Artificial Intelligence Nanodegree

Voice User Interfaces

Project: Speech Recognition with Neural Networks

In this notebook, some template code has already been provided for you, and you will need to implement additional functionality to successfully complete this project. You will not need to modify the included code beyond what is requested. Sections that begin with '(IMPLEMENTATION)' in the header indicate that the following blocks of code will require additional functionality which you must provide. Please be sure to read the instructions carefully!

Note: Once you have completed all of the code implementations, you need to finalize your work by exporting the Jupyter Notebook as an HTML document. Before exporting the notebook to html, all of the code cells need to have been run so that reviewers can see the final implementation and output. You can then export the notebook by using the menu above and navigating to \n",

"**File -> Download as -> HTML (.html)**. Include the finished document along with this notebook as your submission.

In addition to implementing code, there will be questions that you must answer which relate to the project and your implementation. Each section where you will answer a question is preceded by a 'Question X' header. Carefully read each question and provide thorough answers in the following text boxes that begin with 'Answer:'. Your project submission will be evaluated based on your answers to each of the questions and the implementation you provide.

Note: Code and Markdown cells can be executed using the **Shift + Enter** keyboard shortcut. Markdown cells can be edited by double-clicking the cell to enter edit mode.

The rubric contains *optional* "Stand Out Suggestions" for enhancing the project beyond the minimum requirements. If you decide to pursue the "Stand Out Suggestions", you should include the code in this Jupyter notebook.

Introduction

In this notebook, you will build a deep neural network that functions as part of an end-to-end automatic speech recognition (ASR) pipeline! Your completed pipeline will accept raw audio as input and return a predicted transcription of the spoken language. The full pipeline is summarized in the figure below.



- **STEP 1** is a pre-processing step that converts raw audio to one of two feature representations that are commonly used for ASR.
- STEP 2 is an acoustic model which accepts audio features as input and returns a probability distribution over all potential transcriptions. After learning about the basic types of neural networks that are often used for acoustic modeling, you will engage in your own investigations, to design your own acoustic model!
- **STEP 3** in the pipeline takes the output from the acoustic model and returns a predicted transcription.

Feel free to use the links below to navigate the notebook:

- The Data
- STEP 1: Acoustic Features for Speech Recognition
- STEP 2: Deep Neural Networks for Acoustic Modeling
 - Model 0: RNN
 - Model 1: RNN + TimeDistributed Dense
 - Model 2: CNN + RNN + TimeDistributed Dense
 - Model 3: Deeper RNN + TimeDistributed Dense
 - Model 4: Bidirectional RNN + TimeDistributed Dense
 - Models 5+
 - Compare the Models
 - Final Model
- STEP 3: Obtain Predictions

The Data

We begin by investigating the dataset that will be used to train and evaluate your pipeline. LibriSpeech is a large corpus of English-read speech, designed for training and evaluating models for ASR. The dataset contains 1000 hours of speech derived from audiobooks. We will work with a small subset in this project, since larger-scale data would take a long while to train. However, after completing this project, if you are interested in exploring further, you are encouraged to work with more of the data that is provided online.

In the code cells below, you will use the vis_train_features module to visualize a training example. The supplied argument index=0 tells the module to extract the first example in the training set. (You are welcome to change index=0 to point to a

different training example, if you like, but please **DO NOT** amend any other code in the cell.) The returned variables are:

- vis_text transcribed text (label) for the training example.
- vis_raw_audio raw audio waveform for the training example.
- vis_mfcc_feature mel-frequency cepstral coefficients (MFCCs) for the training example.
- vis_spectrogram_feature spectrogram for the training example.
- vis_audio_path the file path to the training example.

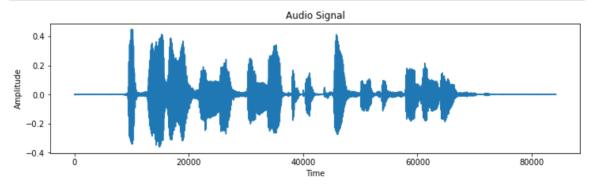
There are 2023 total training examples.

```
In [2]: from data_generator import vis_train_features
# extract label and audio features for a single training example
vis_text, vis_raw_audio, vis_mfcc_feature, vis_spectrogram_feature, vis_a
```

The following code cell visualizes the audio waveform for your chosen example, along with the corresponding transcript. You also have the option to play the audio in the notebook!

```
In [3]: from IPython.display import Markdown, display
    from data_generator import vis_train_features, plot_raw_audio
    from IPython.display import Audio
    %matplotlib inline

# plot audio signal
    plot_raw_audio(vis_raw_audio)
    # print length of audio signal
    display(Markdown('**Shape of Audio Signal**: ' + str(vis_raw_audio.shape
    # print transcript corresponding to audio clip
    display(Markdown('**Transcript**: ' + str(vis_text)))
# play the audio file
    Audio(vis_audio_path)
```



Shape of Audio Signal: (84231,)

Transcript: her father is a most remarkable person to say the least

STEP 1: Acoustic Features for Speech Recognition

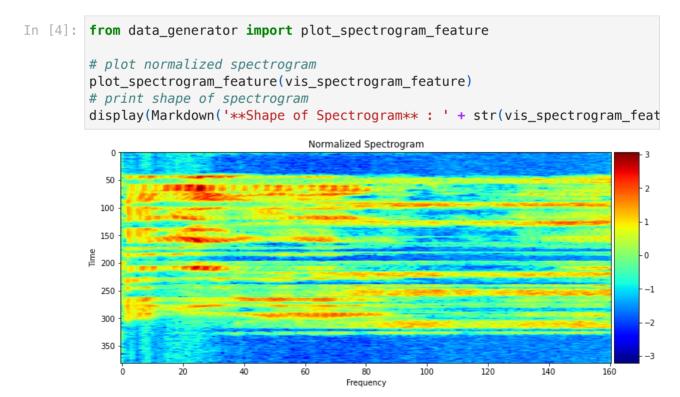
For this project, you won't use the raw audio waveform as input to your model. Instead, we provide code that first performs a pre-processing step to convert the raw audio to a feature representation that has historically proven successful for ASR models. Your acoustic model will accept the feature representation as input.

In this project, you will explore two possible feature representations. *After completing the project*, if you'd like to read more about deep learning architectures that can accept raw audio input, you are encouraged to explore this research paper.

Spectrograms

The first option for an audio feature representation is the spectrogram. In order to complete this project, you will **not** need to dig deeply into the details of how a spectrogram is calculated; but, if you are curious, the code for calculating the spectrogram was borrowed from this repository. The implementation appears in the utils.py file in your repository.

