# Music Classification Using Linear Discriminant Analysis

#### Christina Smith

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## Abstract

In this exercise, we create algorithms to classify audio clips. We break a data set into training data and test data and use principal component analysis and linear discriminant analysis to create and evaluate a model to classify new audio clips.

#### I. Introduction

Our goal is to classify data in three different tests:

- Test 1: Classify three different bands of different genres.

  (Rubblebucket, Nina Simone, Air)
- Test 2: Classify three different bands from the same genre.

  (Al Green, Otis Redding, Sam Cooke)
- Test 1: Classify three different genres.

  (Chill Beats, Country, Rock)

Using singular value decomposition (SVD) and principal component analysis (PCA) we evaluate a small subset of our data (the training set). We chose a value for the number of features (principal components) we would like to evaluate. Next we determine the variance within classes and the covariance for the means of the different classes.

By performing eigenvalue decomposition on these matrices we can determine the bases which give the minimum variance within classes and the maximum covariance for the means of the different classes when we project the data onto it. Since we have three different classes, we can choose a 1 or 2 dimensional basis. If we are evaluating the data on a line, we calculate the thresholds on the line that give us the best separation of the three classes. If we are evaluating in a plane, we compute the mean coordinates of each cluster of data and determine class by finding the minimum distance of a given test sample to the means of the three classes.

# II. Theoretical Background

#### SVD and PCA

Since we want to determine a way to classify data from our three different groups for each test, we will combine our data and use SVD and PCA as way to determine the basis with the most variance, that is, the principal component vectors with the highest corresponding singular values. In this case, the rank of this basis is also described as the number of 'features' we want to evaluate the data for.

When we have chosen the principal component vectors, we project our data onto the new basis.

## Linear Discriminant Analysis

Now that we have a basis with the maximum variance of our complete data set we want to determine another basis to project this data onto such that there is minimal overlap in the data from each class. The way to do this is to determine a basis of projection that maximizes the covariance between each class mean and minimizes the variance within each class (in regards to each dimension of the projection basis).

We first create a matrix of the variance within individual sets of data using the following formula:

$$\mathbf{S}_W = \sum_{j=1}^{N} \sum_{\mathbf{x}} (\mathbf{x} - \boldsymbol{\mu}_j) (\mathbf{x} - \boldsymbol{\mu}_j)^T$$
 (1)

Where N is the number of classes,  $\mu$  is a vector which contains the mean of each row in the submatrix containing the data of a given class and  $\mathbf{x}$  indicates we perform this process for every column in the submatrix.

The matrix of the covariance between the means of the sets of data is computed as follows:

$$\mathbf{S}_B = \sum_{j=1}^{N} (\boldsymbol{\mu}_j - \boldsymbol{\mu})(\boldsymbol{\mu}_j - \boldsymbol{\mu}))^T$$
 (2)

Where  $\mu$  contains the mean across all classes. Using these matrices we compute the eigenvalue decomposition

$$\mathbf{S}_B \mathbf{w} = \lambda \mathbf{S}_W \mathbf{w} \tag{3}$$

Because our data set contains three classes,  $\mathbf{w}$  is a matrix which contains the vectors that represent the basis that maximizes the covariance between each class mean and minimizes the variance within each class.

#### Classification

With the training data projected onto the new basis a model is formed to differentiate the classes. One method is to evaluate the means of the clusters of data on a line and to set threshold values which will predict whether test data belongs to a given class. Another method is to project the training data onto a planar basis of the two vectors of **w** and then evaluate the mean location of each cluster. Test data will be projected on the same plane and evaluated based on its distance from the means of the class data.

# III. Algorithm Implementation and Development

#### Data Acquisition (See App. A: 'get\_data.m' and 'create\_spectrograms.m')

Audio data was processed into wav files each containing 5 seconds of music. For test 1, samples were taken of 4 songs with 10 clips of each song. For tests 2 and 3 samples were taken of 7 songs with 10 clips of each song. This data was then used create spectrograms. The spectrograms had a range in frequency of  $\pm$ 000 Hz in order to capture the most relevant data.

## trainer() (See App. A: 'HW\_04.m' lines 300-490)

This function receives the spectrogram data for the three classes, as well as information about the number of features to evaluate. All of the spectrogram data is combined into a single matrix and then SVD is performed on this matrix. The data is then projected on the principal component basis and the matrix is separated into three separate matrices with the number of rows corresponding to the number of features.

The variance and covariance matrices are computed using Formulas 1 and 2 respectively and SVD is performed using Matlab's eig() function. The eigenvectors corresponding to the eigenvalues with the maximum absolute value are normalized. These unit vectors are the basis for our analysis.

The training data is projected onto these vectors and the means of each class are determined. These means are used to determine where each class falls on the projection basis (maximum, middle value, minimum). One simple way of determining the two thresholds is to find the mean of each class on the projection line the halfway points between the max and middle mean, and between the middle and min mean are used as the threshold values. Another possible method, given that we have three classes, is to project the training data onto the w1 and w2 coordinate plane and determining the coordinates for the minimum of each cluster. When test data is evaluated, these mean coordinates will be used to determine the class of a given test sample.

The function returns the matrices from the SVD of the training data, as well as the **w1**, **w2** vectors, the corresponding eigenvalues, the thresholds, the rank of each class and the projection of the data onto the **w1**, **w2** vectors.

## classify1D() (See App.A: 'HW\_04.m' lines 254-269)

This function takes the projected values (on the  $\mathbf{w1}$  line) of the test data, the threshold values for  $\mathbf{w1}$ , and the rank of each class. It then determines where each value falls in relation to the thresholds and then assigns a class based on the rank information. It returns a vector of classes corresponding to the projected value of each sample.

## classify2D() (See App. A: 'HW\_04.m' lines 276-297)

This function takes the projected values (on the **w1**, **w2** plane) of the test data, the mean coordinate values of the training data. It then determines which mean is the closest for every sample, and assigns a class based on the mean with the minimum distance. It returns a vector of classes corresponding to the projected value of each sample.

# IV. Computational Results & Supplementary Plots

#### Test 1 - Three different bands of different genres (See Fig. 1)

With features set to 50, a training set of 25 samples and a test set of 15 samples, the prediction accuracies using both 1D and 2D classification were as follows:

	Class A	Class B	Class C
1D	86%	20%	53%
2D	100%	20%	73%

Class B was the lowest range of values on the **w1** projection, Class A was the middle range, and Class C was the upper range. The threshold values were -4.1 and 5.3.

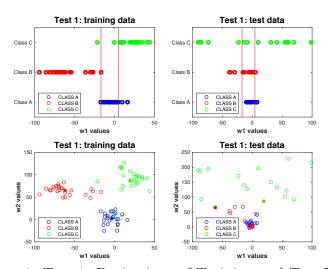


Figure 1: Test 1: Projections of Training and Test Data

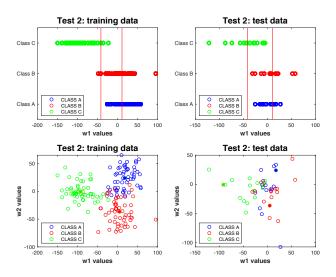


Figure 2: Test 2: Projections of Training and Test Data

## Test 2 - Three different bands from the same genre (See Fig. 2)

With features set to 80, a training set of 55 samples and a test set of 15 samples, the prediction accuracies using both 1D and 2D classification were as follows:

	Class A	Class B	Class C
1D	26.7%	60%	46.7%
2D	46.7%	66.7%	40%

Class C was the lowest range of values on the  $\mathbf{w1}$  projection, Class B was the middle range, and Class A was the upper range. The threshold values were -41.8 and 10.9.

## Test 3 - Three different genres. (See Fig. 3)

With features set to 80, a training set of 55 samples and a test set of 15 samples, the prediction accuracies using both 1D and 2D classification were as follows:

	Class A	Class B	Class C
1D	73.3%	20%	40%
2D	66.7%	33.3%	46.7%

Class C was the lowest range of values on the **w1** projection, Class B was the middle range, and Class A was the upper range. The threshold values were -41.8 and 10.9.

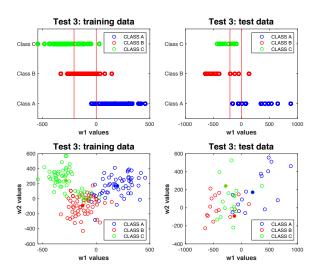


Figure 3: Test 3: Projections of Training and Test Data

# V.Summary and Conclusions

From reviewing the accuracy percentages it appears that evaluating the data using both projection vectors *generally* leads to more accurate classification. It is worth noting that in Tests 2 and 3, the accuracy of one class did go down with prediction in 2 dimensions, however the accuracy gained in predicting each of the other two classes was greater than the accuracy lost by the one class.

