# lab\_music\_partial

April 9, 2019

### 1 Lab: Neural Networks for Music Classification

In addition to the concepts in the MNIST neural network demo, in this lab, you will learn to: \* Load a file from a URL \* Extract simple features from audio samples for machine learning tasks such as speech recognition and classification \* Build a simple neural network for music classification using these features \* Use a callback to store the loss and accuracy history in the training process \* Optimize the learning rate of the neural network

To illustrate the basic concepts, we will look at a relatively simple music classification problem. Given a sample of music, we want to determine which instrument (e.g. trumpet, violin, piano) is playing. This dataset was generously supplied by Prof. Juan Bello at NYU Stenihardt and his former PhD student Eric Humphrey (now at Spotify). They have a complete website dedicated to deep learning methods in music informatics:

http://marl.smusic.nyu.edu/wordpress/projects/feature-learning-deep-architectures/deep-learning-python-tutorial/

You can also check out Juan's course.

### 1.1 Loading Tensorflow

Before starting this lab, you will need to install Tensorflow. If you are using Google colaboratory, Tensorflow is already installed. Run the following command to ensure Tensorflow is installed.

```
In [33]: import tensorflow as tf
   Then, load the other packages.
In [34]: import numpy as np
        import matplotlib
        import matplotlib.pyplot as plt
        %matplotlib inline
```

### 1.2 Audio Feature Extraction with Librosa

The key to audio classification is to extract the correct features. In addition to keras, we will need the librosa package. The librosa package in python has a rich set of methods extracting the features of audio samples commonly used in machine learning tasks such as speech recognition and sound classification.

Installation instructions and complete documentation for the package are given on the librosa main page. On most systems, you should be able to simply use:

```
pip install -u librosa
```

For Unix, you may need to load some additional packages:

```
sudo apt-get install build-essential
sudo apt-get install libxext-dev python-qt4 qt4-dev-tools
pip install librosa
```

After you have installed the package, try to import it.

```
In [35]: import librosa
    import librosa.display
    import librosa.feature
```

In this lab, we will use a set of music samples from the website:

http://theremin.music.uiowa.edu

This website has a great set of samples for audio processing. Look on the web for how to use the requests.get and file.write commands to load the file at the URL provided into your working directory.

You can play the audio sample by copying the file to your local machine and playing it on any media player. If you listen to it you will hear a soprano saxaphone (with vibrato) playing four notes (C, C#, D, Eb).

```
In [36]: import requests
    fn = "SopSax.Vib.pp.C6Eb6.aiff"
    url = "http://theremin.music.uiowa.edu/sound files/MIS/Woodwinds/sopranosaxophone/"+f:

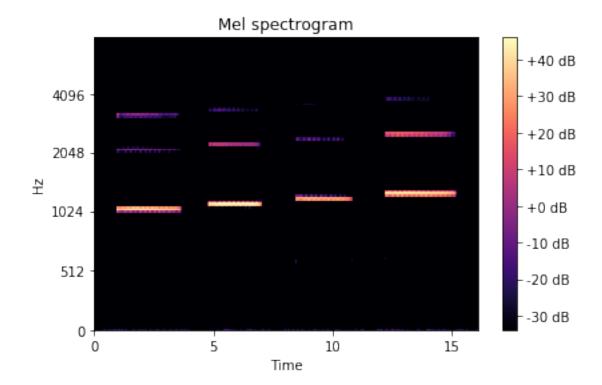
# TODO: Load the file from url and save it in a file under the name fn
    r = requests.get(url)
    if r.status_code == 200:
        with open(fn,"wb") as file:
            file.write(r.content)
            print("success")
    else:
        print(r.status_code)
```

Next, use librosa command librosa.load to read the audio file with filename fn and get the samples y and sample rate sr.

success

Extracting features from audio files is an entire subject on its own right. A commonly used set of features are called the Mel Frequency Cepstral Coefficients (MFCCs). These are derived from the so-called mel spectrogram which is something like a regular spectrogram, but the power and frequency are represented in log scale, which more naturally aligns with human perceptual processing. You can run the code below to display the mel spectrogram from the audio sample.

You can easily see the four notes played in the audio track. You also see the 'harmonics' of each notes, which are other tones at integer multiples of the fundamental frequency of each note.