The code that we give you returns the spectrogram as a 2D tensor, where the first (*vertical*) dimension indexes time, and the second (*horizontal*) dimension indexes frequency. To speed the convergence of your algorithm, we have also normalized the spectrogram. (You can see this quickly in the visualization below by noting that the mean value hovers around zero, and most entries in the tensor assume values close to zero.)



Shape of Spectrogram: (381, 161)

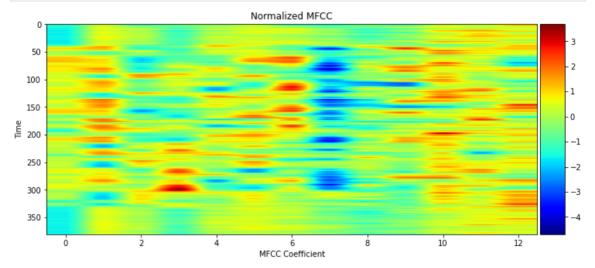
Mel-Frequency Cepstral Coefficients (MFCCs)

The second option for an audio feature representation is MFCCs. You do **not** need to dig deeply into the details of how MFCCs are calculated, but if you would like more information, you are welcome to peruse the documentation of the python_speech_features Python package. Just as with the spectrogram features, the MFCCs are normalized in the supplied code.

The main idea behind MFCC features is the same as spectrogram features: at each time window, the MFCC feature yields a feature vector that characterizes the sound within the window. Note that the MFCC feature is much lower-dimensional than the spectrogram feature, which could help an acoustic model to avoid overfitting to the training dataset.

In [5]: from data_generator import plot_mfcc_feature

plot normalized MFCC
plot_mfcc_feature(vis_mfcc_feature)
print shape of MFCC
display(Markdown('**Shape of MFCC** : ' + str(vis_mfcc_feature.shape)))



Shape of MFCC: (381, 13)

When you construct your pipeline, you will be able to choose to use either spectrogram or MFCC features. If you would like to see different implementations that make use of MFCCs and/or spectrograms, please check out the links below:

- This repository uses spectrograms.
- This repository uses MFCCs.
- This repository also uses MFCCs.
- This repository experiments with raw audio, spectrograms, and MFCCs as features.

STEP 2: Deep Neural Networks for Acoustic Modeling

In this section, you will experiment with various neural network architectures for acoustic modeling.

You will begin by training five relatively simple architectures. **Model 0** is provided for you. You will write code to implement **Models 1**, **2**, **3**, and **4**. If you would like to experiment further, you are welcome to create and train more models under the **Models 5+** heading.

All models will be specified in the sample_models.py file. After importing the sample_models module, you will train your architectures in the notebook.

After experimenting with the five simple architectures, you will have the opportunity to compare their performance. Based on your findings, you will construct a deeper architecture that is designed to outperform all of the shallow models.

For your convenience, we have designed the notebook so that each model can be specified and trained on separate occasions. That is, say you decide to take a break from the notebook after training Model 1. Then, you need not re-execute all prior code cells in the notebook before training Model 2. You need only re-execute the code cell below, that is marked with RUN THIS CODE CELL IF YOU ARE RESUMING THE NOTEBOOK AFTER A BREAK, before transitioning to the code cells corresponding to Model 2.

```
# RUN THIS CODE CELL IF YOU ARE RESUMING THE NOTEBOOK AFTER A BREAK #
      # allocate 50% of GPU memory (if you like, feel free to change this)
      from keras.backend.tensorflow_backend import set_session
      import tensorflow as tf
      config = tf.ConfigProto()
      config.gpu_options.per_process_gpu_memory_fraction = 0.5
      set_session(tf.Session(config=config))
      # watch for any changes in the sample_models module, and reload it automa
      %load ext autoreload
      %autoreload 2
      # import NN architectures for speech recognition
      from sample_models import *
      # import function for training acoustic model
      from train_utils import train_model
```

WARNING:tensorflow:Deprecation warnings have been disabled. Set TF_ENABL E_DEPRECATION_WARNINGS=1 to re-enable them.

Using TensorFlow backend.

Model 0: RNN

Given their effectiveness in modeling sequential data, the first acoustic model you will use is an RNN. As shown in the figure below, the RNN we supply to you will take the time sequence of audio features as input.



At each time step, the speaker pronounces one of 28 possible characters, including each of the 26 letters in the English alphabet, along with a space character (" "), and an apostrophe (').

The output of the RNN at each time step is a vector of probabilities with 29 entries, where the \$i\$-th entry encodes the probability that the \$i\$-th character is spoken in the time sequence. (The extra 29th character is an empty "character" used to pad training examples within batches containing uneven lengths.) If you would like to peek under the hood at how characters are mapped to indices in the probability vector, look at the character used to pad training examples within batches containing uneven lengths.) If you would like to peek under the hood at how characters are mapped to indices in the probability vector, look at the char_map.py file in the repository. The figure below shows an equivalent, rolled depiction of the RNN that shows the output layer in greater detail.



The model has already been specified for you in Keras. To import it, you need only run the code cell below.

In [7]: model_0 = simple_rnn_model(input_dim=161) # change to 13 if you would lik

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:508: The name tf.placeholder is deprecated. P lease use tf.compat.v1.placeholder instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:68: The name tf.get_default_graph is deprecated. Please use tf.compat.v1.get_default_graph instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3837: The name tf.random_uniform is deprecate d. Please use tf.random.uniform instead.

Layer (type)	Output Shape	Param #
the_input (InputLayer)	(None, None, 161)	0
rnn (GRU)	(None, None, 29)	16617
softmax (Activation)	(None, None, 29)	0

Total params: 16,617 Trainable params: 16,617 Non-trainable params: 0

None

As explored in the lesson, you will train the acoustic model with the CTC loss criterion. Custom loss functions take a bit of hacking in Keras, and so we have

implemented the CTC loss function for you, so that you can focus on trying out as many deep learning architectures as possible:). If you'd like to peek at the implementation details, look at the add_ctc_loss function within the

train_utils.py file in the repository.

To train your architecture, you will use the train_model function within the train_utils module; it has already been imported in one of the above code cells. The train_model function takes three required arguments:

- input_to_softmax a Keras model instance.
- pickle_path the name of the pickle file where the loss history will be saved.
- save_model_path the name of the HDF5 file where the model will be saved.

If we have already supplied values for input_to_softmax , pickle_path , and save_model_path , please **DO NOT** modify these values.

There are several **optional** arguments that allow you to have more control over the training process. You are welcome to, but not required to, supply your own values for these arguments.

- minibatch_size the size of the minibatches that are generated while training the model (default: 20).
- spectrogram Boolean value dictating whether spectrogram (True) or MFCC (False) features are used for training (default: True).
- mfcc_dim the size of the feature dimension to use when generating MFCC features (default: 13).
- optimizer the Keras optimizer used to train the model (default: SGD(lr=0.02, decay=1e-6, momentum=0.9, nesterov=True, clipnorm=5)).
- epochs the number of epochs to use to train the model (default: 20). If you choose to modify this parameter, make sure that it is at least 20.
- verbose controls the verbosity of the training output in the model.fit generator method (default: 1).
- sort_by_duration Boolean value dictating whether the training and validation sets are sorted by (increasing) duration before the start of the first epoch (default: False).

The train_model function defaults to using spectrogram features; if you choose to use these features, note that the acoustic model in simple_rnn_model should have input_dim=161. Otherwise, if you choose to use MFCC features, the acoustic model should have input dim=13.

We have chosen to use <code>GRU</code> units in the supplied RNN. If you would like to experiment with <code>LSTM</code> or <code>SimpleRNN</code> cells, feel free to do so here. If you change the <code>GRU</code> units to <code>SimpleRNN</code> cells in <code>simple_rnn_model</code>, you may notice that the loss quickly becomes undefined (<code>nan</code>) - you are strongly encouraged to check this for yourself! This is due to the <code>exploding gradients problem</code>. We have already implemented <code>gradient clipping</code> in your optimizer to help you avoid this issue.

IMPORTANT NOTE: If you notice that your gradient has exploded in any of the models below, feel free to explore more with gradient clipping (the clipnorm argument in your optimizer) or swap out any SimpleRNN cells for LSTM or GRU cells. You can also try restarting the kernel to restart the training process.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3952: The name tf.log is deprecated. Please use tf.math.log instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/opt imizers.py:757: The name tf.train.Optimizer is deprecated. Please use t f.compat.v1.train.Optimizer instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:977: The name tf.assign_add is deprecated. Pl ease use tf.compat.v1.assign_add instead.

WARNING:tensorflow:Variable *= will be deprecated. Use `var.assign(var * other)` if you want assignment to the variable value or `x = x * y` if y ou want a new python Tensor object.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:964: The name tf.assign is deprecated. Please use tf.compat.v1.assign instead.

Epoch 1/20

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:168: The name tf.get_default_session is depre cated. Please use tf.compat.v1.get_default_session instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:184: The name tf.global_variables is deprecated. Please use tf.compat.v1.global_variables instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:193: The name tf.is_variable_initialized is deprecated. Please use tf.compat.v1.is_variable_initialized instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:200: The name tf.variables_initializer is deprecated. Please use tf.compat.v1.variables_initializer instead.

```
660 - val loss: 757.5558
Epoch 2/20
- val loss: 761.6699
Epoch 3/20
- val loss: 757.3842
Epoch 4/20
101/101 [=============== ] - 102s 1s/step - loss: 778.5655
- val_loss: 753.2747
Epoch 5/20
101/101 [============ ] - 101s 1s/step - loss: 777.2697
- val loss: 759.3686
Epoch 6/20
- val_loss: 753.4225
Epoch 7/20
- val loss: 760.3146
Epoch 8/20
- val loss: 757.5974
Epoch 9/20
```

```
- val_loss: 753.5028
Epoch 10/20
462 - val loss: 757.9185
Epoch 11/20
5004 - val_loss: 754.6709
Epoch 12/20
- val loss: 759.6416
Epoch 13/20
101/101 [============= ] - 101s 1s/step - loss: 777.5231
- val_loss: 755.8973
Epoch 14/20
101/101 [============ ] - 103s 1s/step - loss: 778.1245
- val_loss: 753.8410
Epoch 15/20
- val_loss: 748.6742
Epoch 16/20
- val loss: 753.8970
Epoch 17/20
- val loss: 756.9101
Epoch 18/20
411 - val loss: 749.8924
Epoch 19/20
101/101 [============ ] - 102s 1s/step - loss: 777.3247
- val_loss: 756.4204
Epoch 20/20
- val loss: 758.4202
```

(IMPLEMENTATION) Model 1: RNN + TimeDistributed Dense

Read about the TimeDistributed wrapper and the BatchNormalization layer in the Keras documentation. For your next architecture, you will add batch normalization to the recurrent layer to reduce training times. The TimeDistributed layer will be used to find more complex patterns in the dataset. The unrolled snapshot of the architecture is depicted below.



The next figure shows an equivalent, rolled depiction of the RNN that shows the (TimeDistrbuted) dense and output layers in greater detail.



Use your research to complete the rnn_model function within the sample_models.py file. The function should specify an architecture that satisfies the following requirements:

- The first layer of the neural network should be an RNN (SimpleRNN, LSTM, or GRU) that takes the time sequence of audio features as input. We have added GRU units for you, but feel free to change GRU to SimpleRNN or LSTM, if you like!
- Whereas the architecture in simple_rnn_model treated the RNN output as the final layer of the model, you will use the output of your RNN as a hidden layer. Use TimeDistributed to apply a Dense layer to each of the time steps in the RNN output. Ensure that each Dense layer has output_dim units.

Use the code cell below to load your model into the <code>model_1</code> variable. Use a value for <code>input_dim</code> that matches your chosen audio features, and feel free to change the values for <code>units</code> and <code>activation</code> to tweak the behavior of your recurrent layer.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:508: The name tf.placeholder is deprecated. P lease use tf.compat.v1.placeholder instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:68: The name tf.get_default_graph is deprecated. Please use tf.compat.v1.get_default_graph instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3837: The name tf.random_uniform is deprecate d. Please use tf.random.uniform instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:127: The name tf.placeholder_with_default is deprecated. Please use tf.compat.v1.placeholder_with_default instead.

Layer (type)	Output Shape	<u> </u>	Param #
the_input (InputLayer)	(None, None,	161)	0
rnn (GRU)	(None, None,	200)	217200
bn_rnn_1d (BatchNormalizatio	(None, None,	200)	800
time_distributed_1 (TimeDist	(None, None,	29)	5829
softmax (Activation)	(None, None,	29)	0

Total params: 223,829 Trainable params: 223,429 Non-trainable params: 400

None

Please execute the code cell below to train the neural network you specified in input_to_softmax . After the model has finished training, the model is saved in

the HDF5 file model_1.h5 . The loss history is saved in model_1.pickle . You are welcome to tweak any of the optional parameters while calling the train_model function, but this is not required.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3952: The name tf.log is deprecated. Please use tf.math.log instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/opt imizers.py:757: The name tf.train.Optimizer is deprecated. Please use t f.compat.v1.train.Optimizer instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:977: The name tf.assign_add is deprecated. Pl ease use tf.compat.v1.assign add instead.

WARNING:tensorflow:Variable *= will be deprecated. Use `var.assign(var * other)` if you want assignment to the variable value or `x = x * y` if y ou want a new python Tensor object.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:964: The name tf.assign is deprecated. Please use tf.compat.v1.assign instead.