The classifications were most accurate on average in the first test. This follows what I was expecting, because the samples all differed in both artist and genre. The second and third tests had lower average accuracy. This made sense, since these data sets had more sound in common that in Test 1.

Changing the parameters of the feature number and the size of the training and sample sets also had an effect. For example, in Test 3, changing the number of features from 80 to 5 resulted in a 0% accuracy rate for Class B (in 2D). A smaller number of features resulted in a larger variance in the data and a larger number of features had the opposite effect. Changing the size of the training data set from 55 to 10, decreased the accuracy in predicting Class A by almost half. These also affected the spread of the data on the projected line or plane.

The classification model itself can also be optimized. While the methods used here are fairly simple, one possibility would be to evaluate the data in 2D giving more weight to the relationship between means along the **w1** axis, as that is the basis that has best minimized the in-class variance and maximized the between class variance. There is much more to explore in this type of modelling, as well as how the parameters and the data sets themselves affect the accuracy of classification.

# Appendix A: Matlab Code

## $get_data.m$

```
1 % %% Test 3 c
2 % close all; clear all; clc;
3 %
4 \% s = (50000) *2;
5 %
6 % dat_3c = zeros(s,70); % 40 samples total (10/song)
7
8 \% count = 0;
9 % for i=1:7
10 %
         % go through each song
          filename = strcat('test_03/rock/all_songs/0', num2str(i), '.
11
   %
       mp3');
12
  %
          [y, Fs] = audioread(filename);
13 %
          y_{-}mono = y(:,1); % mono
14 %
          b = 1:2: length(y_mono); \% resumpling vector
15 %
          y_resumpled = y_mono(b); % transpose the y vector
16 %
  %
         % get 10 samples of each song
17
   %
18
         for j = 1:10
19
   %
              dat_3c(:,count+1) = y_resumpled(1:s);
                                                        % new 5 second
       clip
20
   %
              last = size(y_resampled);
21 %
              y_resampled = y_resampled(s+1:last(2));
22 %
              if \ count > 9
23
  %
                  wavname = \dots
24
   %
                       strcat('/Users/christinasmith/Desktop/AMATH_482.
       nosync/Homework/HW_04/test_03/rock/all_clips/0,...
   %
25
                  num2str(count), '. wav');
  %
                  audiowrite(wavname, dat_3c(:, count+1), Fs/2);
26
27
   %
              else
  %
28
                  wavname = \dots
29
   %
                      strcat('/Users/christinasmith/Desktop/AMATH\_482.
       nosync/Homework/HW_04/test_03/rock/all_clips/00,...
   %
30
                  num2str(count), '.wav');
  %
31
                  audiowrite(wavname, dat_3c(:, count+1), Fs/2);
32 %
              end
33 %
              count = count + 1;
          end
34
  %
35 \% end
36 %
37
   \% save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
       HW_-04/test_-03/test_-03c.mat\ dat_-3c
38
39 % %% Test 3 b
40 % close all; clear all; clc;
41 %
42 \% s = (50000) * 2;
43 %
44 \% dat_3b = zeros(s,70); \% 70 samples total (10/song)
45 %
```

```
46 \% count = 0;
47 % for i = 1:7
  %
          % go through each song
48
49
   %
          filename = strcat('test_03/country/all_songs/0', num2str(i))
        , '. mp3');
   %
          [y, Fs] = audioread(filename);
50
51
   %
          y_{-}mono = y(:,1); % mono
52
   %
          b = 1:2: length(y_mono); \% resampling vector
   %
53
          y_resumpled = y_mono(b); % transpose the y vector
54
   %
55
   %
          % get 10 samples of each song
   %
56
          for j = 1:10
57
   %
               dat_3b(:,count+1) = y_resampled(1:s);
                                                          % new 5 second
       clip
58
   %
               last = size(y_resampled);
   %
59
               y_resampled = y_resampled(s+1:last(2));
   %
               if count > 9
60
61
   %
                   wavname = \dots
62
   %
                       strcat('/Users/christinasmith/Desktop/AMATH\_482.
       nosync/Homework/HW_-04/test_-03/country/all_-clips/0,...
   %
63
                   num2str(count), '. wav');
   %
64
                   audiowrite(wavname, dat_3b(:, count+1), Fs/2);
   %
65
               else
66
   %
67
  %
                       strcat('/Users/christinasmith/Desktop/AMATH_482.
       nosync/Homework/HW_-04/test_-03/country/all_clips/00,...
                   num2str(count), '.wav');
68
   %
   %
69
                   audiowrite(wavname, dat_3b(:, count+1), Fs/2);
70
   %
               end
71
   %
               count = count + 1;
72
   %
          end
73
   % end
74
75
   % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
       HW_-04/test_-03/test_-03b. mat_-dat_-3b
76
   % %
   % %% Test 3 a
77
78 % close all; clear all; clc;
79 %
80 \% s = (50000) *2;
81
  \%
82 % dat_3a = zeros(s,40); % 40 samples total (10/song)
83 %
84 \% count = 0;
85
   \% \ for \ i=1:7
   %
86
          % go through each song
          filename = strcat('test_03/chill_beats/all_songs/0', num2str('test_03/chill_beats/all_songs/0')
87
   %
       i), '.mp3');
88
   %
          [y, Fs] = audioread(filename);
  %
89
          y_{-}mono = y(:,1); % mono
  %
          b = 1:2:length(y_mono); \% resumpling vector
90
   %
          y_resumpled = y_mono(b); % transpose the y vector
91
92 %
93 %
          % get 10 samples of each song
```

```
94
   %
           for j = 1:10
95 %
               dat_{-}3a(:,count+1) = y_{-}resampled(1:s);
                                                         % new 5 second
        clip
    %
 96
               last = size(y_resampled);
    %
 97
               y_resampled = y_resampled(s+1:last(2));
98
   %
               if \ count > 9
99
    %
                   wavname = \dots
100
    %
                        strcat('/Users/christinasmith/Desktop/AMATH_482.
        nosync/Homework/HW\_04/test\_03/chill\_beats/all\_clips/0,...
    %
101
                   num2str(count), '.wav');
102
    %
                    audiowrite(wavname, dat_3a(:, count+1), Fs/2);
    %
103
               else
104
    %
                   wavname = \dots
105
    %
                        strcat('/Users/christinasmith/Desktop/AMATH_482.
        nosync/Homework/HW_04/test_03/chill_beats/all_clips/00',...
106
    %
                   num2str(count), '.wav');
107
    %
                    audiowrite(wavname, dat_3a(:, count+1), Fs/2);
108
    %
               end
109
    %
               count = count + 1;
    %
110
           end
    % end
111
    %
112
    \% save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
113
        HW_04/test_03/test_03a.mat\ dat_3a
114
    %
115
    %
    % %% Test 2 c
116
117
    % close all; clear all; clc;
118
119
    %
120 \% s = (50000) * 2;
121 %
122 % dat_2c = zeros(s,70); % 70 samples total (10/song)
123
    %
124 \% count = 0;
125
    \% for i = 1:7
126
    %
           % go through each song
    %
           filename = strcat('test_02/sam_cooke/all_songs/0', num2str(i))
127
        , '. wav');
128
    %
           [y, Fs] = audioread(filename);
129
    %
           y_{-}mono = y(:,1); % mono
130 %
           b = 1:2: length(y_mono); \% resumpling vector
   %
           y_resumpled = y_mono(b); % transpose the y vector
131
132 \%
133
    %
           % get 10 samples of each song
    %
134
           for j = 1:10
   %
135
               dat_2c(:,count+1) = y_resampled(1:s);
                                                          % new 5 second
        clip
136 %
               last = size(y_resampled);
    %
137
               y_resampled = y_resampled(s+1:last(2));
138
    %
               if count > 9
139
    %
                   wavname = \dots
140 %
                        strcat('/Users/christinasmith/Desktop/AMATH\_482.
        nosync/Homework/HW_-04/test_-02/sam_-cooke/all_clips/0,...
```

```
141 %
                    num2str(count), '. wav');
142 %
                    audiowrite(wavname, dat_2c(:, count+1), Fs/2);
143 %
               else
144 %
                    wavname = \dots
   %
145
                        strcat('/Users/christinasmith/Desktop/AMATH_482.
        nosync/Homework/HW\_04/test\_02/sam\_cooke/all\_clips/00,...
    %
                    num2str(count), '.wav');
146
147
    %
                    audiowrite(wavname, dat_2c(:, count+1), Fs/2);
    %
148
               end
149
    %
               count = count + 1;
    %
150
           end
    % end
151
152
    %
153
    %
154
    % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
        HW_04/test_02/test_02c.mat\ dat_2c
155
    %
156
    % Test 2 b
157
158
    % close all; clear all; clc;
159
    %
160
   % s = (50000) *2;
161 %
162 \% dat_{-}2b = zeros(s,70); \% 70 samples total (10/song)
163 %
164
    % count = 0;
   % for i = 1:7
165
166
   %
           % go through each song
167
    %
           filename = strcat('test_02/otis_redding/all_songs/0', num2str
        (i), '. wav');
168 %
           [y, Fs] = audioread(filename);
   %
169
           y_{-}mono = y(:,1); % mono
           b = 1:2: length(y_mono); \% resampling vector
170 %
171
    %
           y_resumpled = y_mono(b); % transpose the y vector
172 %
           % get 10 samples of each song
173
    %
    %
174
           for j = 1:10
    %
               dat_{-}2b(:, count+1) = y_{-}resampled(1:s);
175
                                                          % new 5 second
        clip
176
    %
               last = size(y_resampled);
177
    %
               y_resampled = y_resampled(s+1:last(2));
178
    %
               if \ count > 9
    %
179
                    wavname = \dots
180
    %
