Music Genre Classification with LSTMs

Feature extraction Spectral features % chroma_stft ([y, sr, S, norm, n_fft, ...]) Compute a chromagram from a waveform or power spec chroma_cqt ([y, sr, C, hop_length, fmin, ...]) Constant-Q chromagram Computes the chroma variant "Chroma Energy Normaliz chroma_cens ([y, sr, C, hop_length, fmin, ...]) melspectrogram ([y, sr, S, n_fft, ...]) Compute a mel-scaled spectrogram. mfcc ([y, sr, S, n_mfcc, dct_type, norm]) Mel-frequency cepstral coefficients (MFCCs) rmse ([y, S, frame_length, hop_length, ...]) Compute root-mean-square (RMS) energy for each fram Compute the spectral centroid. spectral_centroid ([y, sr, S, n_fft, ...]) spectral_bandwidth ([y, sr, S, n_fft, ...]) Compute p'th-order spectral bandwidth: spectral_contrast ([y, sr, S, n_fft, ...]) Compute spectral contrast [R6ffcc01153df-1] spectral_flatness ([y, S, n_fft, hop_length, ...]) Compute spectral flatness spectral_rolloff ([y, sr, S, n_fft, ...]) Compute roll-off frequency poly_features ([y, sr, S, n_fft, hop_length, ...]) Get coefficients of fitting an nth-order polynomial to the tonnetz ([y, sr, chroma]) Computes the tonal centroid features (tonnetz), followin zero crossing rate (y[, frame length, ...]) Compute the zero-crossing rate of an audio time series.

经大量研究表明, MFCCs(Mel Frequency Cepstral Coefficients)是一种在自动语音和说话人识别中广泛使用的特征。在语音识别领域,MFCCs在人工特征方面被证明十分有效,所以在此保留mfcc这一特征。

此外, chroma_stft, chroma_cqt, chroma_cens 相似性很高, 可视为一种特征, 同理, spectral_centroid, spectral_bandwidth, spectral_contrast, spectral_flatness, spectral_rolloff 也可视为一种特征。

因此可得如下特征组合实验结果。

```
for i, file in enumerate(list_of_audiofiles):
  y, sr = librosa.load(file)
   mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
   spectral_center = librosa.feature.spectral_centroid(y=y, sr=sr, hop_length=self.hop_length)
   chroma = 1ibrosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
   rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)
   splits = re.split('[.]', file)
   genre = re.split('[/]', splits[1])[3]
   target.append(genre)
   data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
   data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
   data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
   data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
Epoch 400/400
 35/420 [=>.....] - ETA: 4s - loss: 0.3981 - acc: 0.8286
 70/420 [====>.....] - ETA: 4s - 1oss: 0.3539 - acc: 0.8714
105/420 [=====>.....] - ETA: 3s - 1oss: 0.3192 - acc: 0.8857
140/420 [=======>)......] - ETA: 3s - loss: 0.3459 - acc: 0.8643
175/420 [======>....] - ETA: 2s - loss: 0.3430 - acc: 0.8571
210/420 [=======>....] - ETA: 2s - loss: 0.3372 - acc: 0.8571
245/420 [=======>.....] - ETA: 2s - loss: 0.3379 - acc: 0.8612
280/420 [=======> .....] - ETA: 1s - loss: 0.3226 - acc: 0.8679
350/420 [===========>....] - ETA: Os - loss: 0.3336 - acc: 0.8686
385/420 [============>...] - ETA: Os - loss: 0.3415 - acc: 0.8675
420/420 [============ ] - 5s 12ms/step - 1oss: 0.3838 - acc: 0.8595
Validating ...
 35/120 [======>.....] - ETA: 1s
120/120 [======== ] - 1s 7ms/step
Dev loss: 1.373569389184316
Dev accuracy: 0.5833333544433117
Testing ...
35/60 [=======>....] - ETA: Os
60/60 [======== ] - 0s 5ms/step
Test loss: 1.2082031120856602
```

Test accuracy: 0.6166666721304258

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_bandwidth(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)

splits = re.split('[ , ]', file)
    genre = re.split('[ , ]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
```

```
## Table ##
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)

spectral_center = librosa.feature.spectral_flatness(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)

splits = re.split('[.]', file)
    genre = re.split('[.]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
```

```
Epoch 400/400
35/420 [=>.....] - ETA: 4s - loss: 0.0208 - acc: 1.0000
70/420 [====>.....] - ETA: 4s - loss: 0.0180 - acc: 1.0000
105/420 [=====>.....] - ETA: 3s - loss: 0.0215 - acc: 1.0000
140/420 [======>>.....] - ETA: 3s - loss: 0.0181 - acc: 1.0000
175/420 [=======>.....] - ETA: 3s - loss: 0.0173 - acc: 1.0000
210/420 [=======>....] - ETA: 2s - loss: 0.0177 - acc: 1.0000
245/420 [=======>.....] - ETA: 2s - 1oss: 0.0222 - acc: 0.9959
280/420 [=========>.....] - ETA: 1s - loss: 0.0394 - acc: 0.9929
315/420 [=======>>.....] - ETA: 1s - 1oss: 0.0366 - acc: 0.9937
385/420 [===============>...] - ETA: Os - loss: 0.0354 - acc: 0.9922
420/420 [============== ] - 5s 12ms/step - loss: 0.0331 - acc: 0.9929
Validating ...
70/120 [=======>....] - ETA: Os
105/120 [============>....] - ETA: 0s
120/120 [======] - 1s 7ms/step
Dev loss: 2.27315000196298
Dev accuracy: 0.6166666829958558
Testing ...
60/60 [======] - 0s 5ms/step
Test loss: 2.655836800734202
Test accuracy: 0.5333333512147268
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_rolloff(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)

splits = re.split('[ ]', file)
    genre = re.split('[ ]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    zero_crossing_rate = librosa.feature.zero_crossing_rate(y=y)
    splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)
    data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = zero_crossing_rate.T[0:timeseries_length, :]
Epoch 400/400
35/420 [=>.....] - ETA: 4s - loss: 0.5204 - acc: 0.7714
70/420 [====>.....] - ETA: 4s - loss: 0.4260 - acc: 0.8143
105/420 [=====>.....] - ETA: 3s - loss: 0.4514 - acc: 0.8095
140/420 [======>....] - ETA: 3s - 1oss: 0.4494 - acc: 0.8214
175/420 [=======>.....] - ETA: 2s - loss: 0.4155 - acc: 0.8457
210/420 [=======>.....] - ETA: 2s - loss: 0.3908 - acc: 0.8524
245/420 [==========>.....] - ETA: 2s - loss: 0.3719 - acc: 0.8571
280/420 [=======>: .....] - ETA: 1s - 1oss: 0.3807 - acc: 0.8571
315/420 [==========>.....] - ETA: 1s - loss: 0.3891 - acc: 0.8476
350/420 [===========>....] - ETA: Os - loss: 0.4125 - acc: 0.8429
385/420 [============>...] - ETA: Os - loss: 0.4103 - acc: 0.8468
420/420 [===========] - 5s 12ms/step - loss: 0.4061 - acc: 0.8476
Validating ...
35/120 [======>....] - ETA: 1s
120/120 [======== ] - 1s 8ms/step
Dev loss: 1.0691008294622104
Dev accuracy: 0.6416666842997074
Testing ...
60/60 [=======] - 0s 5ms/step
Test loss: 1.193656325340271
```

Test accuracy: 0.5500000094374021

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_bandwidth(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    zero_crossing_rate = librosa.feature.zero_crossing_rate(y=y)

splits = re.split('[ , ]' , file)
    genre = re.split('[ , ]' , splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = zero_crossing_rate.T[0:timeseries_length, :]
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_flatness(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    zero_crossing_rate = librosa.feature.zero_crossing_rate(y=y, hop_length=self.hop_length)

splits = re.split('[ , ]', file)
    genre = re.split('[ , ]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = zero_crossing_rate.T[0:timeseries_length, :]
```

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### Table #### Table #
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```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_rolloff(r=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    zero_crossing_rate = librosa.feature.zero_crossing_rate(y=y)

splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = zero_crossing_rate.T[0:timeseries_length, :]
```