Epoch 1/20

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:168: The name tf.get_default_session is depre cated. Please use tf.compat.v1.get_default_session instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:184: The name tf.global_variables is deprecated. Please use tf.compat.v1.global_variables instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:193: The name tf.is_variable_initialized is deprecated. Please use tf.compat.v1.is_variable_initialized instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:200: The name tf.variables_initializer is deprecated. Please use tf.compat.v1.variables_initializer instead.

```
- val loss: 234.6396
Epoch 2/20
- val loss: 212.3043
Epoch 3/20
- val loss: 183.2065
Epoch 4/20
101/101 [=============== ] - 104s 1s/step - loss: 171.9545
- val_loss: 166.0209
Epoch 5/20
- val loss: 163.2990
Epoch 6/20
101/101 [============ ] - 104s 1s/step - loss: 153.6530
- val_loss: 155.9617
Epoch 7/20
- val loss: 152.7398
Epoch 8/20
- val_loss: 149.0897
Epoch 9/20
```

```
101/101 [============= ] - 104s 1s/step - loss: 138.9864
- val_loss: 147.5790
Epoch 10/20
- val loss: 147.5971
Epoch 11/20
- val loss: 144.3826
Epoch 12/20
- val loss: 143.0201
Epoch 13/20
- val_loss: 146.3248
Epoch 14/20
101/101 [============ ] - 103s 1s/step - loss: 126.6580
- val_loss: 138.8765
Epoch 15/20
- val_loss: 143.2524
Epoch 16/20
- val_loss: 142.2260
Epoch 17/20
- val loss: 139.2108
Epoch 18/20
- val loss: 141.7481
Epoch 19/20
- val_loss: 140.3511
Epoch 20/20
- val loss: 141.7228
```

(IMPLEMENTATION) Model 2: CNN + RNN + TimeDistributed Dense

The architecture in cnn_rnn_model adds an additional level of complexity, by introducing a 1D convolution layer.



This layer incorporates many arguments that can be (optionally) tuned when calling the <code>cnn_rnn_model</code> module. We provide sample starting parameters, which you might find useful if you choose to use spectrogram audio features.

If you instead want to use MFCC features, these arguments will have to be tuned.

Note that the current architecture only supports values of 'same' or 'valid' for the conv_border_mode argument.

When tuning the parameters, be careful not to choose settings that make the convolutional layer overly small. If the temporal length of the CNN layer is shorter than the length of the transcribed text label, your code will throw an error.

Before running the code cell below, you must modify the <code>cnn_rnn_model</code> function in <code>sample_models.py</code>. Please add batch normalization to the recurrent layer, and provide the same <code>TimeDistributed</code> layer as before.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:508: The name tf.placeholder is deprecated. P lease use tf.compat.v1.placeholder instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3837: The name tf.random_uniform is deprecate d. Please use tf.random.uniform instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:125: The name tf.get_default_graph is deprecated. Please use tf.compat.v1.get_default_graph instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:127: The name tf.placeholder_with_default is deprecated. Please use tf.compat.v1.placeholder_with_default instead.

Layer (type)	Output	Shape		Param #
the_input (InputLayer)	(None,	None,	161)	0
conv1d (Conv1D)	(None,	None,	200)	354400
bn_conv_1d (BatchNormalizati	(None,	None,	200)	800
rnn (SimpleRNN)	(None,	None,	200)	80200
bn_rnn_1d (BatchNormalizatio	(None,	None,	200)	800
time_distributed_1 (TimeDist	(None,	None,	29)	5829
softmax (Activation)	(None,	None,	29)	0

Total params: 442,029 Trainable params: 441,229 Non-trainable params: 800

None

/usr/local/lib/python3.8/dist-packages/keras/layers/recurrent.py:989: Us erWarning: The `implementation` argument in `SimpleRNN` has been depreca ted. Please remove it from your layer call.

warnings.warn('The `implementation` argument '

Please execute the code cell below to train the neural network you specified in input_to_softmax . After the model has finished training, the model is saved in the HDF5 file model_2.h5 . The loss history is saved in model_2.pickle . You

are welcome to tweak any of the optional parameters while calling the train_model function, but this is not required.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3952: The name tf.log is deprecated. Please use tf.math.log instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/opt imizers.py:757: The name tf.train.Optimizer is deprecated. Please use t f.compat.v1.train.Optimizer instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:977: The name tf.assign_add is deprecated. Pl ease use tf.compat.v1.assign add instead.

WARNING:tensorflow:Variable *= will be deprecated. Use `var.assign(var * other)` if you want assignment to the variable value or `x = x * y` if y ou want a new python Tensor object.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:964: The name tf.assign is deprecated. Please use tf.compat.v1.assign instead.

Epoch 1/20

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:168: The name tf.get_default_session is depre cated. Please use tf.compat.v1.get_default_session instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:184: The name tf.global_variables is deprecated. Please use tf.compat.v1.global_variables instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:193: The name tf.is_variable_initialized is deprecated. Please use tf.compat.v1.is_variable_initialized instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:200: The name tf.variables_initializer is deprecated. Please use tf.compat.v1.variables_initializer instead.

```
51 - val loss: 199.5135
Epoch 2/20
79 - val loss: 163.2025
Epoch 3/20
28 - val loss: 150.6068
Epoch 4/20
05 - val_loss: 147.0995
Epoch 5/20
101/101 [============= ] - 23s 232ms/step - loss: 132.69
81 - val loss: 141.3714
Epoch 6/20
27 - val_loss: 138.6319
Epoch 7/20
00 - val_loss: 135.5060
Epoch 8/20
02 - val_loss: 136.3418
Epoch 9/20
```

```
14 - val_loss: 138.8069
Epoch 10/20
92 - val loss: 137.1206
Epoch 11/20
73 - val loss: 134.9496
Epoch 12/20
98 - val loss: 135.0715
Epoch 13/20
5 - val_loss: 137.3323
Epoch 14/20
9 - val_loss: 138.2602
Epoch 15/20
101/101 [============ ] - 23s 227ms/step - loss: 93.007
1 - val_loss: 141.1420
Epoch 16/20
7 - val_loss: 139.2737
Epoch 17/20
3 - val loss: 141.3206
Epoch 18/20
4 - val loss: 141.8945
Epoch 19/20
3 - val_loss: 142.9868
Epoch 20/20
8 - val loss: 149.2594
```

(IMPLEMENTATION) Model 3: Deeper RNN + TimeDistributed Dense

Review the code in <code>rnn_model</code>, which makes use of a single recurrent layer. Now, specify an architecture in <code>deep_rnn_model</code> that utilizes a variable number <code>recur_layers</code> of recurrent layers. The figure below shows the architecture that should be returned if <code>recur_layers=2</code>. In the figure, the output sequence of the first recurrent layer is used as input for the next recurrent layer.



Feel free to change the supplied values of units to whatever you think performs best. You can change the value of recur_layers, as long as your final value is greater than 1. (As a quick check that you have implemented the additional functionality in deep_rnn_model correctly, make sure that the architecture that you specify here is identical to rnn_model if recur_layers=1.)

```
recur_layers=2)
```

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:68: The name tf.get_default_graph is deprecated. Please use tf.compat.v1.get_default_graph instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3837: The name tf.random_uniform is deprecate d. Please use tf.random.uniform instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:127: The name tf.placeholder_with_default is deprecated. Please use tf.compat.v1.placeholder_with_default instead.

Layer (type)	Output	Shape		Param #
the_input (InputLayer)	(None,	None,	161)	0
rnn_0 (GRU)	(None,	None,	200)	217200
bn_rnn_1d_0 (BatchNormalizat	(None,	None,	200)	800
rnn_1 (GRU)	(None,	None,	200)	240600
bn_rnn_1d_1 (BatchNormalizat	(None,	None,	200)	800
time_distributed_1 (TimeDist	(None,	None,	29)	5829
softmax (Activation)	(None,	None,	29)	0

Total params: 465,229 Trainable params: 464,429 Non-trainable params: 800

None

Please execute the code cell below to train the neural network you specified in input_to_softmax. After the model has finished training, the model is saved in the HDF5 file model_3.h5. The loss history is saved in model_3.pickle. You are welcome to tweak any of the optional parameters while calling the train_model function, but this is not required.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3952: The name tf.log is deprecated. Please use tf.math.log instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/opt imizers.py:757: The name tf.train.Optimizer is deprecated. Please use t f.compat.v1.train.Optimizer instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:977: The name tf.assign_add is deprecated. Pl ease use tf.compat.v1.assign add instead.

WARNING:tensorflow:Variable *= will be deprecated. Use `var.assign(var * other)` if you want assignment to the variable value or `x = x * y` if y ou want a new python Tensor object.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:964: The name tf.assign is deprecated. Please use tf.compat.v1.assign instead.

Epoch 1/20

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:168: The name tf.get_default_session is depre cated. Please use tf.compat.v1.get_default_session instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:184: The name tf.global_variables is deprecated. Please use tf.compat.v1.global_variables instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:193: The name tf.is_variable_initialized is deprecated. Please use tf.compat.v1.is_variable_initialized instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:200: The name tf.variables_initializer is deprecated. Please use tf.compat.v1.variables_initializer instead.

```
- val loss: 234.3806
Epoch 2/20
- val loss: 209.7589
Epoch 3/20
- val loss: 182.6410
Epoch 4/20
- val_loss: 165.4866
Epoch 5/20
- val loss: 155.3851
Epoch 6/20
- val_loss: 143.9949
Epoch 7/20
- val loss: 143.6280
Epoch 8/20
- val_loss: 139.0685
Epoch 9/20
```

```
- val_loss: 140.2469
Epoch 10/20
- val loss: 133.7471
Epoch 11/20
- val loss: 133.7968
Epoch 12/20
- val loss: 131.8538
Epoch 13/20
- val_loss: 136.9488
Epoch 14/20
- val_loss: 131.5040
Epoch 15/20
- val_loss: 136.2721
Epoch 16/20
- val_loss: 128.7077
Epoch 17/20
- val loss: 131.0174
Epoch 18/20
- val loss: 132.4113
Epoch 19/20
- val_loss: 130.8649
Epoch 20/20
- val loss: 131.1669
```

(IMPLEMENTATION) Model 4: Bidirectional RNN + TimeDistributed Dense

Read about the Bidirectional wrapper in the Keras documentation. For your next architecture, you will specify an architecture that uses a single bidirectional RNN layer, before a (TimeDistributed) dense layer. The added value of a bidirectional RNN is described well in this paper.

One shortcoming of conventional RNNs is that they are only able to make use of previous context. In speech recognition, where whole utterances are transcribed at once, there is no reason not to exploit future context as well. Bidirectional RNNs (BRNNs) do this by processing the data in both directions with two separate hidden layers which are then fed forwards to the same output layer.



Before running the code cell below, you must complete the bidirectional_rnn_model function in sample_models.py . Feel free to use

SimpleRNN, LSTM, or GRU units. When specifying the Bidirectional wrapper, use merge_mode='concat'.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:488: The name tf.placeholder is deprecated. P lease use tf.compat.v1.placeholder instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:63: The name tf.get_default_graph is deprecated. Please use tf.compat.v1.get_default_graph instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3626: The name tf.random_uniform is deprecate d. Please use tf.random.uniform instead.

Layer (type)	Output Shape		Param #
the_input (InputLayer)	(None, None,	161)	0
bidirectional_1 (Bidirection	(None, None,	400)	434400
time_distributed_1 (TimeDist	(None, None,	29)	11629
softmax (Activation)	(None, None,	29)	0

Total params: 446,029 Trainable params: 446,029 Non-trainable params: 0

None

Please execute the code cell below to train the neural network you specified in input_to_softmax. After the model has finished training, the model is saved in the HDF5 file model_4.h5. The loss history is saved in model_4.pickle. You are welcome to tweak any of the optional parameters while calling the train_model function, but this is not required.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3741: The name tf.log is deprecated. Please use tf.math.log instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/opt imizers.py:711: The name tf.train.Optimizer is deprecated. Please use t f.compat.v1.train.Optimizer instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:949: The name tf.assign_add is deprecated. Pl ease use tf.compat.v1.assign add instead.

WARNING:tensorflow:Variable *= will be deprecated. Use `var.assign(var * other)` if you want assignment to the variable value or `x = x * y` if y ou want a new python Tensor object.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:936: The name tf.assign is deprecated. Please use tf.compat.v1.assign instead.