                        strcat('/Users/christinasmith/Desktop/AMATH_482.
        nosync/Homework/HW_-04/test_-02/otis_redding/all_clips/0',...
   %
181
                    num2str(count), '. wav');
   %
                    audiowrite(wavname, dat_2b(:, count+1), Fs/2);
182
183
    %
               else
184
    %
                    wavname = \dots
    %
185
                        strcat('/Users/christinasmith/Desktop/AMATH_482.
        nosync/Homework/HW_04/test_02/otis_redding/all_clips/00',...
    %
                    num2str(count), '. wav');
186
187
    %
                    audiowrite(wavname, dat_2b(:, count+1), Fs/2);
188 %
               end
```

```
189 %
               count = count + 1;
190 %
           end
191
    % end
    %
192
    \% save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
193
        HW_04/test_02/test_02b. mat dat_2b
194
195
    % %% Test 2 a
196
197 %
198 % close all; clear all; clc;
199 %
200
   % s = (50000) *2;
201
   %
202 % dat_{-}2a = zeros(s, 40); % 40 samples total (10/song)
203 %
204 \% count = 0;
205 % for i = 1:7
206 %
          % go through each song
    %
           filename = strcat('test_02/al_green/all_songs/0', num2str(i))
207
        , '. wav ');
           [y, Fs] = audioread(filename);
208
    %
    %
209
           y_{-}mono = y(:,1); % mono
210
   %
           b = 1:2: length(y_mono); \% resumpling vector
211 %
           y_resumpled = y_mono(b); % transpose the y vector
212
    \%
213 %
          % get 10 samples of each song
   %
214
           for i = 1:10
               dat_{-}2a(:,count+1) = y_{-}resampled(1:s);
215
    %
                                                          % new 5 second
        clip
216 %
               last = size(y_resampled);
    %
217
               y_resampled = y_resampled(s+1:last(2));
    %
218
               if count > 9
219
    %
                   wavname = \dots
220
    %
                        strcat(')/Users/christinasmith/Desktop/AMATH\_482.
        nosync/Homework/HW_04/test_02/al_green/all_clips/0,...
221
    %
                   num2str(count), '.wav');
222
    %
                    audiowrite(wavname, dat_2a(:, count+1), Fs/2);
   %
223
               else
224
   %
                   wavname = \dots
    %
225
                        strcat('/Users/christinasmith/Desktop/AMATH\_482.
        nosync/Homework/HW\_04/test\_02/al\_green/all\_clips/00',...
226
    %
                   num2str(count), '. wav');
227
    %
                    audiowrite(wavname, dat_2a(:, count+1), Fs/2);
228
    %
               end
229
    %
               count = count + 1;
    %
           end
230
231
    % end
232
    %
    %
233
    % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
234
        HW_04/test_02/test_02a. mat dat_2a
235
    %
236 %
```

```
237 % %% Test 1 c
238 %
239 % close all; clear all; clc;
240 %
   % s = (50000) *2;
241
242 \%
243
    \% dat_1c = zeros(s,40); \% 40 samples total (10/song)
244 %
245 \% count = 0;
246 % for i = 1:4
247
           % go through each song
    %
           filename = strcat('test_01/air/all_songs/0', num2str(i), '.wav
248
        <sup>,</sup>);
249
    %
           [y, Fs] = audioread(filename);
250
   %
           y_{-}mono = y(:,1); % mono
251 %
           b = 1:2:length(y_mono); \% resumpling vector
252 %
           y_resumpled = y_mono(b); % transpose the y vector
253 \%
254
    %
           % get 10 samples of each song
   %
255
           for j = 1:10
    %
               dat_{-}1c(:,count+1) = y_{-}resampled(1:s);
256
                                                          % new 5 second
        clip
257
   %
               last = size(y_resampled);
258
   %
               y_resampled = y_resampled(s+1:last(2));
259 %
               if count > 9
260 %
                    wavname = \dots
                        strcat (\ '/\ Users/christinasmith/Desktop/AMATH\_482.
    \%
261
        nosync/Homework/HW_04/test_01/air/all_clips/0,...
262
    %
                    num2str(count), '. wav');
263
    %
                    audiowrite(wavname, dat_1c(:, count+1), Fs/2);
264
    %
               else
265
   %
                    wavname = \dots
    %
266
                        strcat('/Users/christinasmith/Desktop/AMATH_482.
        nosync/Homework/HW_-04/test_-01/air/all_-clips/00',...
267
    %
                    num2str(count), '. wav');
268
    %
                    audiowrite(wavname, dat_1c(:, count+1), Fs/2);
    %
269
               end
270
    %
               count = count + 1;
    %
271
           end
272
    % end
273
    %
274
    \% save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
        HW_04/test_01/test_01c.mat\ dat_1c
275
    %
276
    % %% Test 1 b
277
    % close all; clear all; clc;
278 %
279 \% s = (50000) * 2;
280 %
281 % dat_{-}1b = zeros(s, 40); % 40 samples total (10/song)
282 %
283 \% count = 0;
284 % for i = 1:4
285 \%
           % go through each song
```

```
286 %
           filename = strcat('test_01/nina/all_songs/0', num2str(i), '.
        wav');
287
    %
           [y, Fs] = audioread(filename);
288 %
           y_{-}mono = y(:,1); % mono
289 %
           b = 1:2: length(y_mono); \% resumpling vector
290 %
           y_resumpled = y_mono(b); % transpose the y vector
    %
291
    %
292
          % get 10 samples of each song
   %
293
          for j = 1:10
294
    %
               dat_{-}1b(:,count+1) = y_{-}resampled(1:s);
                                                         % new 5 second
        clip
295
    %
               last = size(y_resampled);
296
   %
               y_resampled = y_resampled(s+1:last(2));
297
    %
               if count > 9
298
    %
                   wavname = \dots
299 %
                       strcat('/Users/christinasmith/Desktop/AMATH_482.
        nosync/Homework/HW_04/test_01/nina/all_clips/0',...
300
   %
                   num2str(count), '. wav');
301
   %
                   audiowrite(wavname, dat_1b(:, count+1), Fs/2);
302 %
               else
303 %
                   wavname = \dots
    %
                        strcat('/Users/christinasmith/Desktop/AMATH_482.
304
        nosync/Homework/HW_04/test_01/nina/all_clips/00',...
305
    %
                   num2str(count), '. wav');
306
   %
                   audiowrite(wavname, dat_1b(:, count+1), Fs/2);
    %
307
               end
308 %
               count = count + 1;
   %
309
310
   %
           end
311
    % end
312 %
313 %
    % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
314
        HW_04/test_01/test_01b. mat dat_1b
    %
315
    % %% Test 1 a
316
    %
317
318 %
319 % close all; clear all; clc;
320 \%
321 \% s = (50000) *2;
322 \%
323 \%
324 \% dat_1a = zeros(s,40); \% 40 samples total (10/song)
325
    %
326 \% count = 0;
327 \% for i=1:3
328 %
          % Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
        HW_04/test_01/air/all_songs/01.wav
329
    %
          \% go through each song
    %
           filename = strcat('test_01/rubblebucket/all_songs/0', num2str
330
        (i), '. wav');
           [y, Fs] = audioread(filename);
331 %
332 %
           y_{-}mono = y(:,1); % mono
```

```
333
   %
           b = 1:2:length(y_mono); \% resampling vector
334 %
           y_resumpled = y_mono(b); % transpose the y vector
335 %
   %
           % get 10 samples of each song
336
    %
           for j = 1:10
337
    %
338
               dat_1a(:, count+1) = y_resumpled(1:s);
                                                          % new 5 second
        clip
339
    %
               last = size(y_resampled);
    %
340
               y_resampled = y_resampled(s+1:last(2));
    %
               if \ count > 9
341
342
    %
                   wavname = \dots
    %
343
                        strcat('/Users/christinasmith/Desktop/AMATH\_482.
        nosync/Homework/HW\_04/test\_01/rubblebucket/all\_clips/0',...
344
    %
                    num2str(count), '.wav');
345
    %
                    audiowrite(wavname, dat_1a(:, count+1), Fs/2);
    %
346
               else
347
    %
                    wavname = \dots
348
    %
                        strcat('/Users/christinasmith/Desktop/AMATH_482.
        nosync/Homework/HW\_04/test\_01/rubblebucket/all\_clips/00',...
    %
349
                    num2str(count), '. wav');
    %
                    audiowrite(wavname, dat_1a(:, count+1), Fs/2);
350
    %
351
               end
    %
               count = count + 1;
352
353
    %
           end
354
    % end
    %
355
    %
356
357
    \% filename = strcat('test_01/rubblebucket/all_songs/la.wav');
    % [y, Fs] = audioread(filename);
358
359
    \% y_{-}mono = y(:,1); \% mono
360
    \% b = 1:2:length(y_mono); \% resampling vector
361
    \% y_resumpled = y_mono(b); \% transpose the y vector
   %
362
    \% \ for \ j=1:9
363
   %
364
           dat_{-}1a(:,count+1) = y_{-}resampled(1:s);
                                                      % new 5 second clip
           last = size(y_resampled);
365
   %
    %
366
           y_resampled = y_resampled(s+1:last(2));
    %
367
           if \ count > 9
   %
368
               wavname = \dots
369
    %
                    strcat(')/Users/christinasmith/Desktop/AMATH\_482.
        nosync/Homework/HW_04/test_01/rubblebucket/all_clips/0',...
370
    %
               num2str(count), '.wav');
    %
371
               audiowrite(wavname, dat_1a(:, count+1), Fs/2);
372
    %
           else
373
    %
   %
374
                    strcat(')/Users/christinasmith/Desktop/AMATH\_482.
        nosync/Homework/HW\_04/test\_01/rubblebucket/all\_clips/00',...
375
    %
               num2str(count), '.wav');
376
    %
               audiowrite(wavname, dat_1a(:, count+1), Fs/2);
    %
377
           end
378
   %
           count = count + 1;
    % end
379
380 %
381 %
```

382 % save /Users/christinasmith/Desktop/AMATH\_482.nosync/Homework/  $HW_-04/test_-01/test_-01a.mat\ dat_-1a$ 

#### create\_spectrograms.m

```
%% Test 3 c
 1
 3
    close all; clear all; clc;
 4
   load test_03/test_03c.mat
 5
 6
 7
   % set the time and frequency domains
 8
 9
10 \operatorname{\mathbf{size}} = \operatorname{\mathbf{size}}(\operatorname{dat}_{-3}\operatorname{c});
11 n=size(2);
                    % number of samples
12 s = size(1); % length of samples
13 L=s/(44100/2); % time domain
14 t2 = linspace(0, L, s+1);
15 t=t2(1:s);
16 k_Hz = (1/(2*L))*[0:(s/2-1) -s/2:-1];
17 ks_Hz = \mathbf{fftshift}(k_Hz);
18 ks_Hz = ks_Hz((s/2-4000):(s/2+4000));
19
20 \quad a = 2e4;
21 tstep = 0:0.3:L;
22
23
24 % array of spectrogram data
25 \text{ sp-3c} = \mathbf{zeros}(8001*\mathbf{length}(tstep),n);
26 spect_array = zeros(8001, length(tstep), n);
27
28
   for j=1:n
29
         % spectrogram of clip
30
          spect = zeros(s, length(tstep));
31
          for k=1:length(tstep)
32
               \mathbf{g} = \mathbf{exp}(-\mathbf{a} * (\mathbf{t} - \mathbf{t} \mathbf{s} \mathbf{t} \mathbf{e} \mathbf{p} (\mathbf{k})) . ^2);
33
               Sg=g'.*dat_3c(:,j); % apply new filter to the current 5sec
34
               Sgt=fft (Sg); % transform to frequency domain
35
               spect(:,k) = fftshift(abs(Sgt));
36
         end
37
          spect = spect((s/2-4000):(s/2+4000),:);
38
          spect_array(:,:,j) = spect;
          \operatorname{sp_3c}(:,j) = \operatorname{\mathbf{reshape}}(\operatorname{spect},[8001*\operatorname{\mathbf{length}}(\operatorname{tstep}),1]);
39
40 end
41
42 figure (1)
43 subplot (1,1,1)
44 pcolor(tstep , ks_Hz , spect_array(:,:,20)),
45 shading interp
46 title ('Home-made a = 1e03, inc = 0.0446', 'Fontsize', 16)
47 set(gca, 'Fontsize', 11)
48 colormap(hot)
```

```
49
    % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
        HW_{-}04/test_{-}03/test_{-}03c_{-}sp.matsp_{-}3c
51
    % Test 3 b
52
53
54
    close all; clear all; clc;
55
    load test_03/test_03b.mat
56
57
58
59 % set the time and frequency domains
60
61 \quad \mathbf{size} = \mathbf{size} (dat_3b);
62 n=size(2); % number of samples
63 s = size(1); % length of samples
64 L=s/(44100/2); % time domain
65 t2 = linspace(0, L, s+1);
66 \quad t=t2(1:s);
67 k_{-}Hz = (1/(2*L))*[0:(s/2-1) -s/2:-1];
68 ks_Hz=fftshift(k_Hz);
69 ks_Hz = ks_Hz ((s/2-4000):(s/2+4000));
70
71 a = 2e4;
72 tstep = 0:0.3:L;
73
74
75 % array of spectrogram data
76 \text{ sp-3b} = \mathbf{zeros}(8001*\mathbf{length}(tstep),n);
77
    spect_array = zeros(8001, length(tstep), n);
78
79
    for j=1:n
80
         % spectrogram of clip
81
         spect = zeros(s, length(tstep));
82
         for k=1:length(tstep)
83
              \mathbf{g} = \mathbf{exp}(-\mathbf{a} * (\mathbf{t} - \mathbf{tstep}(\mathbf{k})) . ^2);
              Sg=g'.*dat_3b(:,j); \% apply new filter to the current 5sec
84
              Sgt=fft (Sg); % transform to frequency domain
85
86
              spect(:,k) = fftshift(abs(Sgt));
87
         end
88
         spect = spect((s/2-4000):(s/2+4000),:);
89
         \operatorname{spect\_array}(:,:,j) = \operatorname{spect};
         sp_3b(:,j) = reshape(spect,[8001*length(tstep),1]);
90
91
    end
92
93
94 figure (1)
    pcolor(tstep , ks_Hz , spect_array(:,:,20)),
96 shading interp
97 title ('Unfiltered: a = 1e03, inc = 0.0446', 'Fontsize', 16)
98 set (gca, 'Fontsize', 11)
99
    colormap(hot)
100
```

```
101
    \% save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
         HW_{-}04/test_{-}03/test_{-}03b_{-}sp.matsp_{-}3b
103
104
    % Test 3 a
105
106
107
    close all; clear all; clc;
108
109 load test_03/test_03a.mat
110
111
112 % set the time and frequency domains
113
114 \operatorname{\mathbf{size}} = \operatorname{\mathbf{size}}(\operatorname{dat}_{-}3a);
                   % number of samples
115 n=size(2);
116 s = size(1); % length of samples
117 L=s/(44100/2); % time domain
118 t2 = linspace(0, L, s+1);
119 t=t2(1:s);
120 k_Hz= (1/(2*L))*[0:(s/2-1)-s/2:-1];
121 ks_Hz = \mathbf{fftshift} (k_Hz);
122 ks_Hz = ks_Hz((s/2-4000):(s/2+4000));
123
124 \quad a = 2e4;
125 tstep = 0:0.3:L;
126
127
128 % array of spectrogram data
     sp_3a = zeros(8001*length(tstep),n);
130 spect_array = zeros(8001, length(tstep),n);
131
132 for j=1:n
133
         % spectrogram of clip
134
          spect = zeros(s, length(tstep));
          for k=1:length(tstep)
135
              g=exp(-a*(t-tstep(k)).^2);
136
137
              Sg=g'.*dat_3a(:,j); % apply new filter to the current 5sec
138
              Sgt=fft (Sg); % transform to frequency domain
              spect(:,k) = fftshift(abs(Sgt));
139
140
         end
          spect = spect((s/2-4000):(s/2+4000),:);
141
142
          spect_array(:,:,j) = spect;
143
          \operatorname{sp-3a}(:,j) = \operatorname{reshape}(\operatorname{spect},[8001*\operatorname{length}(\operatorname{tstep}),1]);
144
    end
145
146
147
     figure (1)
148 pcolor(tstep, ks_Hz, spect_array(:,:,60)),
149 shading interp
150 title ('Unfiltered: a = 1e03, inc = 0.0446', 'Fontsize', 16)
151
     set (gca, 'Fontsize', 11)
152 colormap(hot)
```

```
153
    \% save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
154
         HW_{-}04/test_{-}03/test_{-}03a_{-}sp.matsp_{-}3a
155
156
    % Test 2 c
157
158
159
    close all; clear all; clc;
160
161 load test_02/test_02c.mat
162
163
164 % set the time and frequency domains
165
166 \quad size = size(dat_2c);
                   % number of samples
167 n=size(2);
168 s = size(1); % length of samples
169 L=s/(44100/2); % time domain
170 t2 = linspace(0, L, s+1);
171 t=t2(1:s);
172 k_{\text{Hz}} = (1/(2*L)) * [0:(s/2-1) - s/2:-1];
173 ks_Hz = \mathbf{fftshift} (k_Hz);
174 ks_Hz = ks_Hz((s/2-4000):(s/2+4000));
175
176 \quad a = 2e4;
177
    tstep = 0:0.3:L;
178
179
180 % array of spectrogram data
     sp_2c = zeros(8001*length(tstep),n);
182 spect_array = zeros(8001, length(tstep),n);
183
184
    for j=1:n
185
         % spectrogram of clip
186
          spect = zeros(s, length(tstep));
187
          for k=1:length(tstep)
              g=exp(-a*(t-tstep(k)).^2);
188
189
              Sg=g'.*dat_2c(:,j); % apply new filter to the current 5sec
              Sgt{=}\mathbf{fft}\left(Sg\right);\ \%\ transform\ to\ frequency\ domain
190
              spect(:,k) = fftshift(abs(Sgt));
191
192
         end
          spect = spect((s/2-4000):(s/2+4000),:);
193
194
          spect_array(:,:,j) = spect;
195
          \operatorname{sp-2c}(:,j) = \operatorname{reshape}(\operatorname{spect}, [8001 * \operatorname{length}(\operatorname{tstep}), 1]);
196
    end
197
198
199
     figure (1)
     pcolor(tstep , ks_Hz , spect_array(:,:,20)),
200
201 shading interp
202 title ('Unfiltered: a = 1e03, inc = 0.0446', 'Fontsize', 16)
203 set (gca, 'Fontsize', 11)
204 colormap(hot)
```

```
205
     save /Users/christinasmith/Desktop/AMATH-482.nosync/Homework/HW-04
206
          / \text{test}_02 / \text{test}_02 \text{c}_s \text{p.mat sp}_2 \text{c}
     %% Test 2 b
207
208
209
     close all; clear all; clc;
210
211
     load test_02/test_02b.mat
212
213
214 % set the time and frequency domains
215
216 \operatorname{\mathbf{size}} = \operatorname{\mathbf{size}} (\operatorname{dat}_{-}2b);
217 n=size(2);
                       % number of samples
218 s = size(1); % length of samples
219 L=s/(44100/2); % time domain
220 t2 = linspace(0, L, s+1);
221 \quad t=t2(1:s);
222 k_Hz = (1/(2*L))*[0:(s/2-1) -s/2:-1];
223 ks_Hz = \mathbf{fftshift} (k_Hz);
224 ks_Hz = ks_Hz((s/2-4000):(s/2+4000));
225
226 \quad a = 2e4;
227 tstep = 0:0.3:L;
228
229
230 % array of spectrogram data
231 \operatorname{sp}_2 \operatorname{b} = \operatorname{zeros}(8001 * \operatorname{length}(\operatorname{tstep}), \operatorname{n});
232 spect_array = zeros(8001, length(tstep), n);
233
234
    for j=1:n
235
           % spectrogram of clip
236
           spect = zeros(s,length(tstep));
237
           for k=1:length(tstep)
238
                \mathbf{g} = \mathbf{exp}(-\mathbf{a} * (\mathbf{t} - \mathbf{tstep}(\mathbf{k})) . \hat{2});
