

EHB328E – Machine Learning for Signal Processing Matlab HW1

Project Members

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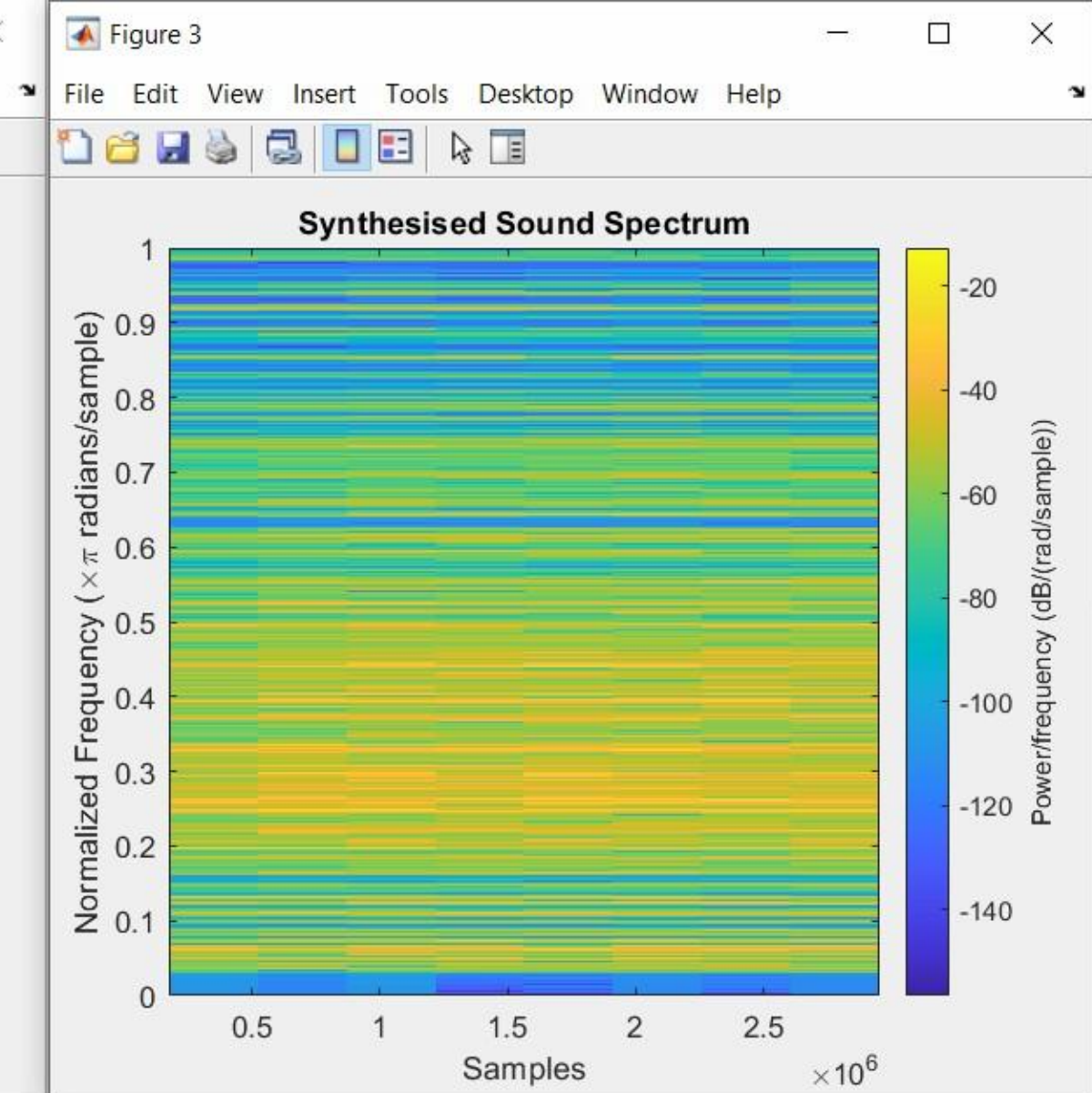
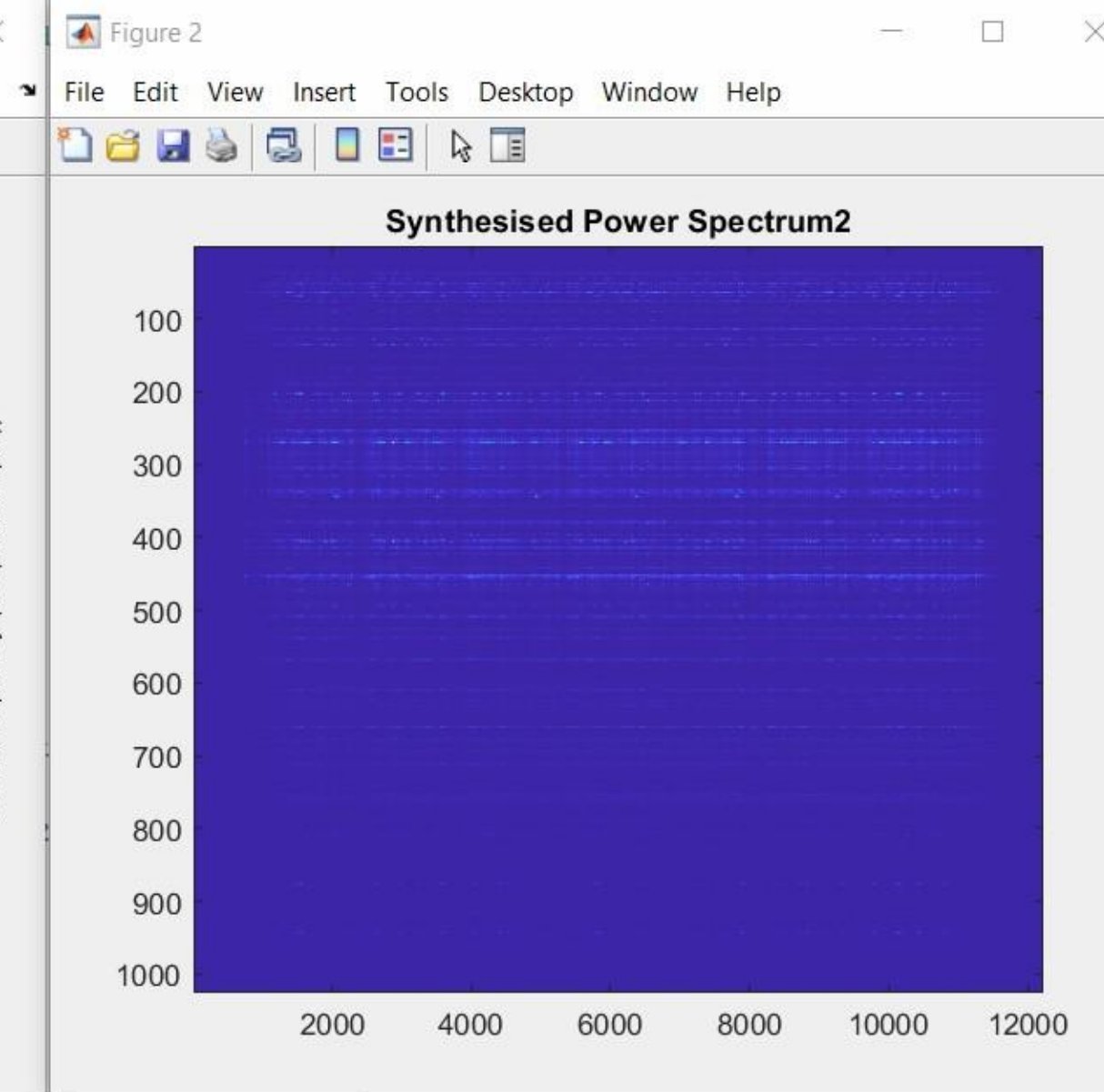
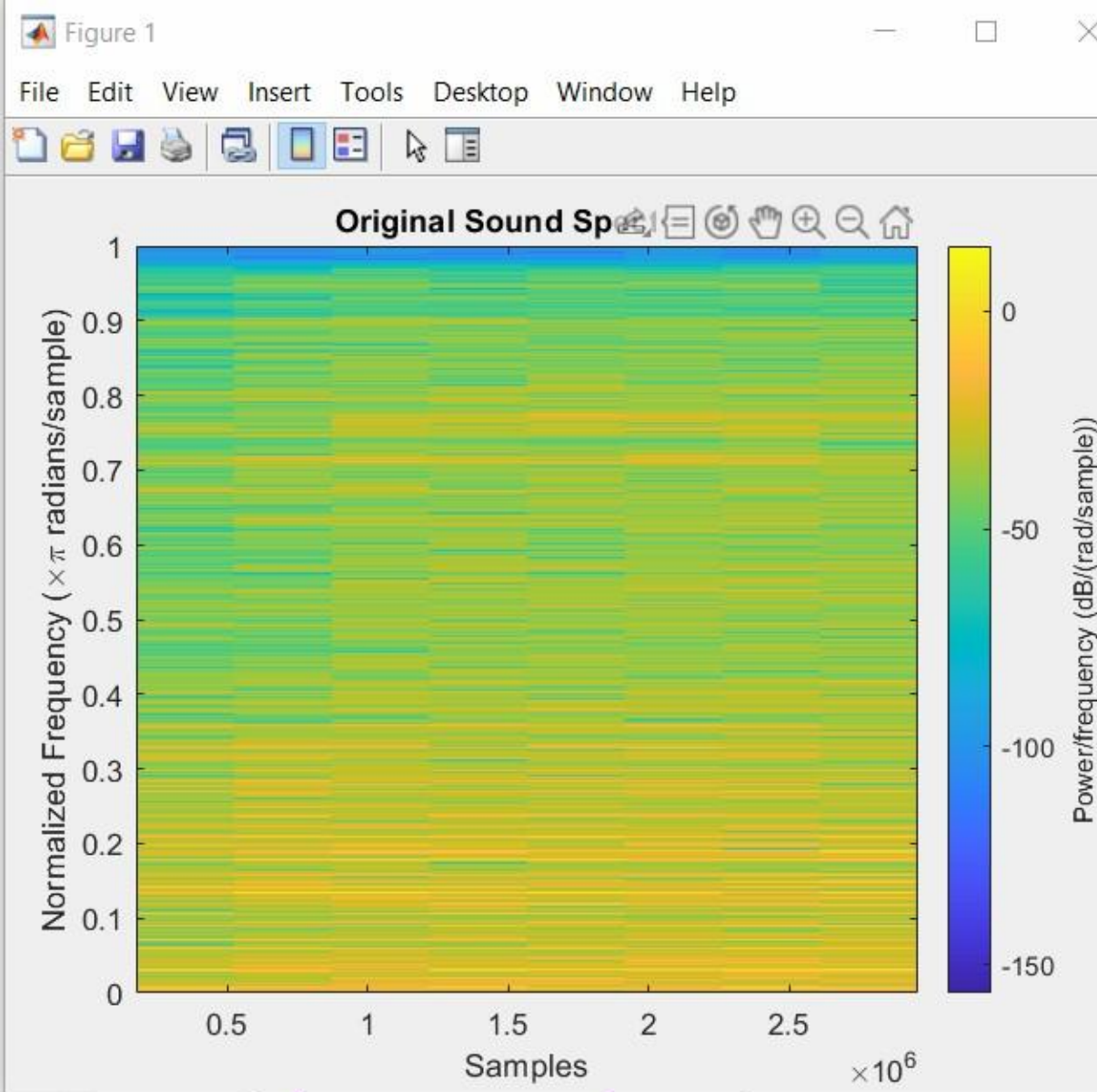
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```
1 - clear, clc, close all;
2
3 - % Firstly read the audio.
4 - materialfolder = 'hw1materials';
5 - soundname = dir([materialfolder filesep 'polyushka.wav']);
6 - [poly, Fs] = audioread([materialfolder filesep soundname.name]);
7
8 - % Figure the original spectrum of audio.
