

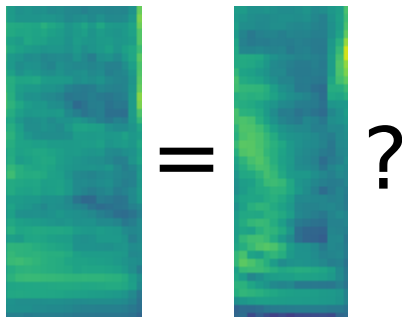


Universität Hamburg
DER FORSCHUNG | DER LEHRE | DER BILDUNG

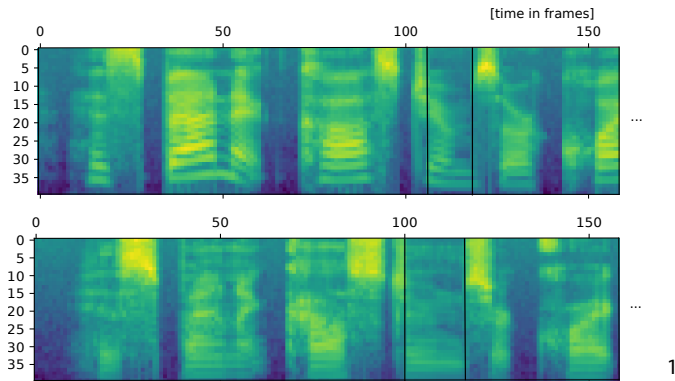
Benjamin Milde, Chris Biemann

UNSPEECH: UNSUPERVISED SPEECH CONTEXT EMBEDDINGS

Motivation

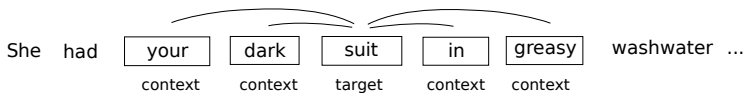


Motivation - Context



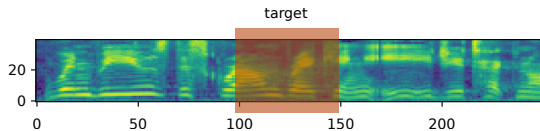
¹ Example in the style of: Aren Jansen, Samuel Thomas, and Hynek Hermansky. 2013. Weak top-down constraints for unsupervised acoustic model training. In ICASSP, pages 8091–8095.

Inspiration - Negative sampling

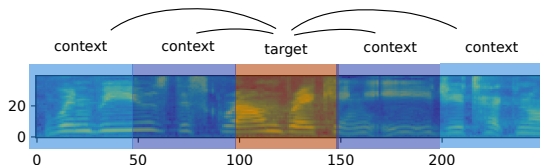


- Word2vec, skipgram with negative sampling
- Binary task instead of directly predicting surrounding words
- Is "dark" + "suit" a context pair?

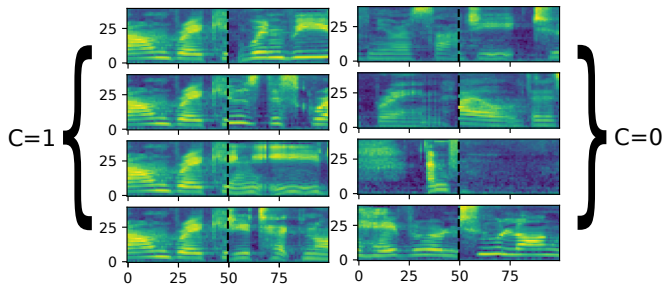
Context example



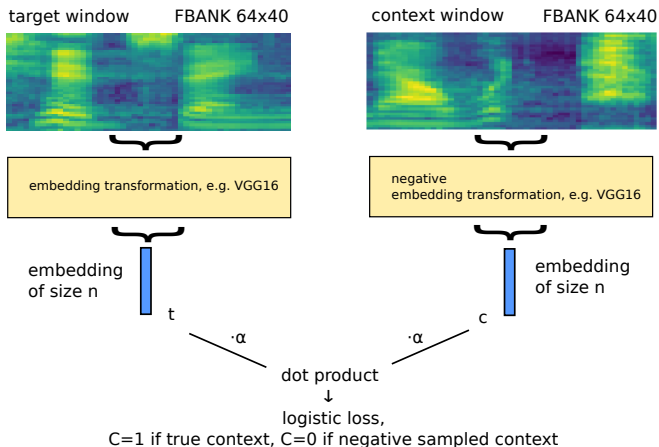
Context example



Example samples



Proposed model



Negative sampling loss

$$\begin{aligned} NEG_{loss} = & -k \cdot \log(\sigma(\mathit{emb}_t^T \mathit{emb}_c)) \\ & - \sum_{i=1}^k \log(1 - \sigma(\mathit{emb}_{neg1_i}^T \mathit{emb}_{neg2_i})) \end{aligned} \tag{1}$$

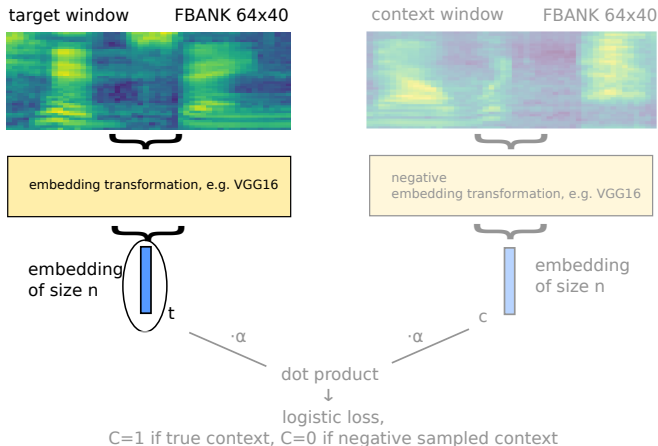
- The objective function is similar to negative sampling in word2vec
- But we are not contrasting emb_t with a emb_{neg} and choose two random unrelated samples instead for the negative sum.

