

# A TRANSFER LEARNING APPROACH FOR PRONUNCIATION SCORING

Marcelo Sancinetti<sup>2</sup>, Jazmín Vidal<sup>1,2\*</sup>, Cyntia Bonomi<sup>2</sup>, Luciana Ferrer<sup>1</sup>

<sup>1</sup> Laboratorio de Inteligencia Artificial Aplicada, Instituto de Ciencias de la Computación, UBA-CONICET, Argentina <sup>2</sup> Departamento de Computación, Universidad de Buenos Aires, Argentina

\*jvidal@dc.uba.ar



### PRONUNCIATION SCORING

### Given a phrase uttered by a language learner, return a pronunciation quality score for each phone.

- Challenging task with room for improvement.
- Standard systems use models trained for automatic speech recognition (ASR) with native data only.
- Better performance using systems trained specifically for the task using native data.
- Datasets labelled for the task are scarce and usually small.

### NATIVE DATA

- Rely on ASR technology to generate native models.
- Measures similarity between student's speech and native sounding speech.

#### NATIVE + NONNATIVE DATA

- Use non-native data with pronunciation quality labels.
- Directly trained to distinguish correctly from incorrectly pronounced segments.
- Variety of input features and classifiers.

#### TRANSFER LEARNING

- DNNs for pronunciation scoring show improvements over traditional methods of both groups
- Rely on transfer learning to mitigate data scarcity

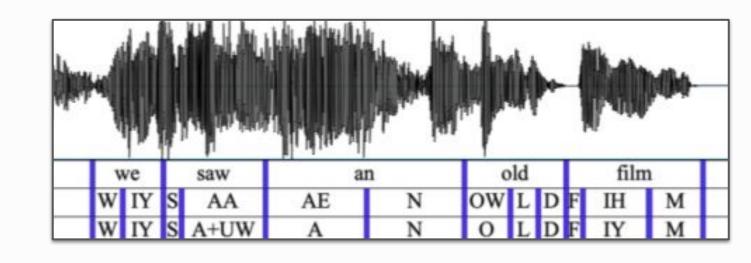
### CONTRIBUTIONS

- Finetune the ASR model to the task of pronunciation scoring.
- Explore 2 different fine-tuning approaches and 6 design choices.
- Propose a loss function that compensates for inherent imbalance across phones and classes present in pronunciation scoring datasets.
- Measure performance using an alternative cost function designed to encourage low false correction rates.
- Share dataset and code to replicate the results at:

https://github.com/MarceloSancinetti/epa-gop-pykaldi

# DATABASE

- 3200 nonnative English phrases by 50 speakers from Argentina.
- Manually annotated at detailed phonetic level using ARPAbet symbols.
- Correctly- and incorrectly-pronounced labels are assigned to each of the target phones determined by the forced-alignment system



### BASELINE METHOD: GOP

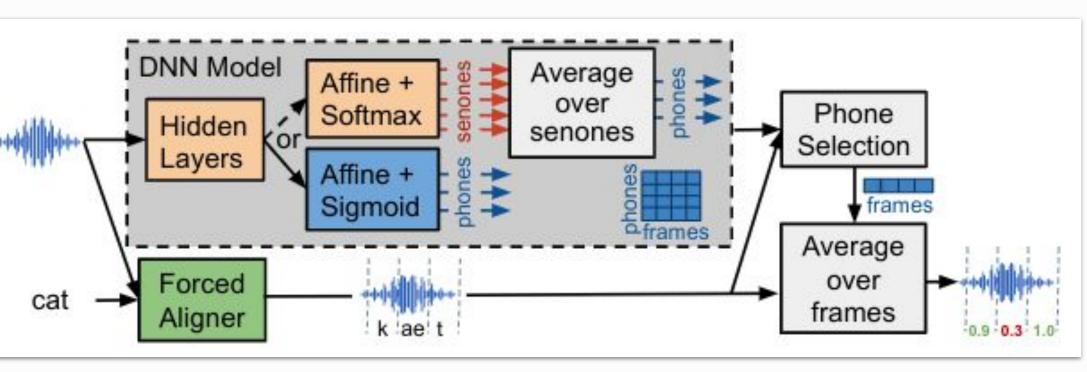
- GOP scores: for each phone, the averaged posterior probability of the target phone for each frame.
- Computed using the outputs of a senone acoustic model.

$$GOP(p) = -\frac{1}{D} \sum_{t=T}^{T+D-1} \log P_t(p|O)$$

 Start and end frames are obtained from forced alignment.

- Official Kaldi recipe reproduced in Pykaldi
- Features: 40-dimensional MFCCs + I-vectors.
- Acoustic model: TDNN-F trained on Librispeech (960 hours) (decoding and forced alignment)
- 17 layers + output layer of size 6024 senones + softmax

#### **TOP BRANCH - GOP**



**BOTTOM BRANCH- GOP-FT** 

### PROPOSED METHOD: GOP-FT

- Replace baseline output layer with a layer that predicts per phone per frame probability of correctly pronounced
- GOP-FT scores: for each frame the probability of being correctly pronounced for the target phone in that frame. Then average over the frames.
- Weighted cross-entropy loss:

$$L = -\sum_{p \in P} \sum_{y \in Y} w_{py} \sum_{t \in T_{py}} y_t \log \hat{y}_t + (1 - y_t) \log (1 - \hat{y}_t)$$
 w<sub>py</sub> adjust the influence of the samples from each phone and class.