Epoch 1/20

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:158: The name tf.get_default_session is depre cated. Please use tf.compat.v1.get_default_session instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:172: The name tf.global_variables is deprecated. Please use tf.compat.v1.global_variables instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:180: The name tf.is_variable_initialized is deprecated. Please use tf.compat.v1.is_variable_initialized instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:187: The name tf.variables_initializer is deprecated. Please use tf.compat.v1.variables_initializer instead.

```
- val loss: 231.5425
Epoch 2/20
- val loss: 200.4790
Epoch 3/20
- val loss: 195.0213
Epoch 4/20
101/101 [=============== ] - 199s 2s/step - loss: 190.4837
- val_loss: 185.3153
Epoch 5/20
- val loss: 178.3045
Epoch 6/20
- val_loss: 175.4351
Epoch 7/20
- val loss: 164.5784
Epoch 8/20
- val_loss: 166.1030
Epoch 9/20
```

```
- val_loss: 159.0771
Epoch 10/20
- val loss: 159.2844
Epoch 11/20
- val loss: 154.6488
Epoch 12/20
- val loss: 154.7036
Epoch 13/20
- val_loss: 150.8596
Epoch 14/20
- val_loss: 148.5861
Epoch 15/20
- val_loss: 148.4838
Epoch 16/20
- val_loss: 143.8347
Epoch 17/20
- val loss: 148.9774
Epoch 18/20
- val loss: 144.9319
Epoch 19/20
- val loss: 144.7219
Epoch 20/20
- val loss: 144.6623
```

(OPTIONAL IMPLEMENTATION) Models 5+

If you would like to try out more architectures than the ones above, please use the code cell below. Please continue to follow the same convention for saving the models; for the \$i\$-th sample model, please save the loss at **model_i.pickle** and saving the trained model at **model_i.h5**.

```
In []: ## (Optional) TODO: Try out some more models!
### Feel free to use as many code cells as needed.
```

Compare the Models

Execute the code cell below to evaluate the performance of the drafted deep learning models. The training and validation loss are plotted for each model.

```
In [8]: from glob import glob
import numpy as np
import _pickle as pickle
```

```
import seaborn as sns
          import matplotlib.pyplot as plt
          %matplotlib inline
          sns.set_style(style='white')
          # obtain the paths for the saved model history
          all_pickles = sorted(glob("results/*.pickle"))
          # extract the name of each model
          model_names = [item[8:-7] for item in all_pickles]
          # extract the loss history for each model
          valid_loss = [pickle.load( open( i, "rb" ) )['val_loss'] for i in all_pic
          train_loss = [pickle.load( open( i, "rb" ) )['loss'] for i in all_pickles
          # save the number of epochs used to train each model
          num_epochs = [len(valid_loss[i]) for i in range(len(valid_loss))]
          fig = plt.figure(figsize=(16,5))
          # plot the training loss vs. epoch for each model
          ax1 = fig.add_subplot(121)
          for i in range(len(all_pickles)):
              ax1.plot(np.linspace(1, num_epochs[i], num_epochs[i]),
                      train loss[i], label=model names[i])
          # clean up the plot
          ax1.legend()
          ax1.set_xlim([1, max(num_epochs)])
          plt.xlabel('Epoch')
          plt.ylabel('Training Loss')
          # plot the validation loss vs. epoch for each model
          ax2 = fig.add_subplot(122)
          for i in range(len(all_pickles)):
              ax2.plot(np.linspace(1, num_epochs[i], num_epochs[i]),
                      valid loss[i], label=model names[i])
          # clean up the plot
          ax2.legend()
          ax2.set_xlim([1, max(num_epochs)])
          plt.xlabel('Epoch')
          plt.ylabel('Validation Loss')
          plt.show()
                                                   700
           700
                                                   600
           600
                                                 Loss
                                                  500
                                                                                 model_1
                                          model_1
           500
                                         model 2
                                                                                 model 2
                                                                                 model 3
                                                   400
          400
                                                                                 model 4
                                                   300
           200
           100
         from glob import glob
In [11]:
          import numpy as np
          import _pickle as pickle
          import seaborn as sns
          import matplotlib.pyplot as plt
          %matplotlib inline
          sns.set_style(style='white')
```

```
# obtain the paths for the saved model history
all_pickles = sorted(glob("results/*.pickle"))
# extract the name of each model
model_names = [item[8:-7] for item in all_pickles]
# extract the loss history for each model
valid_loss = [pickle.load( open( i, "rb" ) )['val_loss'] for i in all_pic
train_loss = [pickle.load( open( i, "rb" ) )['loss'] for i in all_pickles
# save the number of epochs used to train each model
num_epochs = [len(valid_loss[i]) for i in range(len(valid_loss))]
fig = plt.figure(figsize=(16,5))
# plot the training loss vs. epoch for each model
ax1 = fig.add_subplot(121)
for i in range(len(all_pickles)):
    ax1.plot(np.linspace(1, num_epochs[i], num_epochs[i]),
             train_loss[i], label=model_names[i])
# clean up the plot
ax1.legend()
ax1.set_xlim([1, max(num_epochs)])
plt.xlabel('Epoch')
plt.ylabel('Training Loss')
plt.ylim(50, 300)
# plot the validation loss vs. epoch for each model
ax2 = fig.add_subplot(122)
for i in range(len(all pickles)):
    ax2.plot(np.linspace(1, num_epochs[i], num_epochs[i]),
             valid_loss[i], label=model_names[i])
# clean up the plot
ax2.legend()
ax2.set_xlim([1, max(num_epochs)])
plt.xlabel('Epoch')
plt.ylabel('Validation Loss')
plt.ylim(100, 270)
plt.show()
                                model_1
                                                                        model 1
                               model 2
                                                                       model 2
                                         240
 250
                                model 3
                                                                        model 3
                                         220
S 200
                                        -055
                                         200
                                         180
 150
                                         160
 100
                                         120
                                         100
                               17.5
                      12.5
                                                 5.0
                                                          10.0
```

Question 1: Use the plot above to analyze the performance of each of the attempted architectures. Which performs best? Provide an explanation regarding why you think some models perform better than others.

Answer: Since Model 0 (RNN) tends to underfit, it can be seen that it is not a sufficient model to learn the features of the data. On the other hand, all other models show an overfit tendency because the valid loss starts to increase for training data up to 20 epochs. In particular, Model 2 (CNN + RNN + TimeDistributed Dense) shows

a clear overfit tendency and records the minimum loss at the 7th epoch. All models except model 0 are likely to record less loss with appropriate regularization.

(IMPLEMENTATION) Final Model

Now that you've tried out many sample models, use what you've learned to draft your own architecture! While your final acoustic model should not be identical to any of the architectures explored above, you are welcome to merely combine the explored layers above into a deeper architecture. It is **NOT** necessary to include new layer types that were not explored in the notebook.

However, if you would like some ideas for even more layer types, check out these ideas for some additional, optional extensions to your model:

- If you notice your model is overfitting to the training dataset, consider adding dropout! To add dropout to recurrent layers, pay special attention to the dropout_W and dropout_U arguments. This paper may also provide some interesting theoretical background.
- If you choose to include a convolutional layer in your model, you may get better results by working with **dilated convolutions**. If you choose to use dilated convolutions, make sure that you are able to accurately calculate the length of the acoustic model's output in the model.output_length lambda function. You can read more about dilated convolutions in Google's WaveNet paper. For an example of a speech-to-text system that makes use of dilated convolutions, check out this GitHub repository. You can work with dilated convolutions in Keras by paying special attention to the padding argument when you specify a convolutional layer.
- If your model makes use of convolutional layers, why not also experiment with adding **max pooling**? Check out this paper for example architecture that makes use of max pooling in an acoustic model.
- So far, you have experimented with a single bidirectional RNN layer. Consider stacking the bidirectional layers, to produce a deep bidirectional RNN!