                Sg=g'.*dat_2b\left(:\,,\,j\right);\ \%\ apply\ new\ filter\ to\ the\ current\ 5sec
239
                       clip
240
                Sgt=fft(Sg); % transform to frequency domain
241
                spect(:,k) = fftshift(abs(Sgt));
242
           end
243
           spect = spect((s/2-4000):(s/2+4000),:);
244
           spect_array(:,:,j) = spect;
245
           \operatorname{sp}_{-}2b(:,j) = \operatorname{reshape}(\operatorname{spect}, [8001 * \operatorname{length}(\operatorname{tstep}), 1]);
246
     end
247
248
249
     figure (1)
250 \mathbf{pcolor}(tstep, ks\_Hz, spect\_array(:,:,20)),
     shading interp
252 title ('Unfiltered: a = 1e03, inc = 0.0446', 'Fontsize', 16)
     set (gca, 'Fontsize', 11)
254
    colormap (hot)
255
256 % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
```

```
HW_04/test_02/test_02b_sp.matsp_2b
257
258
    % Test 2 a
259
260
    close all; clear all; clc;
261
262
    load test_02/test_02a. mat
263
264
265 % set the time and frequency domains
266
267 \quad \mathbf{size} = \mathbf{size} (dat_2a);
268 n=size(2);
                    % number of samples
269 s = size(1); % length of samples
270 L=s/(44100/2); % time domain
271 t2 = linspace(0, L, s+1);
272 \quad t=t2(1:s);
273 k_Hz= (1/(2*L))*[0:(s/2-1)-s/2:-1];
274 ks_Hz = \mathbf{fftshift}(k_Hz);
275 ks_Hz = ks_Hz ((s/2-4000):(s/2+4000));
276
277 \quad a = 2e4:
278 tstep = 0:0.3:L;
279
280
281 % array of spectrogram data
    sp_2a = zeros(8001*length(tstep),n);
283 spect_array = zeros(8001, length(tstep), n);
284
285
    for j=1:n
286
          % spectrogram of clip
287
          spect = zeros(s, length(tstep));
288
          for k=1:length(tstep)
289
              \mathbf{g} = \mathbf{exp}(-\mathbf{a} * (\mathbf{t} - \mathbf{tstep}(\mathbf{k})) . ^2);
290
              Sg=g'.*dat_2a(:,j); % apply new filter to the current 5sec
291
               Sgt=fft(Sg); % transform to frequency domain
292
               spect(:,k) = fftshift(abs(Sgt));
293
          end
294
          spect = spect((s/2-4000):(s/2+4000),:);
295
          spect_array(:,:,j) = spect;
296
          \operatorname{sp}_{-}2a(:,j) = \operatorname{reshape}(\operatorname{spect}, [8001 * \operatorname{length}(\operatorname{tstep}), 1]);
297
    end
298
299
300
    figure (1)
301 pcolor(tstep, ks_Hz, spect_array(:,:,20)),
302 shading interp
    title ('Unfiltered: a = 1e03, inc = 0.0446', 'Fontsize', 16)
304
    set (gca, 'Fontsize', 11)
305
    colormap (hot)
306
307 % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
         HW_{-}04/test_{-}02/test_{-}02a_{-}sp.matsp_{-}2a
```

```
308
309
310 %% Test 1 c
311
     close all; clear all; clc;
312
313
314 load test_01/test_01c.mat
315
316
317 % set the time and frequency domains
318
319 \operatorname{\mathbf{size}} = \operatorname{\mathbf{size}} (\operatorname{dat}_{-}1c);
320 n=size(2);
                      % number of samples
321 	ext{ s} = \mathbf{size}(1); \% length of samples
322 L=s/(44100/2); % time domain
323 t2 = linspace(0, L, s+1);
324 \quad t=t2(1:s);
325 k_Hz= (1/(2*L))*[0:(s/2-1)-s/2:-1];
326 ks_Hz=fftshift(k_Hz);
327 \text{ ks}_{-}\text{Hz} = \text{ks}_{-}\text{Hz} ((\text{s}/2-4000):(\text{s}/2+4000));
328
329 \quad a = 2e4:
330 tstep = 0:0.3:L;
331
332
333 % array of spectrogram data
334 \text{ sp_1c} = \mathbf{zeros}(8001*\mathbf{length}(tstep),n);
335 spect_array = zeros(8001, length(tstep), n);
336
337
     for j=1:n
338
           \% spectrogram of clip
339
           spect = zeros(s, length(tstep));
           for k=1:length(tstep)
340
341
                \mathbf{g} = \mathbf{exp}(-\mathbf{a} * (\mathbf{t} - \mathbf{tstep}(\mathbf{k})) . ^2);
342
                Sg=g'.*dat_1c(:,j); % apply new filter to the current 5sec
343
                Sgt=fft(Sg); % transform to frequency domain
344
                spect(:,k) = fftshift(abs(Sgt));
345
           end
346
           spect = spect((s/2-4000):(s/2+4000),:);
347
           \operatorname{spect\_array}(:,:,j) = \operatorname{spect};
348
           \operatorname{sp_-1c}(:,j) = \operatorname{reshape}(\operatorname{spect},[8001*\operatorname{length}(\operatorname{tstep}),1]);
349
     end
350
351
    figure (1)
352
353 pcolor(tstep, ks_Hz, spect_array(:,:,20)),
354 shading interp
     title ('Unfiltered: a = 1e03, inc = 0.0446', 'Fontsize', 16)
356
     set (gca, 'Fontsize', 11)
357
     colormap(hot)
358
359 % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
          HW_{-}04/test_{-}01/test_{-}01c_{-}sp.matsp_{-}1c
```

```
360
    %% Test 1 b
361
362
363 close all; clear all; clc;
364
    load test_01/test_01b.mat
365
366
367 % set the time and frequency domains
368
369 \quad \mathbf{size} = \mathbf{size} (dat_1b);
370 n=size(2); % number of samples
371 s = size(1); % length of samples
372 L=s/(44100/2); % time domain
373 t2 = linspace(0, L, s+1);
374 \quad t=t2(1:s);
375 k_Hz= (1/(2*L))*[0:(s/2-1)-s/2:-1];
376 \text{ ks-Hz} = \mathbf{fftshift} (\text{k-Hz});
377 ks_Hz = ks_Hz ((s/2-4000):(s/2+4000));
378
379 \quad a = 2e4;
    tstep = 0:0.3:L;
380
381
382
383 % array of spectrogram data
384 \text{ sp.1b} = \mathbf{zeros}(8001*\mathbf{length}(tstep), n);
385
     spect_array = zeros(8001, length(tstep), n);
386
387
    for j=1:n
388
          % spectrogram of clip
389
          spect = zeros(s, length(tstep));
390
          for k=1:length(tstep)
391
               \mathbf{g} = \mathbf{exp}(-\mathbf{a} * (\mathbf{t} - \mathbf{t} \mathbf{s} \mathbf{t} \mathbf{e} \mathbf{p} (\mathbf{k})) . \hat{2});
               Sg=g'.*dat_1b(:,j); % apply new filter to the current 5sec
392
393
               Sgt=fft (Sg); % transform to frequency domain
394
               spect(:,k) = fftshift(abs(Sgt));
395
          end
396
          spect = spect((s/2-4000):(s/2+4000),:);
397
          spect_array(:,:,j) = spect;
398
          \operatorname{sp-1b}(:,j) = \operatorname{reshape}(\operatorname{spect},[8001*\operatorname{length}(\operatorname{tstep}),1]);
399
     end
400
401
402
     figure (1)
403
     pcolor (tstep, ks_Hz, spect_array(:,:,20)),
404
    shading interp
     title ('Unfiltered: a = 1e03, inc = 0.0446', 'Fontsize', 16)
    set (gca, 'Fontsize', 11)
406
407
    colormap(hot)
408
409
     % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
         HW_04/test_01/test_01b_sp.mat\ sp_1b
410
411 %% Test 1 a
```

```
412
413
     close all; clear all; clc;
414
    load test_01/test_01a.mat
415
416
417 % set the time and frequency domains
418
419 \operatorname{\mathbf{size}} = \operatorname{\mathbf{size}}(\operatorname{dat}_{-}1a);
                    % number of samples
420 n=size(2);
421 s = size(1); % length of samples
422 L=s/(44100/2); % time domain
423 t2 = linspace(0, L, s+1);
424 \quad t=t2(1:s);
425 k_{-}Hz = (1/(2*L))*[0:(s/2-1) -s/2:-1];
     ks_Hz = fftshift(k_Hz);
427
    ks_Hz = ks_Hz((s/2-4000):(s/2+4000));
428
429 \quad a = 2e4;
430
    tstep = 0:0.3:L;
431
432
433
    % array of spectrogram data
     sp_1a = zeros(8001*length(tstep),n);
435
     spect_array = zeros(8001, length(tstep), n);
436
437
     for j=1:n
438
         % spectrogram of clip
          spect = zeros(s, length(tstep));
439
440
          for k=1:length(tstep)
441
              g=exp(-a*(t-tstep(k)).^2);
442
              Sg=g'.*dat_1a(:,j); \% apply new filter to the current 5sec
443
              Sgt=fft (Sg); % transform to frequency domain
444
              spect(:,k) = fftshift(abs(Sgt));
445
         end
446
          spect = spect((s/2-4000):(s/2+4000),:);
447
          spect_array(:,:,j) = spect;
448
          \operatorname{sp-1a}(:,j) = \operatorname{reshape}(\operatorname{spect},[8001*\operatorname{length}(\operatorname{tstep}),1]);
449
    end
450
    %spect = reshape(sp_1a(:,1), [length(ks_Hz), length(tstep)]);
451
452
453
     figure (1)
     pcolor(tstep , ks_Hz , spect_array(:,:,20)),
454
     shading interp
     title ('Unfiltered: a = 1e03, inc = 0.0446', 'Fontsize', 16)
456
457
     set (gca, 'Fontsize', 11)
458
    colormap(hot)
459
460
    % save /Users/christinasmith/Desktop/AMATH_482.nosync/Homework/
         HW_04/test_01/test_01a_sp.matsp_1a
```