```
Epoch 400/400
35/420 [=>.....] - ETA: 4s - loss: 0.4505 - acc: 0.8000
70/420 [====>.....] - ETA: 4s - loss: 0.4440 - acc: 0.8143
105/420 [=====>.....] - ETA: 3s - loss: 0.4322 - acc: 0.8286
140/420 [======>.....] - ETA: 3s - loss: 0.4767 - acc: 0.8143
175/420 [=======>....] - ETA: 2s - loss: 0.4647 - acc: 0.8171
210/420 [=======>....] - ETA: 2s - loss: 0.4810 - acc: 0.8048
245/420 [=======>.....] - ETA: 2s - loss: 0.4850 - acc: 0.8122
280/420 [=======>>.....] - ETA: 1s - loss: 0.5203 - acc: 0.8071
315/420 [=======>:.....] - ETA: 1s - loss: 0.5180 - acc: 0.8032
350/420 [==========>....] - ETA: Os - loss: 0.5303 - acc: 0.7971
385/420 [===========>...] - ETA: Os - loss: 0.5399 - acc: 0.7870
Validating ...
35/120 [======>.....] - ETA: 1s
105/120 [==========>....] - ETA: Os
120/120 [========== ] - 1s 7ms/step
Dev loss: 0.9134505813320478
Dev accuracy: 0.6583333536982536
Testing ...
60/60 [======] - 0s 5ms/step
Test loss: 1.199605683485667
Test accuracy: 0.5500000044703484
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cqt(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)
    splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)
    data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
Epoch 400/400
35/420 [=>......] - ETA: 4s - 1oss: 0.1938 - acc: 0.9143
70/420 [====>.....] - ETA: 4s - loss: 0.2412 - acc: 0.9000
105/420 [=====>.....] - ETA: 3s - 1oss: 0.2878 - acc: 0.8952
140/420 [=======>.....] - ETA: 3s - 1oss: 0.2778 - acc: 0.9071
175/420 [=======>....] - ETA: 3s - loss: 0.2677 - acc: 0.9086
210/420 [======>: .....] - ETA: 2s - loss: 0.2925 - acc: 0.9048
245/420 [=========>.....] - ETA: 2s - 1oss: 0.2869 - acc: 0.8980
280/420 [============>....] - ETA: 1s - loss: 0.2831 - acc: 0.9000
315/420 [==============>, .....] - ETA: 1s - loss: 0.2996 - acc: 0.8921
385/420 [==========================>...] - ETA: Os - loss: 0.3332 - acc: 0.8753
420/420 [===============] - 5s 12ms/step - loss: 0.3233 - acc: 0.8810
Validating ...
35/120 [======>.....] - ETA: 1s
70/120 [========>....] - ETA: Os
105/120 [============>....] - ETA: Os
120/120 [======== ] - 1s 7ms/step
Dev loss: 1.1678892870744069
Dev accuracy: 0.6583333536982536
Testing ...
35/60 [=======>....] - ETA: Os
60/60 [=======] - 0s 5ms/step
Test loss: 1.2402155896027882
Test accuracy: 0.6500000109275182
```

```
for i, file in enumerate(list_of_audiofiles):
   y, sr = librosa.load(file)
   mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
   spectral_center = librosa.feature.spectral_bandwidth(y=y, sr=sr, hop_length=self.hop_length)
   chroma = librosa.feature.chroma_cqt(y=y, sr=sr, hop_length=self.hop_length)
   rmse = librosa.feature.rmse(y=y)
   splits = re.split('[.]', file)
   genre = re.split('[/]', splits[1])[3]
   target.append(genre)
   data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
   data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
   data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
   data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
Epoch 400/400
70/420 [====>.....] - ETA: 4s - loss: 0.5802 - acc: 0.7286
105/420 [=====>.....] - ETA: 3s - loss: 0.4968 - acc: 0.8000
140/420 [=======>.....] - ETA: 3s - loss: 0.5133 - acc: 0.7929
175/420 [=======>.....] - ETA: 3s - loss: 0.4896 - acc: 0.8000
210/420 [=======>.....] - ETA: 2s - loss: 0.5001 - acc: 0.7952
245/420 [=======>....] - ETA: 2s - loss: 0.4949 - acc: 0.7918
280/420 [=========>.....] - ETA: 1s - loss: 0.5077 - acc: 0.7893
315/420 [==========>.....] - ETA: 1s - loss: 0.5133 - acc: 0.7841
350/420 [===========>....] - ETA: Os - loss: 0.5151 - acc: 0.7829
385/420 [============>...] - ETA: Os - loss: 0.5163 - acc: 0.7870
420/420 [=================] - 5s 12ms/step - 1oss: 0.4985 - acc: 0.7976
Validating ...
35/120 [======>.....] - ETA: 1s
70/120 [=======>....] - ETA: 0s
105/120 [===========>....] - ETA: 0s
120/120 [=========== ] - 1s 7ms/step
Dev loss: 1.0489776755372684
Dev accuracy: 0.6083333510905504
Testing ...
35/60 [========>....] - ETA: Os
60/60 [======] - 0s 5ms/step
Test loss: 1.5920234620571136
```

Test accuracy: 0.5333333512147268

```
12.
 for i, file in enumerate(list_of_audiofiles):
     y, sr = librosa.load(file)
     mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
     spectral_center = librosa.feature.spectral_rolloff(y=y, hop_length=self.hop_length)
     chroma = librosa.feature.chroma_cqt(y=y, sr=sr, hop_length=self.hop_length)
     rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)
     splits = re.split('[.]', file)
     genre = re.split('[/]', splits[1])[3]
     target.append(genre)
     data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
     data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
     data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
     data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
 Epoch 400/400
 35/420 [=>.....] - ETA: 4s - loss: 0.3650 - acc: 0.8571
  70/420 [====>.....] - ETA: 3s - loss: 0.3188 - acc: 0.8714
 105/420 [=====>......] - ETA: 3s - 1oss: 0.4842 - acc: 0.8286
 140/420 [=======>.....] - ETA: 3s - loss: 0.4743 - acc: 0.8143
 175/420 [=======>.....] - ETA: 2s - 1oss: 0.4710 - acc: 0.8229
```

```
210/420 [=======>....] - ETA: 2s - loss: 0.4542 - acc: 0.8333
245/420 [=========>.....] - ETA: 2s - loss: 0.4430 - acc: 0.8367
280/420 [=======>.....] - ETA: 1s - 1oss: 0.4404 - acc: 0.8393
315/420 [=========>.....] - ETA: 1s - loss: 0.4376 - acc: 0.8349
350/420 [=========>....] - ETA: Os - loss: 0.4282 - acc: 0.8429
385/420 [===========>...] - ETA: Os - loss: 0.4110 - acc: 0.8494
Validating ...
35/120 [======>.....] - ETA: 1s
70/120 [========>....] - ETA: Os
105/120 [=============>....] - ETA: Os
Dev loss: 1.1132373834649723
Dev accuracy: 0.6166666857898235
Testing ...
35/60 [========>....] - ETA: 0s
60/60 [======] - 0s 5ms/step
Test loss: 0.9893808513879776
Test accuracy: 0.6500000208616257
```

```
for i, file in enumerate(list_of_audiofiles):
   y, sr = librosa.load(file)
   mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
   spectral_center = librosa.feature.spectral_centroid(y=y, sr=sr, hop_length=self.hop_length)
   {\tt chroma = librosa.feature.chroma\_cens(y=y, sr=sr, hop\_length=self.hop\_length)}
   rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)
   splits = re.split('[.]', file)
   genre = re.split('[/]', splits[1])[3]
   target.append(genre)
   data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
   data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
   data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
   data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
Epoch 400/400
 70/420 [====>.....] - ETA: 1s - 1oss: 0.3148 - acc: 0.9000
 105/420 [=====>.....] - ETA: 1s - 1oss: 0.4082 - acc: 0.8571
 140/420 [=======>.....] - ETA: 1s - loss: 0.3624 - acc: 0.8786
 175/420 [======>....] - ETA: Os - loss: 0.3770 - acc: 0.8743
210/420 [========>.....] - ETA: 0s - loss: 0.3632 - acc: 0.8810
245/420 [=======>.....] - ETA: Os - loss: 0.3586 - acc: 0.8816
280/420 [=======>:.....] - ETA: Os - loss: 0.3301 - acc: 0.8929
315/420 [==========>.....] - ETA: 0s - loss: 0.3383 - acc: 0.8889
350/420 [===========>....] - ETA: Os - loss: 0.3478 - acc: 0.8800
385/420 [============>...] - ETA: Os - loss: 0.3613 - acc: 0.8701
420/420 [===========] - 2s 4ms/step - loss: 0.3626 - acc: 0.8643
Validating ...
 35/120 [======>....] - ETA: 0s
 105/120 [============>....] - ETA: 0s
120/120 [======== ] - 0s 3ms/step
Dev loss: 1.1321315790216129
Dev accuracy: 0.666666828095913
Testing ...
60/60 [=======] - Os 909us/step
Test loss: 1.2191942036151886
Test accuracy: 0.6833333422740301
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cens(y=y, sr=sr, hop_length=self.hop_length)
    spectral_contrast = librosa.feature.spectral_contrast(y=y, sr=sr, hop_length=self.hop_length)

splits = re.split('[.]', file)
    genre = re.split('[.]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = spectral_contrast.T[0:timeseries_length, :]
```

```
Epoch 400/400
35/420 [=>.....] - ETA: 1s - loss: 0.3663 - acc: 0.8857
70/420 [====>.....] - ETA: 1s - 1oss: 0.2992 - acc: 0.9000
105/420 [=====>.....] - ETA: 1s - loss: 0.3019 - acc: 0.9048
175/420 [=======>....] - ETA: 1s - loss: 0.3143 - acc: 0.8800
210/420 [========>.............] - ETA: Os - loss: 0.3222 - acc: 0.8810
245/420 [=======>....] - ETA: Os - loss: 0.3409 - acc: 0.8735
280/420 [=======>.....] - ETA: Os - loss: 0.3224 - acc: 0.8857
350/420 [=========>....] - ETA: Os - loss: 0.3051 - acc: 0.8914
385/420 [==========>...] - ETA: Os - loss: 0.3115 - acc: 0.8857
420/420 [============] - 2s 4ms/step - 1oss: 0.3237 - acc: 0.8810
Validating ...
105/120 [=======>....] - ETA: 0s
120/120 [============ ] - 0s 3ms/step
Dev loss: 1.0959213078022003
Dev accuracy: 0.6416666842997074
Testing ...
60/60 [=======] - 0s 884us/step
Test loss: 1.1959176063537598
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cens(y=y, sr=sr, hop_length=self.hop_length)
    spectral_bandwidth = librosa.feature.spectral_bandwidth(y=y, sr=sr, hop_length=self.hop_length)

splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = spectral_bandwidth.T[0:timeseries_length, :]
```

```
Epoch 400/400
35/420 [=>.....] - ETA: 1s - loss: 1.1163 - acc: 0.5143
70/420 [====>.....] - ETA: 1s - 1oss: 0.9843 - acc: 0.6286
105/420 [=====>.....] - ETA: 1s - 1oss: 1.0822 - acc: 0.6000
140/420 [======>.....] - ETA: 1s - loss: 1.0109 - acc: 0.6357
175/420 [======>....] - ETA: Os - loss: 0.9601 - acc: 0.6514
210/420 [=======>....] - ETA: Os - loss: 0.9193 - acc: 0.6667
245/420 [========>.....] - ETA: Os - loss: 0.9080 - acc: 0.6694
280/420 [===========>.....] - ETA: Os - loss: 0.8786 - acc: 0.6750
315/420 [=======>>.....] - ETA: 0s - loss: 0.8894 - acc: 0.6667
350/420 [=========>....] - ETA: Os - loss: 0.8771 - acc: 0.6714
385/420 [===========>...] - ETA: Os - loss: 0.8778 - acc: 0.6649
Validating ...
105/120 [==========>....] - ETA: Os
120/120 [========== ] - 0s 3ms/step
Dev loss: 1.020694226026535
Dev accuracy: 0.6166666895151138
Testing ...
35/60 [=======>....] - ETA: Os
60/60 [======= ] - 0s 901us/step
Test accuracy: 0.5833333432674408
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cens(y=y, sr=sr, hop_length=self.hop_length)
    spectral_flatness = librosa.feature.spectral_flatness(y=y, hop_length=self.hop_length)

splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 26:33] = spectral_flatness.T[0:timeseries_length, :]
```