Negative sampling loss

$$\begin{aligned} NEG_{loss} &= -k \cdot \log(\sigma(\mathit{emb}_t^T \mathit{emb}_c)) \\ &\quad - \sum_{i=1}^k \log(1 - \sigma(\mathit{emb}_{neg1_i}^T \mathit{emb}_{neg2_i})) \end{aligned} \tag{2}$$

- The objective function is similar to negative sampling in word2vec
- But we are not contrasting emb_t with a emb_{neg} and choose two random unrelated samples instead for the negative sum.

Applying a trained unspeech model



Unspeech rep. of an utterance

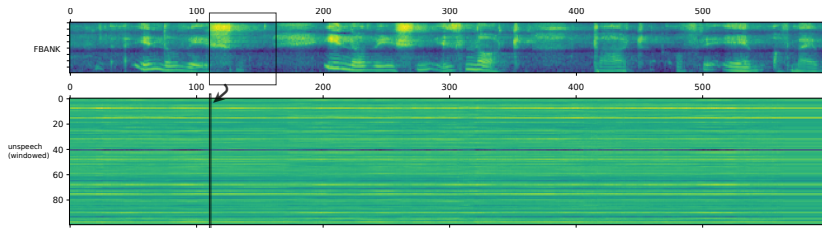


Figure: Windowed unspeech-64 representation

TSNE plot TED-LIUM dev set

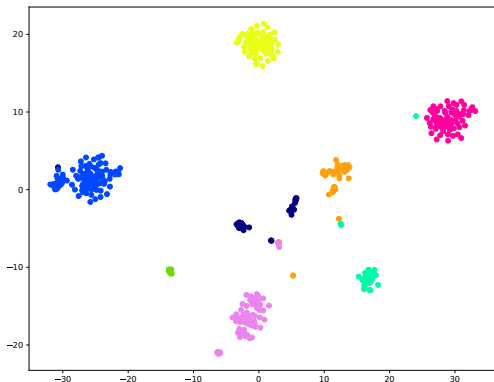
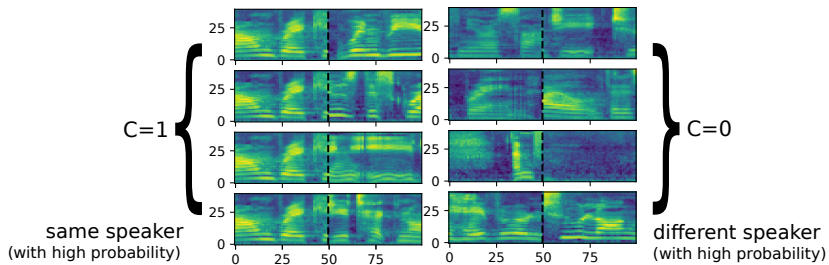


Figure: TSNE plot of unspeech vectors averaged across utterances, TED-LIUM dev set

Example Samples



Evaluation

- Speaker embedding
- Context clustering
- ASR evaluations with Kaldi:
 - Context clustering → cluster-ids in speaker adaptation
 - Providing TDNN-HMM acoustic models with unspeech context embeddings

Evaluation: datasets

Table: Comparison of English speech data sets used in our evaluations

| dataset | hours | | | speakers | | |
|-----------------|-------|-----|------|-------------|------|------|
| | train | dev | test | train | dev | test |
| TED-LIUM V2 | 211 | 2 | 1 | 1273+3 | 14+4 | 13+2 |
| Common Voice V1 | 242 | 5 | 5 | 16677 | 2728 | 2768 |
| TEDx (crawled) | 9505 | | | 41520 talks | | |

Same/different speaker experiment

Table: Equal error rates (EER) on TED-LIUM V2 – Unspeech embeddings correlate with speaker embeddings.

| Embedding | EER | | |
|-----------------------|--------------|--------------|--------------|
| | train | dev | test |
| TED-LIUM: | | | |
| (1) i-vector | 7.59% | 0.46% | 1.09% |
| (2) i-vector-sp | 7.57% | 0.47% | 0.93% |
| (3) unspeech-32-sp | 13.84% | 5.56% | 3.73% |
| (4) unspeech-64 | 15.42% | 5.35% | 2.40% |
| (5) unspeech-64-sp | 13.92% | 3.4% | 3.31% |
| (6) unspeech-64-tedx | 19.56% | 7.96% | 4.96% |
| (7) unspeech-128-tedx | 20.32% | 5.56% | 5.45% |

EER = equal error rate, point on a false positive / false negative curve, where both error rates are equal -32 = 32 input frames, -64 = 64 input frames, ...

Context clustering

- Averaged unspeech vectors across time: one 100d vector per utterance
- We use HDBSCAN* to cluster, a modern density based cluster algorithm ²
- Scales well to large quantities (average case complexity $\approx N \log N$)
- