Speech autoencoder using deep fully-connected networks

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Implementation Summary

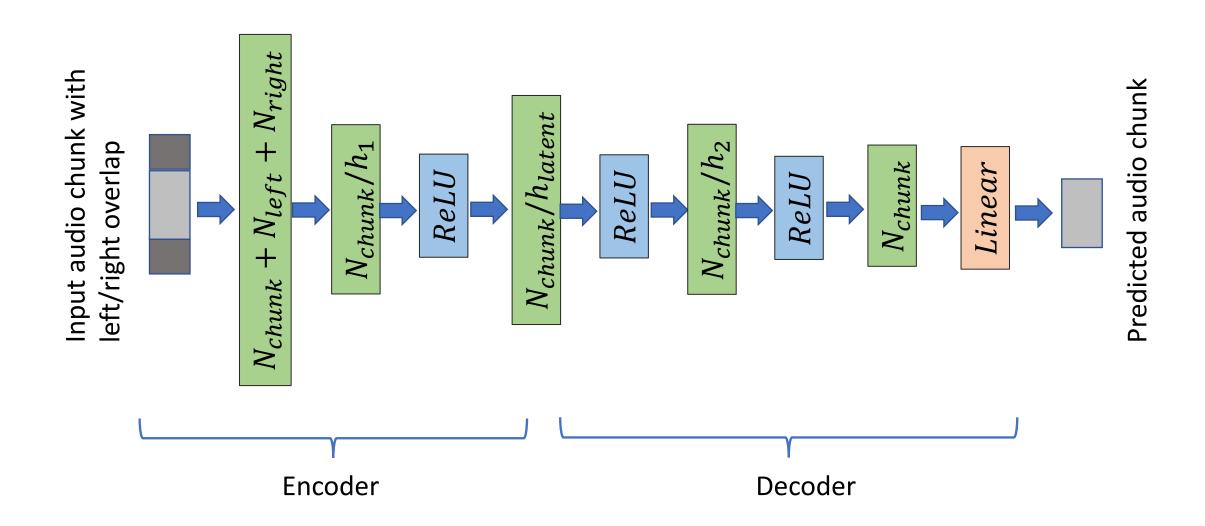
- Vanilla autoencoder built using Keras/TensorFlow
- Custom pipeline for pre-processing and on-the-fly batch generation
- Specialized utility functions for training on LibriSpeech corpuses
- Mini-batches generated by traversing the raw audio signal for each speaker and breaking it into chunks of a specified size which become the feature vectors.
- Overlap allowed on input chunks as a means for incorporating temporal structure into the autoencoder. The corresponding non-overlapping output chunks are used as labels.

Pre-processing Workflow

build_speech_dict.py

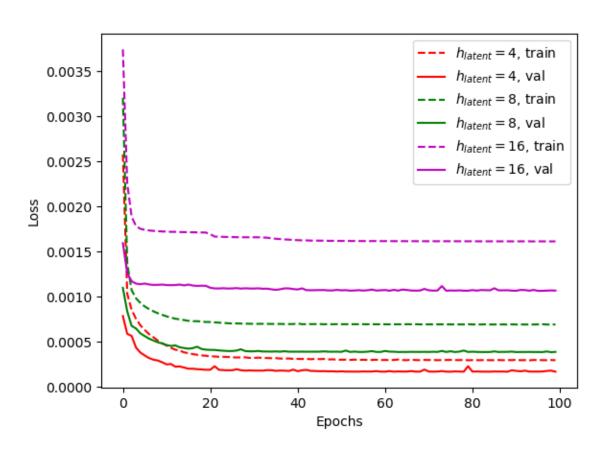
- Generates a "speech_dict" data structure from the corpus
 - Reflects the internal structure of the LibriSpeech corpus. Useful for traversing the speakers, chapters, and utterances.
- Concatenates each speaker's utterances into a master 1D numpy array for the speaker. Writes array to disk.
 - The sequence of the utterances is preserved in the array.
 - Speaker array used for fast on-the-fly batch generation

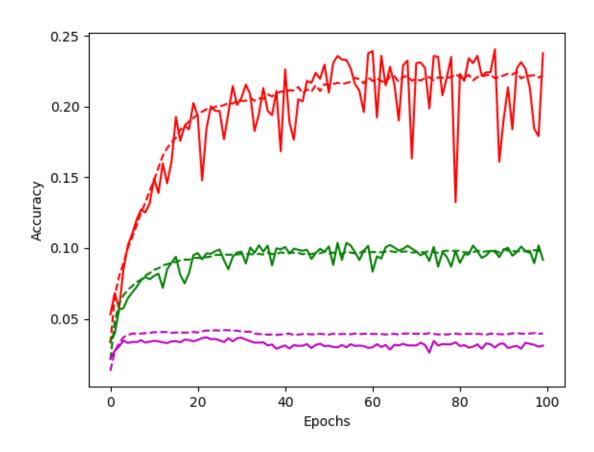
Example Network Architecture



Training Experiments

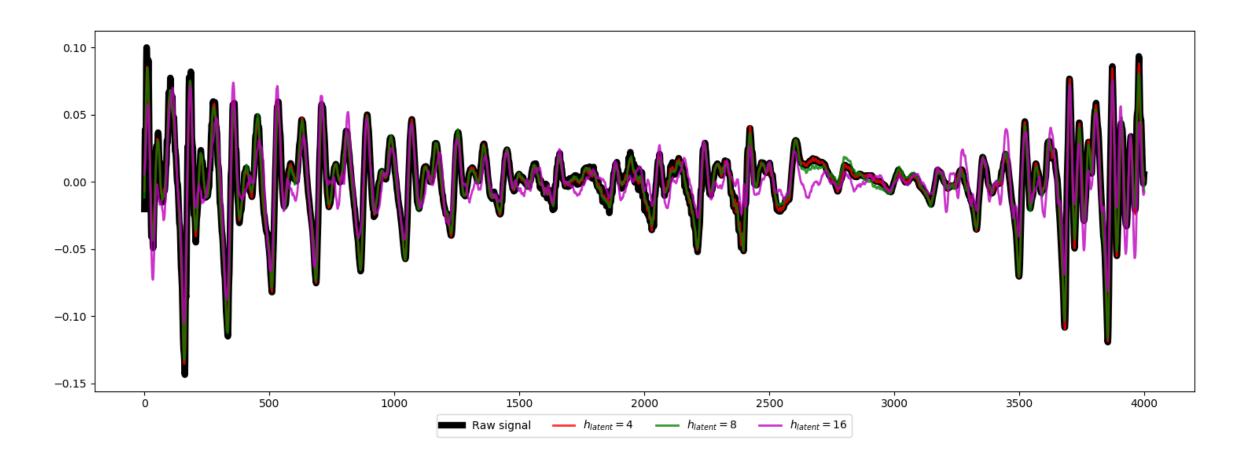
- dev-clean dataset (40 speakers, 5.1 hrs total, 294M samples @16kHz)
- Tried chunk size, batch size from Chorowski, et al. (2019)
 - $N_{chunk} = 5120 \ (\Delta t_{chunk} = 320 ms, 16 \text{kHz}), \text{ batch size} = 64$
 - This chunk size is too large for the vanilla autoencoder and training fails with a non-decreasing loss function
- Reducing chunk size and increasing batch size proved successful
 - $N_{chunk} = 800 \ (\Delta t_{chunk} = 50 ms, \ 16 \text{kHz}), \ \text{batch size} = 128$
- Training parameters:
 - 100 epochs, learning rate = 1e-4, Adam optimizer, MSE loss