#### $HW_04_combined.m$

```
1 % Audio classifier
3
   % Load in spectrogram files
4
5
   clear; close all; clc
6
7
   load test_01/test_01a_sp.mat
8
   load test_01/test_01b_sp.mat
9
   load test_01/test_01c_sp.mat
10
11
   load test_02/test_02a_sp.mat
12
   load test_02/test_02b_sp.mat
   load test_02/test_02c_sp.mat
14
15
  load test_03/test_03a_sp.mat
   load test_03/test_03b_sp.mat
17
   load test_03/test_03c_sp.mat
18
19
   %% Train classifier
20
  feature = 50;
21
22
   size_train = 25;
23
   size\_test = 15;
24
25
  feature = 80;
26
   size_train = 55;
27
   size_test = 15;
28
29 \% train_a = sp_1a(:, 1: size_train);
30 \% train_b = sp_1b(:, 1:size_train);
31 \% train_c = sp_1c(:,1:size_train);
32 %
33 % train_a = sp_2a(:, 1: size_train);
34 \% train_b = sp_2b(:,1:size_train);
35 \% train_c = sp_2c(:,1:size_train);
36
   train_a = sp_3a(:,1:size_train);
37
38
   train_b = sp_3b(:,1:size_train);
39
   train_c = sp_3c(:,1:size_train);
40
41
42
43
   [U,S,V,thresholds,d_arr,w_arr,rank,proj_w1, proj_w2] = trainer(
44
       train_3a, train_3b, train_3c, feature);
45
46 % compute the mean position in the w1/w2 plane
   class_1 = [mean(proj_w1(1,:)), mean(proj_w2(1,:))];
47
   class_2 = [mean(proj_w1(2,:)), mean(proj_w2(2,:))];
48
   class_3 = [mean(proj_w1(3,:)), mean(proj_w2(3,:))];
49
50
51
   means = [class_1; class_2; class_3];
52
53 %% Test 3
```

```
54
 55
    \dim = \mathbf{size}(\mathrm{sp}_{-}3a);
 56
   n = \dim(2);
 57
 58 % % % test the different classes
 59 \quad \text{test\_a} = \text{sp\_3a}(:, \text{size\_train} + 1:n);
 60 \operatorname{test_b} = \operatorname{sp_3b}(:, \operatorname{size\_train} + 1:n);
 61 \quad test_c = sp_3c(:, size_train + 1:n);
 62
 63 % project the test data on the principal components {\it E\!E} w vectors
 64 \text{ pval}_a = \text{w}_{arr}*\text{U}'*\text{test}_a;
 65 \text{ pval_b} = \text{w_arr*U'*test_b};
 66 pval_c = w_arr*U'* test_c;
 67
 68
 69 \% class_a 1 = classify_1 D(pval_a(1,:), thresholds(1,:), rank);
 70 % class_b1 = classify_1D(pval_b(1,:), thresholds(1,:), rank);
 71 \% class_c1 = classify_1D(pval_c(1,:), thresholds(1,:), rank);
 72 \%
 73 % tc_a = sum(class_a1 == 1);
 74 % tt_a = length(class_a1);
 75 % pc_a = tc_a/tt_a;
 76 %
 77 \% \ tc_b = sum(class_b1 == 2);
 78 % tt_b = length(class_b1);
 79 \% pc_b = tc_b/tt_b;
 80 %
 81 % tc_{-}c = sum(class_{-}c1 == 3);
 82 % tt_c = length(class_c1);
 83 \% pc_c = tc_c/tt_c;
 84
 85 \quad class_a2 = classify_2D (pval_a, means);
 86 \quad class_b2 = classify_2D (pval_b, means);
 87
    class_c2 = classify_2D (pval_c, means);
 88
 89 tc_a = sum(class_a2 == 1);
 90 tt_a = length(class_a2);
 91 	 pc_a = tc_a/tt_a;
 92
 93 tc_b = sum(class_b2 == 2);
 94 \text{ tt_b} = \text{length}(\text{class_b2});
 95 pc_b = tc_b/tt_b;
 97 tc_c = sum(class_c2 == 3);
 98
    tt_c = length(class_c2);
 99
     pc_{-c} = tc_{-c}/tt_{-c};
100
101 \mathref{7}\mathref{7} Plot the test and training data in 1D
102
103 figure(1)
104 subplot (1,2,1)
105
     plot (proj_w1 (1,:), ones (size_train), 'ob', 'Linewidth', 2)
106 hold on
     plot (proj_w1 (2,:), ones (size_train) *2, 'or', 'Linewidth', 2)
```

```
108 hold on
109 plot (proj_w1 (3,:), ones (size_train) *3, 'og', 'Linewidth', 2)
110 ylim ([0.5 \ 3.5])
111 plot ([thresholds (1,1); thresholds (1,1)], get (gca, 'vlim'), 'r')
112 hold on
113 plot ([thresholds (1,2); thresholds (1,2)], get (gca, 'ylim'), 'r')
114
     title ('Test 3: training data', 'Fontsize', 16)
115 LH(1) = \mathbf{plot}(\operatorname{nan}, \operatorname{nan}, \operatorname{ob}');
116 L\{1\} = 'CLASS A';
117 LH(2) = \mathbf{plot}(\operatorname{nan}, \operatorname{nan}, \operatorname{or});
118 L\{2\} = 'CLASS B';
119 LH(3) = \mathbf{plot}(\operatorname{nan}, \operatorname{nan}, \operatorname{og});
120 L{3} = 'CLASS C';
121 legend(LH, L, 'FontSize', 12, 'Location', 'northeast');
     xlabel('w1 values', 'FontSize', 12, 'Fontweight', 'bold');
     yticks ([1 2 3])
     yticklabels ({ 'Class A', 'Class B', 'Class C'})
125 hold off
126
127 % Plot the test data 1D
128 subplot (1,2,2)
129 plot (pval_a (1,:), ones (length (pval_a (1,:))), 'ob', 'Linewidth', 2)
130 hold on
131 plot (pval_b (1,:), ones (length (pval_b (1,:))) *2, 'or', 'Linewidth', 2)
132 hold on
     plot (pval_c (1,:), ones (length (pval_c (1,:))) *3, 'og', 'Linewidth', 2)
133
134 hold on
135 ylim ([0.5 3.5])
    \operatorname{plot}([\operatorname{thresholds}(1,1);\operatorname{thresholds}(1,1)],\operatorname{\mathbf{get}}(\operatorname{\mathbf{gca}}, '\operatorname{ylim}'), '\operatorname{r}')
136
137
     hold on
138 plot ([thresholds (1,2); thresholds (1,2)], get (gca, 'ylim'), 'r')
139 title ('Test 3: test data', 'Fontsize', 16)
140 LH(1) = plot (nan, nan, 'ob');
141 L\{1\} = 'CLASS A';
142 LH(2) = plot(nan, nan, 'or');
143 L\{2\} = 'CLASS B';
144 LH(3) = \mathbf{plot}(\operatorname{nan}, \operatorname{nan}, \operatorname{og});
145 \text{ L}{3} = \text{'CLASS C'};
146 legend(LH, L, 'FontSize', 12, 'Location', 'northeast');
147
    xlabel('w1 values', 'FontSize', 12, 'Fontweight', 'bold');
148
     yticks ([1 2 3])
149
     yticklabels ({ 'Class A', 'Class B', 'Class C'})
150
     hold off
151
152
     % Plot the test and training data in 2D
153
154
155 figure (2)
156 subplot (1,2,1)
157
     plot (proj_w1 (1,:), proj_w2 (1,:), 'bo')
158 hold on;
     plot (class_1 (1), class_1 (2), 'o', 'MarkerFaceColor', 'b')
159
160 hold on;
     plot (proj_w1 (2,:), proj_w2 (2,:), 'ro')
```

```
162 hold on;
163 plot (class_2(1), class_2(2), 'o', 'MarkerFaceColor', 'r')
164 hold on;