All models that you specify in this repository should have output_length defined as an attribute. This attribute is a lambda function that maps the (temporal) length of the input acoustic features to the (temporal) length of the output softmax layer. This function is used in the computation of CTC loss; to see this, look at the add_ctc_loss function in train_utils.py. To see where the output_length attribute is defined for the models in the code, take a look at the sample_models.py file. You will notice this line of code within most models:

```
model.output_length = lambda x: x
```

The acoustic model that incorporates a convolutional layer (cnn_rnn_model) has a line that is a bit different:

In the case of models that use purely recurrent layers, the lambda function is the identity function, as the recurrent layers do not modify the (temporal) length of their input tensors. However, convolutional layers are more complicated and require a specialized function (cnn_output_length in sample_models.py) to determine the temporal length of their output.

You will have to add the output_length attribute to your final model before running the code cell below. Feel free to use the cnn_output_length function, if it suits your model.

```
# RUN THIS CODE CELL IF YOU ARE RESUMING THE NOTEBOOK AFTER A BREAK #
      # allocate 50% of GPU memory (if you like, feel free to change this)
      from keras.backend.tensorflow backend import set session
      import tensorflow as tf
      config = tf.ConfigProto()
      config.gpu_options.per_process_gpu_memory_fraction = 0.5
      set_session(tf.Session(config=config))
      # watch for any changes in the sample models module, and reload it automa
      %load ext autoreload
      %autoreload 2
      # import NN architectures for speech recognition
      from sample models import *
      # import function for training acoustic model
      from train_utils import train_model
```

WARNING:tensorflow:Deprecation warnings have been disabled. Set TF_ENABL E_DEPRECATION_WARNINGS=1 to re-enable them.

Using TensorFlow backend.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:488: The name tf.placeholder is deprecated. P lease use tf.compat.v1.placeholder instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3626: The name tf.random_uniform is deprecate d. Please use tf.random.uniform instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:117: The name tf.get_default_graph is deprecated. Please use tf.compat.v1.get_default_graph instead.

/home/sample_models.py:154: UserWarning: Update your `GRU` call to the K eras 2 API: `GRU(200, activation="relu", return_sequences=True, name="rn n_0", dropout=0.5, recurrent_dropout=0.5)`
 GRU(units, activation='relu', return_sequences=True, name='rnn_' + str (ith),
/home/sample_models.py:154: UserWarning: Update your `GRU` call to the K eras 2 API: `GRU(200, activation="relu", return_sequences=True, name="rn n_1", dropout=0.5, recurrent_dropout=0.5)`
 GRU(units, activation='relu', return_sequences=True, name='rnn_' + str (ith),

Layer (type)	0utput	Shape		Param #
the_input (InputLayer)	(None,	None,	161)	0
conv1d (Conv1D)	(None,	None,	200)	354400
bn_conv_1d (BatchNormalizati	(None,	None,	200)	800
bidirectional_1 (Bidirection	(None,	None,	400)	481200
bn_rnn_1d_0 (BatchNormalizat	(None,	None,	400)	1600
bidirectional_2 (Bidirection	(None,	None,	400)	721200
bn_rnn_1d_1 (BatchNormalizat	(None,	None,	400)	1600
time_distributed_1 (TimeDist	(None,	None,	29)	11629
softmax (Activation)	(None,	None,	29)	0

Total params: 1,572,429
Trainable params: 1,570,429
Non-trainable params: 2,000

None

Please execute the code cell below to train the neural network you specified in input_to_softmax. After the model has finished training, the model is saved in the HDF5 file model_end.h5. The loss history is saved in model_end.pickle. You are welcome to tweak any of the optional parameters while calling the train_model function, but this is not required.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:3741: The name tf.log is deprecated. Please use tf.math.log instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/opt imizers.py:711: The name tf.train.Optimizer is deprecated. Please use t f.compat.v1.train.Optimizer instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:949: The name tf.assign_add is deprecated. Pl ease use tf.compat.v1.assign add instead.

WARNING:tensorflow:Variable *= will be deprecated. Use `var.assign(var * other)` if you want assignment to the variable value or `x = x * y` if y ou want a new python Tensor object.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:936: The name tf.assign is deprecated. Please use tf.compat.v1.assign instead.

Epoch 1/20

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:158: The name tf.get_default_session is depre cated. Please use tf.compat.v1.get_default_session instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:172: The name tf.global_variables is deprecated. Please use tf.compat.v1.global_variables instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:180: The name tf.is_variable_initialized is deprecated. Please use tf.compat.v1.is_variable_initialized instead.

WARNING:tensorflow:From /usr/local/lib/python3.8/dist-packages/keras/backend/tensorflow_backend.py:187: The name tf.variables_initializer is deprecated. Please use tf.compat.v1.variables_initializer instead.

```
- val loss: 247.6694
Epoch 2/20
- val loss: 223.8520
Epoch 3/20
- val loss: 208.5439
Epoch 4/20
101/101 [============== ] - 232s 2s/step - loss: 204.7624
- val_loss: 187.1865
Epoch 5/20
- val loss: 179.9056
Epoch 6/20
- val_loss: 166.6414
Epoch 7/20
- val loss: 152.8091
Epoch 8/20
- val_loss: 145.5870
Epoch 9/20
```

```
- val_loss: 142.2745
Epoch 10/20
- val loss: 139.2816
Epoch 11/20
- val loss: 135.7911
Epoch 12/20
- val loss: 134.8015
Epoch 13/20
- val_loss: 130.6167
Epoch 14/20
101/101 [============= ] - 233s 2s/step - loss: 140.2559
- val_loss: 130.0604
Epoch 15/20
- val_loss: 126.6415
Epoch 16/20
- val_loss: 124.7123
Epoch 17/20
- val loss: 124.3876
Epoch 18/20
- val loss: 121.8541
Epoch 19/20
- val_loss: 120.8134
Epoch 20/20
- val loss: 120.9789
```

Question 2: Describe your final model architecture and your reasoning at each step.

Answer: My final model consists of a multi-layer CNN, a bidirectional RNN, and a **TimeDistributedDNN**. As can be easily predicted in the first training, at the 4th epoch, the valid loss stopped monotonically decreasing and gradually increased overfit. The minimum valid loss was 127.0196 in the 7th epoch.