Flat Weights: Zero Weight: Balanced  $W_{DV} = 1/N_{DV}$  $W_{py} = 0$ 

- Features: 40-dimensional MFCCs + I-vectors.
- Acoustic model: Kaldi's TDNN-F trained on Librispeech (960 hours) ported to Pytorch (decoding and forced alignment)
- 17 layers + affine output layer of size 39 + sigmoid

# **EXPERIMENTS**

LayO: only the new output layer • BN: batch-normalization in the is trained, keeping all other parameters frozen at their pre-trained values.

• LayO+1: the last hidden layer is

also trained.

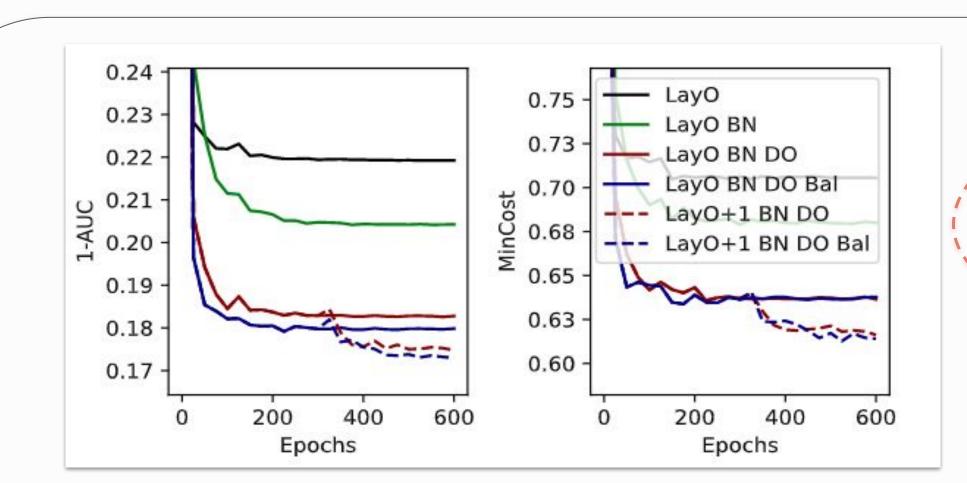
- output layer.
- **DO**: dropout in all layers.
- Bal: the loss with balanced weights is used in training.

### COST FUNCTION

Allows to control false negatives / useful for pedagogical reasons

$$Cost = 0.5 FPR + FNR$$

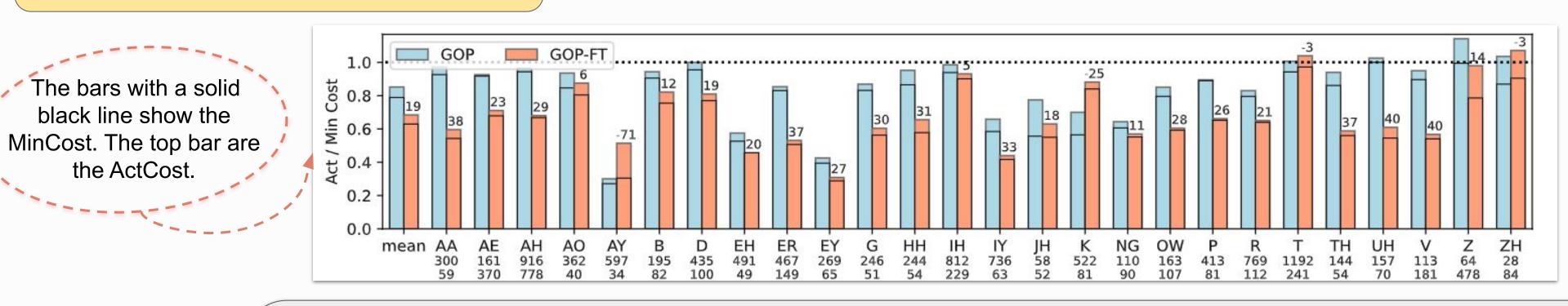
- Allows to to see the effect of the threshold selection.
- MinCost: computed on test data / ActCost: computed on dev data.



- Average 1-AUC and MinCost (phones with more than 50 samples of each class for the development data)
- GOP system has 1-AUC of 0.286 and MinCost of 0.801.
- Best configuration: LayO+1 BN DO BAL

## RESULTS / CONCLUSIONS

the ActCost.



- ActCost is within 10% of the MinCost for most phones (thresholds on development speakers) generalize well to the unseen speakers).
- Average FNR rate is 10% (GOP) and 13% (GOP-FT). Acceptable level for real use scenarios.
- Average FPR is 64% (GOP) and 41% (GOP-FT). 23% relative improvement from fine-tuning approach.