## 1.3 Downloading the Data

Using the MFCC features described above, Eric Humphrey and Juan Bellow have created a complete data set that can used for instrument classification. Essentially, they collected a number of data files from the website above. For each audio file, the segmented the track into notes and then extracted 120 MFCCs for each note. The goal is to recognize the instrument from the 120 MFCCs. The process of feature extraction is quite involved. So, we will just use their processed data provided at:

https://github.com/marl/dl4mir-tutorial/blob/master/README.md

Note the password. Load the four files into some directory, say instrument\_dataset. Then, load them with the commands.

Looking at the data files: \* What are the number of training and test samples? \* What is the number of features for each sample? \* How many classes (i.e. instruments) are there per class?

Before continuing, you must scale the training and test data, Xtr and Xts. Compute the mean and std deviation of each feature in Xtr and create a new training data set, Xtr\_scale, by subtracting the mean and dividing by the std deviation. Also compute a scaled test data set, Xts\_scale using the mean and std deviation learned from the training data set.

```
In [41]: # TODO Scale the training and test matrices
     from sklearn import preprocessing
     Xtr_mean = np.mean(Xtr, axis = 0)
     Xtr_std = np.std(Xtr, axis = 0)
     Xts_mean = np.mean(Xts, axis = 0)
     Xts_std = np.std(Xts, axis = 0)
     Xtr_scale = (Xtr - Xtr_mean)/ Xtr_std
     Xts_scale = (Xts - Xtr_mean)/ Xtr_std
```

### 1.4 Building a Neural Network Classifier

Following the example in MNIST neural network demo, clear the keras session. Then, create a neural network model with: \* nh=256 hidden units \* sigmoid activation \* select the input and output shapes correctly \* print the model summary

```
In [45]: from tensorflow.keras.models import Model, Sequential from tensorflow.keras.layers import Dense, Activation import tensorflow.keras.backend as K
```

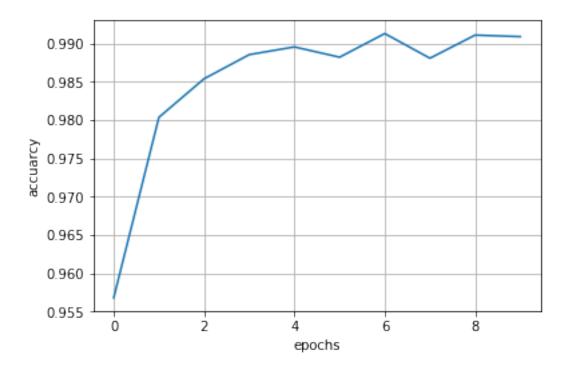
```
In [46]: # TODO clear session
      K.clear_session()
In [47]: # TODO: construct the model
      nin = Xtr.shape[1] # dimension of input data
      nh = 256 # number of hidden units
      nout = int(np.max(ytr)+1) # number of outputs = 10 since there are 10 classes
      model = Sequential()
      model.add(Dense(units=nh, input_shape=(nin,), activation='sigmoid', name='hidden'))
      model.add(Dense(units=nout, activation='sigmoid', name='output'))
      # print("dd")
In [48]: # TODO: Print the model summary
      model.summary()
Layer (type)
             Output Shape
                                        Param #
______
hidden (Dense)
                     (None, 256)
                                         30976
_____
output (Dense) (None, 10)
______
Total params: 33,546
Trainable params: 33,546
Non-trainable params: 0
```

Create an optimizer and compile the model. Select the appropriate loss function and metrics. For the optimizer, use the Adam optimizer with a learning rate of 0.001

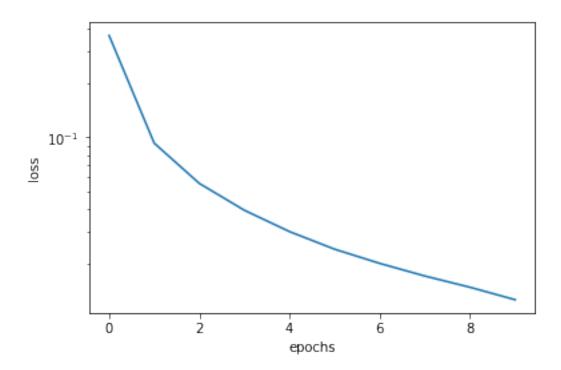
Fit the model for 10 epochs using the scaled data for both the training and validation. Use the validation\_data option to pass the test data. Also, pass the callback class create above. Use a batch size of 100. Your final accuracy should be >99%.

```
Epoch 2/10
Epoch 3/10
Epoch 4/10
Epoch 5/10
Epoch 6/10
Epoch 7/10
Epoch 8/10
Epoch 9/10
Epoch 10/10
```

Plot the validation accuracy saved in hist.history dictionary. This gives one accuracy value per epoch. You should see that the validation accuracy saturates at a little higher than 99%. After that it "bounces around" due to the noise in the stochastic gradient descent.