It effectively alleviated the overfit by adding dropout using the dropout of the recurrent layer. If 0.5 is given for both the dropout_U and dropout_W parameters, the minimum value of the valid loss is 121.3858, and it is recorded at the 20th epoch.

```
In [4]: from glob import glob
   import numpy as np
   import _pickle as pickle
   import seaborn as sns
   import matplotlib.pyplot as plt
%matplotlib inline
   sns.set_style(style='white')
# obtain the paths for the saved model history
```

```
all_pickles = sorted(glob("results/*.pickle"))
# extract the name of each model
model_names = [item[8:-7] for item in all_pickles]
# extract the loss history for each model
valid_loss = [pickle.load( open( i, "rb" ) )['val_loss'] for i in all_pic
train_loss = [pickle.load( open( i, "rb" ) )['loss'] for i in all pickles
# save the number of epochs used to train each model
num_epochs = [len(valid_loss[i]) for i in range(len(valid_loss))]
fig = plt.figure(figsize=(16.5))
# plot the training loss vs. epoch for each model
ax1 = fig.add_subplot(121)
for i in range(len(all_pickles)):
    ax1.plot(np.linspace(1, num_epochs[i], num_epochs[i]),
             train_loss[i], label=model_names[i])
# clean up the plot
ax1.legend()
ax1.set_xlim([1, max(num_epochs)])
plt.xlabel('Epoch')
plt.ylabel('Training Loss')
plt.ylim(0, 300)
# plot the validation loss vs. epoch for each model
ax2 = fig.add_subplot(122)
for i in range(len(all_pickles)):
    ax2.plot(np.linspace(1, num_epochs[i], num_epochs[i]),
             valid_loss[i], label=model_names[i])
# clean up the plot
ax2.legend()
ax2.set_xlim([1, max(num_epochs)])
plt.xlabel('Epoch')
plt.ylabel('Validation Loss')
plt.ylim(100, 270)
plt.show()
                                          260
                               model 3
                                                                        model_1
                               model 2
                                                                        model_2
                                          240
                              model_3
                                                                       model 3
                               model 4
                                          220
                                                                        model 4
 200
                               model end
                                                                        model end
                                        055
                                          200
150
                                          180
 100
                                         140
 50
                                          120
                                          100
                                                           Epoch
```

STEP 3: Obtain Predictions

We have written a function for you to decode the predictions of your acoustic model. To use the function, please execute the code cell below.

```
In [5]: import numpy as np
  from data_generator import AudioGenerator
  from keras import backend as K
  from utils import int_sequence_to_text
```

```
from IPython.display import Audio
def get_predictions(index, partition, input_to_softmax, model_path):
    """ Print a model's decoded predictions
        index (int): The example you would like to visualize
        partition (str): One of 'train' or 'validation'
        input to softmax (Model): The acoustic model
        model_path (str): Path to saved acoustic model's weights
    # load the train and test data
    data gen = AudioGenerator()
    data_gen.load_train_data()
    data_gen.load_validation_data()
    # obtain the true transcription and the audio features
    if partition == 'validation':
        transcr = data_gen.valid_texts[index]
        audio_path = data_gen.valid_audio_paths[index]
        data_point = data_gen.normalize(data_gen.featurize(audio_path))
    elif partition == 'train':
        transcr = data gen.train texts[index]
        audio_path = data_gen.train_audio_paths[index]
        data point = data gen.normalize(data gen.featurize(audio path))
    else:
        raise Exception('Invalid partition! Must be "train" or "validati
    # obtain and decode the acoustic model's predictions
    input_to_softmax.load_weights(os.path.join('results', model_path))
    prediction = input_to_softmax.predict(np.expand_dims(data_point, axis
    output_length = [input_to_softmax.output_length(data_point.shape[0])]
    pred_ints = (K.eval(K.ctc_decode(
                prediction, output length)[0][0])+1).flatten().tolist()
    # play the audio file, and display the true and predicted transcripti
    print('-'*80)
    Audio(audio_path)
    print('True transcription:\n' + '\n' + transcr)
    print('-'*80)
    print('Predicted transcription:\n' + '\n' + ''.join(int_sequence_to_t
    print('-'*80)
```

Use the code cell below to obtain the transcription predicted by your final model for the first example in the training dataset.

/home/sample_models.py:154: UserWarning: Update your `GRU` call to the K eras 2 API: `GRU(200, activation="relu", return_sequences=True, name="rn n_0", dropout=0.5, recurrent_dropout=0.5)`

GRU(units, activation='relu', return_sequences=True, name='rnn_' + str (ith),

/home/sample_models.py:154: UserWarning: Update your `GRU` call to the K eras 2 API: `GRU(200, activation="relu", return_sequences=True, name="rn n_1", dropout=0.5, recurrent_dropout=0.5)`

GRU(units, activation='relu', return_sequences=True, name='rnn_' + str (ith),

Layer (type)	Output	Shape		Param #
the_input (InputLayer)	(None,	None,	161)	0
conv1d (Conv1D)	(None,	None,	200)	354400
bn_conv_1d (BatchNormalizati	(None,	None,	200)	800
bidirectional_3 (Bidirection	(None,	None,	400)	481200
bn_rnn_1d_0 (BatchNormalizat	(None,	None,	400)	1600
bidirectional_4 (Bidirection	(None,	None,	400)	721200
bn_rnn_1d_1 (BatchNormalizat	(None,	None,	400)	1600
time_distributed_2 (TimeDist	(None,	None,	29)	11629
softmax (Activation)	(None,	None,	29)	0
T . 1 4 570 400				

Total params: 1,572,429
Trainable params: 1,570,429
Non-trainable params: 2,000

None

True transcription:

her father is a most remarkable person to say the least

Predicted transcription:

e fotthers am mos fr mor a bo porsin to sathe lest

Use the next code cell to visualize the model's prediction for the first example in the validation dataset.

Layer (type)	Output	Shape		Param #
the_input (InputLayer)	(None,	None,	161)	0
conv1d (Conv1D)	(None,	None,	200)	354400
bn_conv_1d (BatchNormalizati	(None,	None,	200)	800
bidirectional_5 (Bidirection	(None,	None,	400)	481200
bn_rnn_1d_0 (BatchNormalizat	(None,	None,	400)	1600
bidirectional_6 (Bidirection	(None,	None,	400)	721200
bn_rnn_1d_1 (BatchNormalizat	(None,	None,	400)	1600
time_distributed_3 (TimeDist	(None,	None,	29)	11629
softmax (Activation)	(None,	None,	29)	0
Total params: 1,572,429 Trainable params: 1,570,429 Non-trainable params: 2,000				
None				
True transcription:				
the bogus legislature numbered thirty six members				

Predicted transcription:

the bel lhis or dis latur nomver tery ses memers

One standard way to improve the results of the decoder is to incorporate a language model. We won't pursue this in the notebook, but you are welcome to do so as an optional extension.

If you are interested in creating models that provide improved transcriptions, you are encouraged to download more data and train bigger, deeper models. But beware - the model will likely take a long while to train. For instance, training this state-of-the-art model would take 3-6 weeks on a single GPU!