165 plot (proj_w1 (3,:), proj_w2 (3,:), 'go')
166 hold on;
    plot (class_3 (1), class_3 (2), 'o', 'MarkerFaceColor', 'g')
167
168
     hold on;
     title ('Test 3: training data', 'Fontsize', 16)
169
170 xlabel('w1 values', 'FontSize', 12, 'Fontweight', 'bold');
    ylabel('w2 values', 'FontSize', 12, 'Fontweight', 'bold');
171
172 LH(1) = plot(nan, nan, 'ob');
173 L\{1\} = 'CLASS A';
174 \text{ LH}(2) = \mathbf{plot}(\text{nan}, \text{nan}, \text{'or'});
175 \text{ L}\{2\} = \text{'CLASS B'};
176 LH(3) = plot(nan, nan, 'og');
177 \text{ L}{3} = \text{'CLASS C'};
178 legend(LH, L, 'FontSize', 12, 'Location', 'southeast');
179 hold off
180
    % Plot test data 2D
181
182
183 subplot (1,2,2)
184 plot (means (1,1), means (1,2), 'o', 'MarkerFaceColor', 'b')
185 hold on;
186 plot (means (2,1), means (2,2), 'o', 'MarkerFaceColor', 'r')
187
    hold on;
188 plot (means (3,1), means (3,2), 'o', 'MarkerFaceColor', 'g')
189 hold on;
    plot (pval_a (1,:), pval_a (2,:), 'bo')
190
191
    hold on;
192 plot (pval_b (1,:), pval_b (2,:), 'ro')
193 hold on;
194 plot (pval_c (1,:), pval_c (2,:), 'go')
     title ('Test 3: test data', 'Fontsize', 16)
196 xlabel('w1 values', 'FontSize', 12, 'Fontweight', 'bold');
    ylabel('w2 values', 'FontSize', 12, 'Fontweight', 'bold');
197
198 LH(1) = plot(nan, nan, 'ob');
199 L\{1\} = 'CLASS A';
200 \text{ LH}(2) = \mathbf{plot}(\text{nan}, \text{nan}, \text{'or'});
201 L\{2\} = 'CLASS B';
202 \text{ LH}(3) = \mathbf{plot}(\text{nan}, \text{nan}, '\text{og}');
203 L{3} = 'CLASS C';
    legend(LH, L, 'FontSize', 12, 'Location', 'southeast');
205
    hold off
206
207
208 % Plot histograms of test data 1D
209
210 \% bins = 1;
211 %
212 % % Plot histogram of test data with thresholds
213 % figure (3)
214 % subplot (1,3,1)
215 % histogram(pval_a(1,:),bins);
```

```
216 \% xlim([-500 1500])
217 \% hold on,
218 \% plot ([thresholds (1,1); thresholds (1,1)], get (gca, 'ylim'), 'r')
219 % hold on
                                                         'ylim'), 'r')
220 % plot([thresholds(1,2);thresholds(1,2)],get(gca,
                                          class 1 range
221 % xlabel('class 2 range
        class 3 range',...
222 %
           'Fontsize', 10, 'Fontweight', 'bold')
223 % title ('Test 1: class 1 data', 'Fontsize', 14)
224 \% hold off
225 \%
226 % subplot (1,3,2)
227 % histogram (pval_b (1,:), bins);
228 \% xlim([-500 1500])
229 \% hold on,
230 % plot([thresholds(1,1); thresholds(1,1)], get(gca, 'ylim'), 'r')
231 \% hold on
232 % plot([thresholds(1,2);thresholds(1,2)],get(gca, 'ylim'),'r')
233 % xlabel ('class 2 range
                                          class 1 range
        class 3 range',...
           'Fontsize', 10, 'Fontweight', 'bold')
234 %
235 % title ('Test 1: class 2 data', 'Fontsize', 14)
236 % hold off
237 %
238 % subplot (1,3,3)
239 \% histogram (pval_c(1,:), bins);
240 \% xlim([-500 1500])
241 \% hold on,
242 % plot([thresholds(1,1);thresholds(1,1)],get(gca, 'ylim'),'r')
243 % hold on
244 % plot([thresholds(1,2); thresholds(1,2)], get(gca, 'ylim'), 'r')
245 % xlabel ('class 2 range
                                          class 1 range
        class 3 range',...
'Fontsize', 10, 'Fontweight', 'bold')
246
    %
    % title ('Test 1: class 3 data', 'Fontsize', 14)
247
248 % hold off
249
250 %% Classify using 1D
251
252 % Classify using first projection vector
253
254
    function classes = classify_1D (pval, threshold, rank)
255
256
        \dim = \mathbf{size}(pval);
257
        n = \dim(2);
258
         classes = zeros(1,n);
259
260
         for j=1:n
261
              if pval(j)<threshold(2) && pval(j)>threshold(1)
262
                 classes(j) = rank(2);
263
             elseif pval(j)>threshold(2)
264
                 classes(j) = rank(3);
265
             else
266
                 classes(j) = rank(1);
```

```
267
             end
268
         end
269
    end
270
271
    % Classify 2D
272
273
274
    % Classify using both projection vectors
275
276
    function classes = classify_2D (pvals, means)
277
278
         \dim = \mathbf{size}(\text{pvals});
279
         n = \dim(2);
280
         classes = zeros(1,n);
281
282
         \% for every spectrogram in the data
283
         for j=1:n
             A = [pvals(:,j)'; means(1,:)];
284
285
             B = [pvals(:,j)'; means(2,:)];
             C = [pvals(:,j)'; means(3,:)];
286
287
             % compute the minimumm distance to a class mean
288
              distances = [pdist(A, 'euclidean') ...
289
290
              pdist (B, 'euclidean')...
291
              pdist(C, 'euclidean')];
292
293
              [\tilde{\ }, \min_{i} dx] = \min(distances);
294
              classes(j) = min_i dx;
295
         end
296
297
    end
298
299
300
    function [U,S,V,thresholds,d_arr,w_arr,rank,proj_w1,proj_w2] =
        trainer (s1, s2, s3, feature)
301
302
         \% a 800k by 30 matrix of the spectrograms from all 3 groups
303
         X = [s1 \ s2 \ s3];
304
305
         ns1 = size(s1,2);
306
         ns2 = size(s2,2);
307
         ns3 = size(s3,2);
308
         [U,S,V] = \mathbf{svd}(X, 'econ');
309
310
         songs = S*V'; % projection onto principal components
311
312
313
         % choose how many singular values we want
314
         U = U(:, 1: feature);
315
316
         % separate the songs now that they are
317
         % in the principal component basis
         song_1 = songs(1:feature, 1:ns1);
318
319
         song_2 = songs(1:feature, ns1+1:ns1+ns2);
```

```
320
         song_3 = songs(1:feature, ns1+ns2+1:ns1+ns2+ns3);
321
322
         ms_all = mean(songs(1:feature,:),2);
323
         ms1 = mean(song_1, 2);
324
         ms2 = mean(song_2, 2);
325
         ms3 = mean(song_3, 2);
326
327
328
        Sw = 0; % within class variances
329
         for k=1:ns1
             Sw = Sw + (song_1(:,k)-ms1)*(song_1(:,k)-ms1)';
330
331
        end
332
         for k=1:ns2
333
             Sw = Sw + (song_2(:,k)-ms2)*(song_2(:,k)-ms2)';
334
         end
335
         for k=1:ns3
336
             Sw = Sw + (song_3(:,k)-ms_3)*(song_3(:,k)-ms_3)';
337
         end
338
339
         Sb = (ms1-ms_all)*(ms1-ms_all)'; \% between class
340
         Sb = Sb + (ms2-ms_all)*(ms2-ms_all)';
341
         Sb = Sb + (ms3-ms_all)*(ms3-ms_all)';
342
343
344
        % get the eigenvalue
         [V2,D] = eig(Sb,Sw); % linear discriminant analysis
345
346
        \max(\max(D));
347
348
         [d1, ind] = \max(abs(diag(D)));
349
350
        % eigenvector with max eigenvalue
351
         w1 = V2(:, ind);
        w1 = w1/norm(w1, 2);
352
353
354
        D(:, ind) = [];
355
         [d2, ind] = \max(abs(diag(D)));
356
357
         d_{arr} = [d1, d2];
358
359
        % eigenvector with 2nd max eigenvalue
         w2 = V2(:, ind);
360
361
         w2 = w2/\mathbf{norm}(w2, 2);
362
         w_{arr} = [w1 \ w2]';
363
364
365
         \% project the song data onto w1
366
         sort_s1a = w1*song_1;
367
         sort_s2a = w1'*song_2;
368
         sort_s3a = w1'*song_3;
369
370
         proj_w1 = [sort_s1a; sort_s2a; sort_s3a];
371
372
        % project the song data onto w2
         sort_s1b = w2'*song_1;
373
```

```
374
         sort_s2b = w2'*song_2;
375
         sort_s3b = w2*song_3;
376
377
         proj_w2 = [sort_s1b; sort_s2b; sort_s3b];
378
379
380
         % figure out which means are 'neighbors'
381
382
         mv_a = [mean(sort_s1a) mean(sort_s2a) mean(sort_s3a)];
383
         for j = 1:3
384
385
              \mathbf{if} \ \mathrm{mv}_{-}\mathrm{a}(\mathrm{j}) = \mathbf{max}(\mathrm{mv}_{-}\mathrm{a})
386
                  \% max_val = mv_a(j);
387
                  max_class_a = j;
388
              elseif mv_a(j) = min(mv_a)
                  \% min_val = mv_a(j);
389
390
                  min_class_a = j;
391
              else
392
                  \% med_-val = mv_-a(j);
393
                  med_class_a = j;
394
              end
395
         end
396
397
         rank = [min_class_a med_class_a max_class_a];
398
399
         vsm = [sort_s1a; sort_s2a; sort_s3a];
400
401
         % sort all values from lowest to highest
402
         sort_max_a = sort(vsm(max_class_a,:));
403
         sort_med_a = sort(vsm(med_class_a,:));
404
         sort_min_a = sort(vsm(min_class_a,:));
405
            thresholds(1,1) = (mean(sort\_med\_a) - mean(sort\_min\_a))/2;
406
    %
407
    %
            thresholds(1,2) = (mean(sort_max_a) - mean(sort_med_a))/2;
         t1a = length(sort_max_a);
408
409
         t2a = 1;
410
         while (t1a > 0 \&\& sort_max_a(t1a) > sort_med_a(t2a))
411
         t1a = t1a - 1;
412
         t2a = t2a+1;
413
         end
414
         if t1a = 0
415
         thresholds(1,2) = (sort_max_a(1) + sort_med_a(length(sort_max_a)))
             ))/2;
416
         else
417
         thresholds(1,2) = (sort_max_a(t1a) + sort_med_a(t2a))/2;
418
419
420
421
         t1b = length(sort_med_a);
422
         t2b = 1;
423
         while (t1b > 0 \&\& sort_med_a(t1b) > sort_min_a(t2b))
424
         t1b = t1b - 1;
425
         t2b = t2b+1;
426
         end
```

```
427
428
         if t1b = 0
429
         thresholds(1,1) = (sort_med_a(1) + sort_min_a(length(sort_med_a))
            ))/2;
430
         else
431
         thresholds (1,1) = (sort_med_a(t1b) + sort_min_a(t2b))/2;
432
         end
433
         % figure out which means are 'neighbors'
434
435
436
         mv_b = [mean(sort_s1b) mean(sort_s2b) mean(sort_s3b)];
437
438
         for j=1:3
439
             if mv_b(j) = max(mv_b)
440
                 \% max_val = mv_a(j);
                 max_class_b = j;
441
442
             elseif mv_b(j) = min(mv_b)
443
                 \% min_val = mv_a(j);
444
                 min_class_b = j;
445
             else
446
                 \% med_val = mv_a(j);
                 med_class_b = j;
447
448
             end
449
         end
450
         vsm = [sort_s1b; sort_s2b; sort_s3b];
451
452
453
         % sort all values from lowest to highest
454
         sort_max_b = sort(vsm(max_class_b,:));
455
         sort_med_b = sort(vsm(med_class_b,:));
456
         sort_min_b = sort(vsm(min_class_b,:));
457
458
459
460
   %
           thresholds(2,1) = (mean(sort\_med\_b) - mean(sort\_min\_b))/2;
461 %
           thresholds(2,2) = (mean(sort_max_b) - mean(sort_med_b))/2;
462
463
464
         t1a = length(sort_max_b);
465
         t2a = 1;
466
         while (t1a > 0 \&\& sort_max_b(t1a) > sort_med_b(t2a))
467
         t1a = t1a-1;
468
         t2a = t2a+1;
469
         end
470
         if t1a = 0
         thresholds(2,2) = (sort_max_b(1) + sort_med_b(length(sort_max_b)))
471
            ))/2;
472
         _{
m else}
473
         thresholds(2,2) = (sort_max_b(t1a) + sort_med_b(t2a))/2;
474
         end
475
476
477
         t1b = length(sort_med_b);
478
         t2b = 1;
```

```
479
         while (t1b > 0 \&\& sort_med_b(t1b) > sort_min_b(t2b))
480
         t1b = t1b - 1;
481
         t2b = t2b+1;
482
        end
483
         if t1b = 0
484
         thresholds(2,1) = (sort\_med\_b(1) + sort\_min\_b(length(sort\_med\_b))
485
            ))/2;
486
         else
487
         thresholds(2,1) = (sort_med_b(t1b) + sort_min_b(t2b))/2;
488
489
490 end
```