Plot the loss values saved in the hist.history dictionary. You should see that the loss is steadily decreasing. Use the semilogy plot.



## 1.5 Optimizing the Learning Rate

One challenge in training neural networks is the selection of the learning rate. Rerun the above code, trying four learning rates as shown in the vector rates. For each learning rate: \* clear the session \* construct the network \* select the optimizer. Use the Adam optimizer with the appropriate learning rate. \* train the model for 20 epochs \* save the accuracy and losses

```
In [53]: rate = [0.01,0.001,0.0001]
         batch_size = 100
         loss_hist = []
         accuracy_hist = []
         # TODO
         for lr in rate:
             K.clear_session()
             nin = Xtr.shape[1] # dimension of input data
             nh = 256
                          # number of hidden units
             nout = int(np.max(ytr)+1)
                                          # number of outputs = 10 since there are 10 classes
             model = Sequential()
             model.add(Dense(units=nh, input_shape=(nin,), activation='sigmoid', name='hidden'
             model.add(Dense(units=nout, activation='sigmoid', name='output'))
             opt = optimizers.Adam(lr = lr)
             model.compile(optimizer=opt, loss='sparse_categorical_crossentropy',metrics=['acc'
             hist = model.fit(Xtr_scale, ytr, epochs=20, batch_size=100, validation_data=(Xts_s
             loss_hist.append(hist.history['loss'])
```

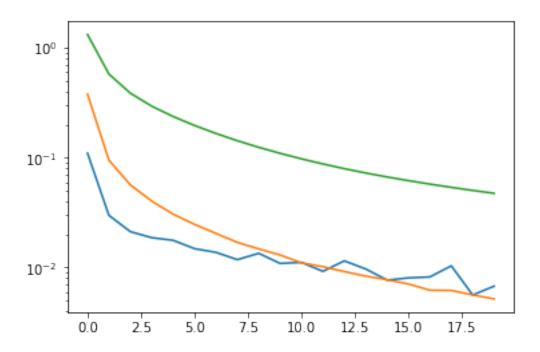
### accuracy\_hist.append(hist.history['val\_acc'])

```
Train on 66247 samples, validate on 14904 samples
Epoch 1/20
Epoch 2/20
Epoch 3/20
Epoch 4/20
Epoch 5/20
Epoch 6/20
Epoch 7/20
Epoch 8/20
Epoch 9/20
Epoch 10/20
Epoch 11/20
Epoch 12/20
Epoch 13/20
Epoch 14/20
Epoch 15/20
Epoch 16/20
Epoch 17/20
Epoch 18/20
Epoch 19/20
Epoch 20/20
Train on 66247 samples, validate on 14904 samples
Epoch 1/20
Epoch 2/20
```

```
Epoch 3/20
Epoch 4/20
Epoch 5/20
Epoch 6/20
Epoch 7/20
Epoch 8/20
Epoch 9/20
Epoch 10/20
Epoch 11/20
Epoch 12/20
Epoch 13/20
Epoch 14/20
Epoch 15/20
Epoch 16/20
Epoch 17/20
Epoch 18/20
Epoch 19/20
Epoch 20/20
Train on 66247 samples, validate on 14904 samples
Epoch 1/20
Epoch 2/20
Epoch 3/20
Epoch 4/20
Epoch 5/20
```

```
Epoch 6/20
Epoch 7/20
Epoch 8/20
Epoch 9/20
Epoch 10/20
Epoch 11/20
Epoch 12/20
Epoch 13/20
Epoch 14/20
Epoch 15/20
Epoch 16/20
Epoch 17/20
Epoch 18/20
Epoch 19/20
Epoch 20/20
```

Plot the loss function vs. the epoch number for all three learning rates on one graph. You should see that the lower learning rates are more stable, but converge slower.



In []:

In []: