampling rate was approximately doubled, and
% the pitch was much higher than the previous sounds for similar
reasons.
```

Section 2.3

```
x.Amp = 7;
x.phase = -pi/2;
x.freq = 100;
x.fs = 11025;
x.timeInterval = 0:(1/x.fs):0.05;
x.values = x.Amp*cos(2*pi*(x.freq)*(x.timeInterval) + x.phase);
x.name = 'My Signal';
x      %---- echo the contents of the structure "x"
plot(x.timeInterval, x.values)
title(x.name)
```

`x =`

```
      Amp: 7  
    phase: -1.5708  
     freq: 100  
      fs: 11025  
timeInterval: [1x552 double]  
  values: [1x552 double]  
    name: 'My Signal'
```



Section 3.1 (key2note function code)

```
%function xx = key2note(X, keynum, dur)  
% KEY2NOTE Produce a sinusoidal waveform corresponding to a  
%   given piano key number  
%  
% usage: xx = key2note (X, keynum, dur)  
%  
%   xx = the output sinusoidal waveform  
%   X = complex amplitude for the sinusoid, X = A*exp(j*phi).  
% keynum = the piano keyboard number of the desired note  
%   dur = the duration (in seconds) of the output note  
%  
%fs = 11025;
```

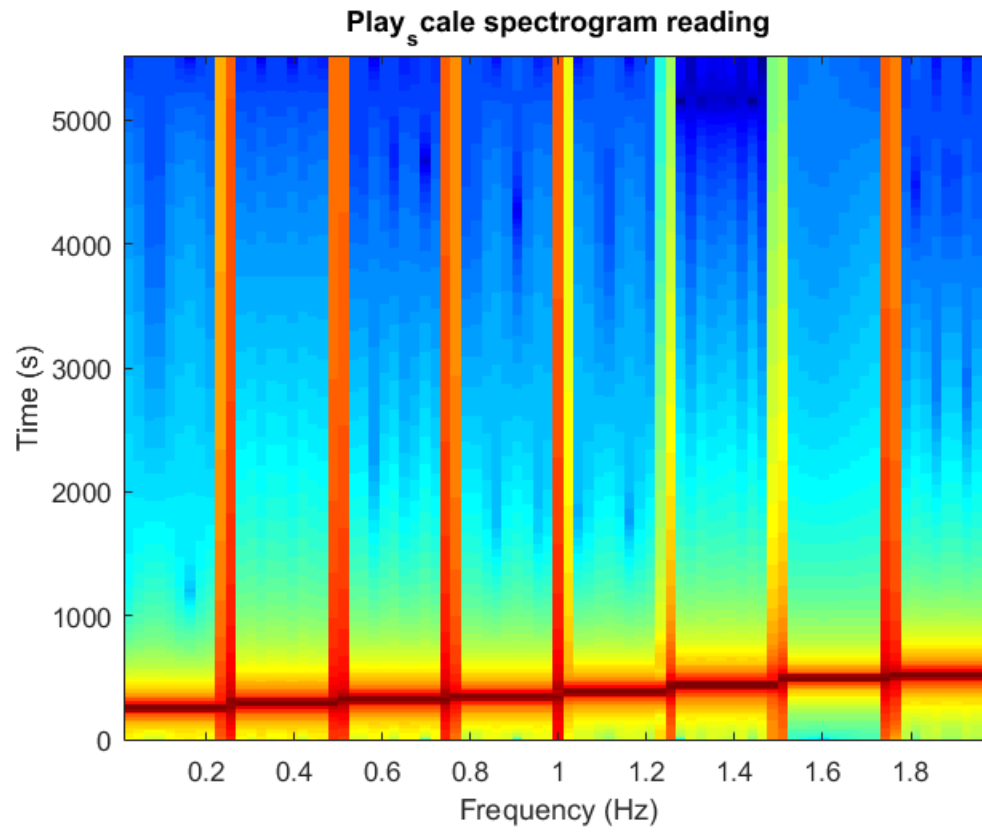
```
%tt = 0:(1/fs):dur;  
%freq = 440*2^((keynum-49)/12);  
%xx = real( X*exp(j*2*pi*freq*tt);
```

Section 3.2

```
%--- play_scale.m  
%---  
scale.keys = [40 42 44 45 47 49 51 52];  
%--- NOTES: C D E F G A B C  
% key #40 is middle-C  
%  
scale.durations = 0.25 * ones(1,length(scale.keys));  
fs = 11025;  
xx = zeros(1, sum(scale.durations)*fs+length(scale.keys));  
n1 = 1;  
for kk = 1:length(scale.keys)  
    keynum = scale.keys(kk);  
    dur = scale.durations(kk);  
    tone = key2note(10, keynum, dur);  
    n2 = n1 + length(tone) - 1;  
    xx(n1:n2) = tone;  
    n1 = n2 + 1;  
end  
soundsc(xx, fs)
```

Section 3.3

```
% The signal for the scale was created in section 3.2  
  
% Plots the scale as a spectrogram  
spectrogram(xx,512,fs)  
title('Play_scale spectrogram reading');  
xlabel('Frequency (Hz)');  
ylabel('Time (s)');  
  
% When zoomed in you can see that the 440 Hz note is played from  
% time 1.25s to 1.5s, meaning that is it key 49 or A-440.
```



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