9 - figure
10 - spectrogram(poly, 'yaxis')
11 - title('Original Sound Spectrum')
12
13 - notesfolder = 'notes15';
14 - portion = 1:Fs*5;
15
16 - poly_spectrum = stft(poly', 2048, 256, 0, hann(2048));
17 - poly_stft = abs(poly_spectrum);
18 - poly_phase = poly_spectrum./(poly_stft+eps);
19
20 - % Adding the 15 notes in to 'notes' array.
21 - listname = dir([materialfolder filesep notesfolder filesep '*.wav']);
22 - notes = [];
23 - for i=1:length(listname)
24 -     [s, Fs] = audioread([materialfolder filesep notesfolder filesep listname(i).name]);
25 -     s = s(:,1);
26 -     s = resample(s, 16000, Fs);
27 -     spectrum = stft(s', 2048, 256, 0, hann(2048));
28 -     middle = ceil(size(spectrum, 2) /2);
29 -     note = abs(spectrum(:, middle));
30 -     note(find(note<max(note(:))/100)) = 0 ;
31 -     note = note/norm(note);
32 -     notes = [notes, note];
33 - end
```

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34
35 % To find the W, pseudoinverse of notes * stft of the audio poly.
36 W = pinv(notes)*poly_stft;
37
38 % W's negative values go to zero.
39 for i=1:numel(W)
40     if W(i)<0
41         W(i) = 0;
42     end
43 end
44
45 polysynth_stft = notes*W;
46
47 % Figure of synthesised power spectrum of audio's stft.
48 figure
49 imagesc(polysynth_stft)
50 title('Synthesised Power Spectrum2')
51
52 % For figure of synthesised sound spectrum, invert the stft of audio.
53 polysynth_spectrum = polysynth_stft.*poly_phase;
54 polysynth = stft(polysynth_spectrum, 2048, 256, 0, hann(2048));
55 polysynth = polysynth';
56
57 % Figure the synthesised sound spectrum.
58 figure
59 spectrogram(polysynth, 'yaxis')
60 title('Synthesised Sound Spectrum')
61
62 % Obtain the synthesised audio in format of wav.
63 audiowrite('poly_synth.wav', polysynth, Fs);
64
65 function [f,fp] = stft( x, sz, hp, pd, w) ...

```




```
1 - clear all; clc; close all;
2
3 % Firstly read the 3 audios.
4 - [audioA, fsA] = audioread('silentnight_piano.aif');
5 - [audioB, fsB] = audioread('silentnight_guitar.aif');
6 - [audioC, fsC] = audioread('littlestar_piano.aif');
7
8 % Given audios have two channels, we need first channel.
9 % Then find audios' stft.
10 - audioAL = audioA(:,1);
11 - spectrumA = stft(audioAL', 1024, 256, 0, hann(1024));
12 - music_stftA = abs(spectrumA);
13
14 - audioBL = audioB(:,1);
15 - spectrumB = stft(audioBL', 1024, 256, 0, hann(1024));
16 - music_stftB = abs(spectrumB);
17
18 - audioCL = audioC(:,1);
19 - spectrumC = stft(audioCL', 1024, 256, 0, hann(1024));
20 - music_stftC = abs(spectrumC);
21 - sphaseC = spectrumC ./ (abs(spectrumC)+eps);
22
23 % Before transformation the audio's negative values go to zero.
24 - for i=1:numel(music_stftC)
25 -     if music_stftC(i) < 0
26 -         music_stftC = 0;
27 -     end
28 - end
29 |
30 % To find the audio D's stft, get these following steps.
31 - X = music_stftB * pinv(music_stftA);
32 - stftD = X * music_stftC;
```

```
33 % Finally invert the audio D's stft.
34 - audioDL = stft(stftD.*sphaseC, 1024, 256, 0, hann(1024));
35 - audioDL = audioDL';
36 - soundsc(audioDL(portion), fsC);
37
38 % Obtain audio D which means 'littlestar_guitar' from the others in format wav.
39 - audiowrite('audioD_syms.wav', audioDL, fsC);
40
41 % Matlab file in the attached zip combines the two codes. Please inspect that one as well.
42 % These codes can be found at https://github.com/velibulur/ehb328hw2 and https://github.com/meserbetcioğlu/ehb328hw2
43 % Written by Veli Bulur and Mehmet Şerbetçioğlu
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