```
Epoch 400/400
35/420 [=>.....] - ETA: 1s - loss: 0.2677 - acc: 0.8857
70/420 [====>.....] - ETA: 1s - loss: 0.3010 - acc: 0.8714
105/420 [=====>.....] - ETA: 1s - 1oss: 0.2863 - acc: 0.8952
140/420 [======>....] - ETA: 1s - loss: 0.2556 - acc: 0.9071
210/420 [=======>....] - ETA: Os - loss: 0.2766 - acc: 0.8952
245/420 [=======>:....] - ETA: Os - loss: 0.2941 - acc: 0.8980
280/420 [=========>.....] - ETA: Os - loss: 0.2730 - acc: 0.9071
315/420 [==========>.....] - ETA: Os - loss: 0.2676 - acc: 0.9079
350/420 [==========>....] - ETA: Os - loss: 0.2711 - acc: 0.9114
420/420 [===========] - 2s 4ms/step - loss: 0.2690 - acc: 0.9119
Validating ...
105/120 [=============>....] - ETA: Os
120/120 [============ ] - 0s 3ms/step
Dev loss: 0.8947887097795805
Dev accuracy: 0.7583333477377892
Testing ...
60/60 [========== ] - 0s 907us/step
Test loss: 1.029723157485326
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cens(y=y, sr=sr, hop_length=self.hop_length)
    spectral_rolloff = librosa.feature.spectral_rolloff(y=y, sr=sr, hop_length=self.hop_length)

splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 26:33] = spectral_rolloff.T[0:timeseries_length, :]
```

```
Epoch 400/400
35/420 [=>.....] - ETA: 2s - loss: 0.8835 - acc: 0.6857
70/420 [====>.....] - ETA: 1s - loss: 1.0689 - acc: 0.5571
140/420 [======>....] - ETA: 1s - loss: 1.0484 - acc: 0.5929
210/420 [=======>.....] - ETA: 1s - loss: 1.0300 - acc: 0.6190
245/420 [=======>>.....] - ETA: Os - loss: 1.0319 - acc: 0.6245
280/420 [=======>.....] - ETA: Os - loss: 1.0403 - acc: 0.6214
315/420 [========>:.....] - ETA: Os - loss: 1.0488 - acc: 0.6222
350/420 [==========>....] - ETA: Os - loss: 1.0733 - acc: 0.6086
420/420 [=========] - 2s 5ms/step - loss: 1.0416 - acc: 0.6143
105/120 [==========>....] - ETA: Os
120/120 [=======] - Os 3ms/step
Dev loss: 1.000188333292802
Dev accuracy: 0.6000000151495138
Testing ...
35/60 [========>....] - ETA: Os
60/60 [======] - 0s 1ms/step
Test loss: 1.244888613621394
Test accuracy: 0.4666666661699613
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cqt(y=y, sr=sr, hop_length=self.hop_length)
    spectral_bandwidth = librosa.feature.spectral_bandwidth(y=y, sr=sr, hop_length=self.hop_length)

splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 26:33] = spectral_bandwidth.T[0:timeseries_length, :]
```

```
Epoch 400/400
35/420 [=>.....] - ETA: 1s - 1oss: 1.0677 - acc: 0.4857
70/420 [====>.....] - ETA: 1s - loss: 1.1393 - acc: 0.5286
105/420 [=====>.....] - ETA: 1s - loss: 1.0420 - acc: 0.5429
140/420 [======>.....] - ETA: 1s - loss: 0.9928 - acc: 0.5643
175/420 [=======>....] - ETA: Os - loss: 0.9586 - acc: 0.5886
210/420 [========>.....] - ETA: 0s - loss: 0.9879 - acc: 0.5810
245/420 [=========>>.....] - ETA: Os - loss: 0.9900 - acc: 0.5878
280/420 [=======>:....] - ETA: Os - loss: 0.9577 - acc: 0.6107
315/420 [========>.....] - ETA: Os - loss: 0.9553 - acc: 0.6063
350/420 [============>.....] - ETA: Os - loss: 0.9623 - acc: 0.6000
420/420 [==================] - 2s 4ms/step - loss: 0.9507 - acc: 0.6024
Validating ...
105/120 [==========>....] - ETA: Os
120/120 [=======] - 0s 3ms/step
Dev loss: 0.9821401908993721
Dev accuracy: 0.6250000136593977
Testing ...
35/60 [======>.....] - ETA: Os
60/60 [======== ] - 0s 904us/step
Test loss: 0.9612920631965002
Test accuracy: 0.6833333472410837
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, sr=sr, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cqt(y=y, sr=sr, hop_length=self.hop_length)
    spectral_contrast = librosa.feature.spectral_contrast(y=y, sr=sr, hop_length=self.hop_length)
   splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)
   data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
   data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
   data[i, :, 14:26] = chroma.T[0:timeseries_length, :]
    data[i, :, 26:33] = spectral_contrast.T[0:timeseries_length, :]
Epoch 400/400
35/420 [=>.....] - ETA: 1s - 1oss: 0.3826 - acc: 0.8286
70/420 [====>.....] - ETA: 1s - loss: 0.3748 - acc: 0.8429
105/420 [=====>.....] - ETA: 1s - 1oss: 0.4167 - acc: 0.8571
140/420 [=======>.....] - ETA: 1s - loss: 0.4069 - acc: 0.8714
175/420 [=======>....] - ETA: Os - loss: 0.4087 - acc: 0.8743
210/420 [=======>....] - ETA: Os - loss: 0.4276 - acc: 0.8619
245/420 [=======>.....] - ETA: Os - loss: 0.4582 - acc: 0.8490
280/420 [=========>.....] - ETA: Os - loss: 0.4495 - acc: 0.8500
315/420 [=======>>.....] - ETA: Os - loss: 0.4462 - acc: 0.8444
350/420 [=========>.....] - ETA: Os - loss: 0.4395 - acc: 0.8400
420/420 [=========] - 2s 4ms/step - 1oss: 0.4264 - acc: 0.8476
Validating ...
35/120 [======>....] - ETA: Os
105/120 [==========>....] - ETA: Os
Dev loss: 0.9936995257933935
Dev accuracy: 0.6750000156462193
Testing ...
35/60 [========>....] - ETA: Os
60/60 [======] - 0s 1ms/step
Test loss: 1.0625159045060475
Test accuracy: 0.6666666865348816
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_bandwidth(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cens(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)

splits = re.split('[',']', file)
    genre = re.split('[',']', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_flatness(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cens(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)

splits = re.split('[']', file)
    genre = re.split('[']', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:14] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 26:33] = rmse.T[0:timeseries_length, :]
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_contrast(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)

splits = re.split('[.]', file)
    genre = re.split('[/]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 20:32] = chroma.T[0:timeseries_length, :]

data[i, :, 32:39] = rmse.T[0:timeseries_length, :]
```

```
Epoch 400/400
35/420 [=>.....] - ETA: 4s - loss: 0.0504 - acc: 0.9714
70/420 [====>.....] - ETA: 4s - loss: 0.0995 - acc: 0.9714
105/420 [=====>.....] - ETA: 3s - loss: 0.1221 - acc: 0.9524
140/420 [=======>.....] - ETA: 3s - loss: 0.0974 - acc: 0.9643
175/420 [======>.....] - ETA: 3s - loss: 0.0836 - acc: 0.9714
210/420 [========>.....] - ETA: 2s - loss: 0.0778 - acc: 0.9762
245/420 [=======>.....] - ETA: 2s - loss: 0.0775 - acc: 0.9714
280/420 [========>.....] - ETA: 1s - loss: 0.0761 - acc: 0.9714
315/420 [=========>.....] - ETA: 1s - loss: 0.0838 - acc: 0.9651
385/420 [===========>...] - ETA: Os - loss: 0.0770 - acc: 0.9662
420/420 [=================== ] - 5s 12ms/step - loss: 0.0786 - acc: 0.9643
Validating ...
35/120 [======>.....] - ETA: 1s
105/120 [============>....] - ETA: Os
120/120 [=======] - 1s 7ms/step
Dev loss: 1.9086458186308544
Dev accuracy: 0.6416666842997074
Testing ...
35/60 [=======>....] - ETA: Os
60/60 [=======] - 0s 5ms/step
Test loss: 2.6331248680750527
Test accuracy: 0.5333333487311999
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_contrast(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    zero_crossing_rate = librosa.feature.zero_crossing_rate(y=y, hop_length=self.hop_length)

splits = re.split('[',']', file)
    genre = re.split('[',']', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 20:32] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 32:39] = zero_crossing_rate.T[0:timeseries_length, :]
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_contrast(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cqt(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)

splits = re.split('[ ]', file)
    genre = re.split('[ ]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 13:20] = spectral_center.T[0:timeseries_length, :]
    data[i, :, 20:32] = chroma.T[0:timeseries_length, :]
    data[i, :, 32:39] = rmse.T[0:timeseries_length, :]
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)
    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_contrast(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_cens(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)

splits = re.split('[.]', file)
    genre = re.split('[.]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 20:32] = chroma.T[0:timeseries_length, :]
    data[i, :, 32:39] = rmse.T[0:timeseries_length, :]
```

```
Epoch 400/400
 35/420 [=>.....] - ETA: 4s - loss: 0.0085 - acc: 1.0000
70/420 [====>.....] - ETA: 4s - loss: 0.0518 - acc: 0.9857
105/420 [=====>.....] - ETA: 3s - loss: 0.0557 - acc: 0.9810
140/420 [======>.....] - ETA: 3s - loss: 0.0604 - acc: 0.9786
175/420 [======>.....] - ETA: 3s - loss: 0.0509 - acc: 0.9829
210/420 [======>.....] - ETA: 2s - loss: 0.0549 - acc: 0.9810
245/420 [=======>.....] - ETA: 2s - loss: 0.0512 - acc: 0.9796
280/420 [=======>.....] - ETA: 1s - 1oss: 0.0522 - acc: 0.9786
315/420 [========>:.....] - ETA: 1s - loss: 0.0473 - acc: 0.9810
350/420 [==========>....] - ETA: 0s - loss: 0.0455 - acc: 0.9800
385/420 [============>...] - ETA: Os - loss: 0.0444 - acc: 0.9818
420/420 [============] - 5s 12ms/step - loss: 0.0445 - acc: 0.9810
Validating ...
35/120 [======>....] - ETA: 1s
70/120 [=======>.....] - ETA: 0s
105/120 [=========>....] - ETA: Os
120/120 [========= ] - 1s 7ms/step
Dev loss: 2.2819462964932122
Dev accuracy: 0.5583333531394601
Testing ...
35/60 [========>....] - ETA: 0s
60/60 [=======] - 0s 5ms/step
Test loss: 2.19755748907725
Test accuracy: 0.5166666830579439
```

```
for i, file in enumerate(list_of_audiofiles):
    y, sr = librosa.load(file)

    mfcc = librosa.feature.mfcc(y=y, sr=sr, hop_length=self.hop_length, n_mfcc=13)
    spectral_center = librosa.feature.spectral_centroid(y=y, hop_length=self.hop_length)
    chroma = librosa.feature.chroma_stft(y=y, sr=sr, hop_length=self.hop_length)
    rmse = librosa.feature.rmse(y=y, hop_length=self.hop_length)
    zero_crossing_rate = librosa.feature.zero_crossing_rate(y=y, hop_length=self.hop_length)
    spectral_bandwidth = librosa.feature.spectral_bandwidth(y=y, hop_length=self.hop_length)

splits = re.split('[./]', splits[1])[3]
    target.append(genre)

data[i, :, 0:13] = mfcc.T[0:timeseries_length, :]
    data[i, :, 20:32] = chroma.T[0:timeseries_length, :]
    data[i, :, 32:39] = rmse.T[0:timeseries_length, :]
    data[i, :, 39:46] = zero_crossing_rate.T[0:timeseries_length, :]
    data[i, :, 46:53] = spectral_bandwidth.T[0:timeseries_length, :]
```