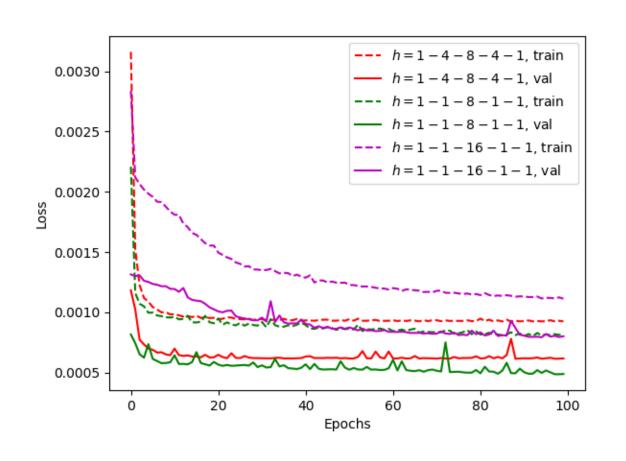


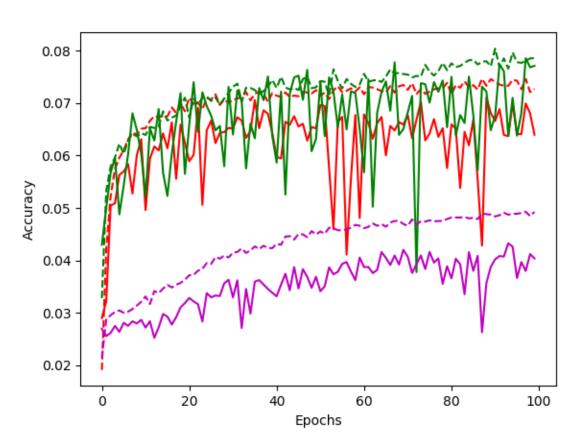


$$N_{chunk} o rac{N_{chunk}}{h_{latent}} o N_{chunk}$$

Speaker 2803 250ms audio sample (5 chunks)

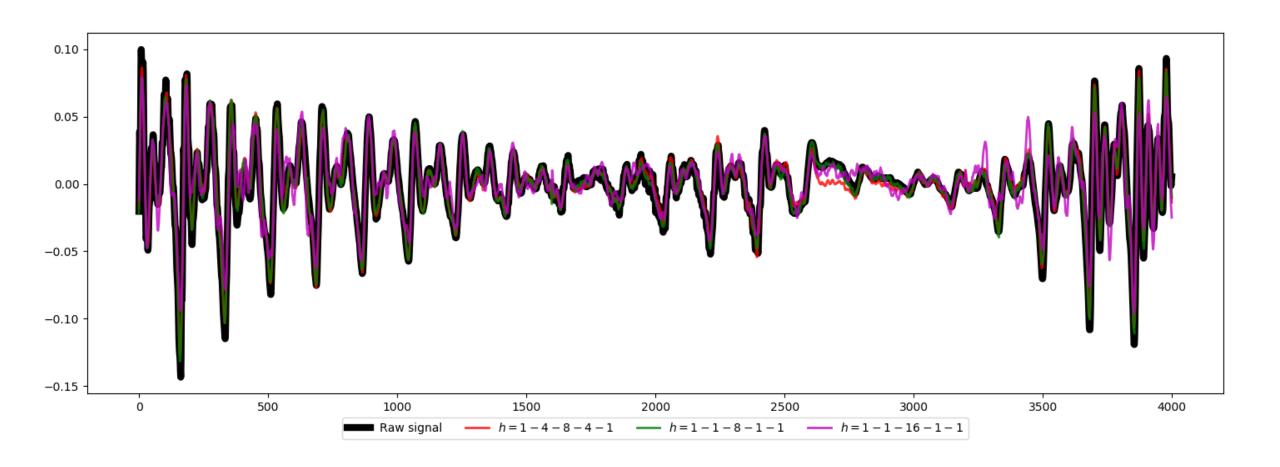




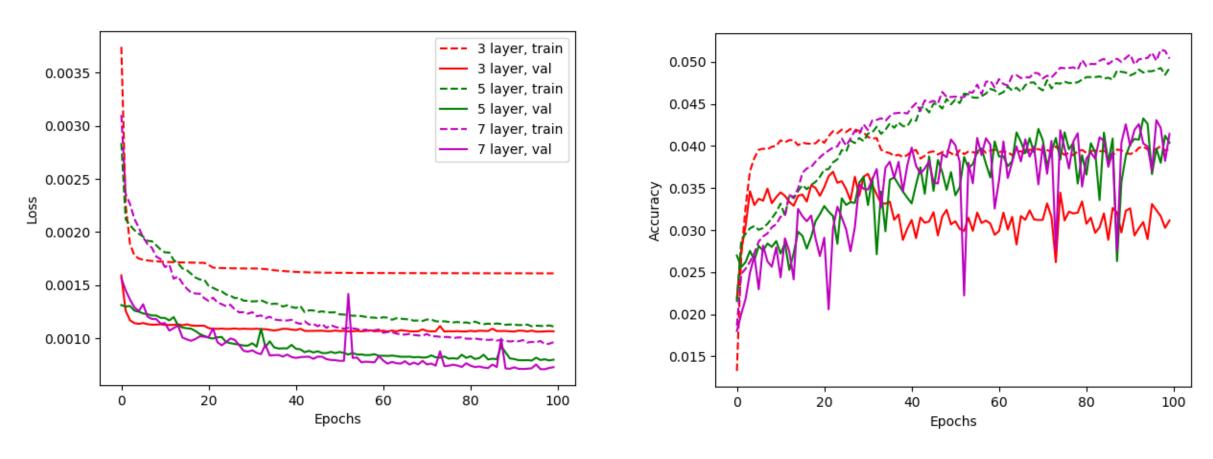


$$N_{chunk} o rac{N_{chunk}}{h_1} o rac{N_{chunk}}{h_{latent}} o rac{N_{chunk}}{h_1} o N_{chunk}$$

Speaker 2803 250ms audio sample (5 chunks)



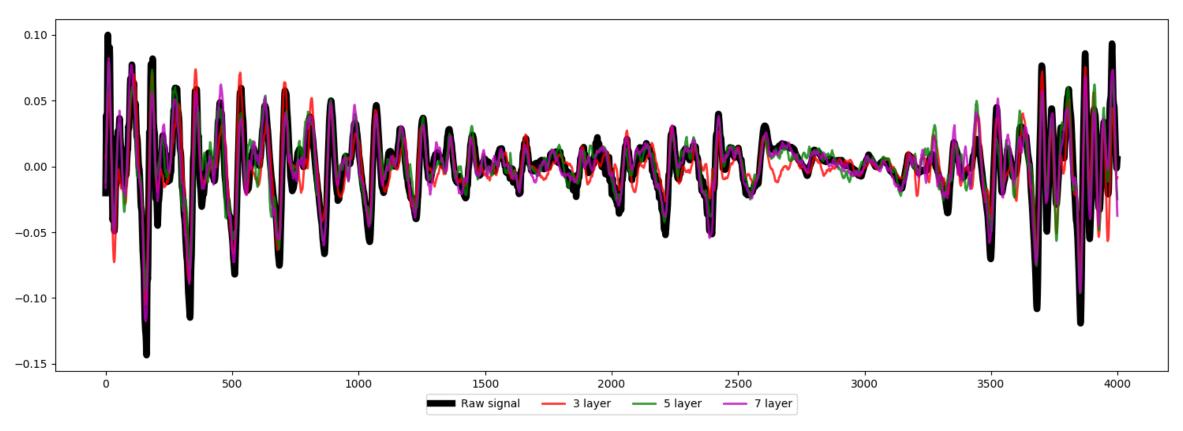
Effect of network depth



Symmetric architectures, h_{latent} =16, size reduction only in the latent layer

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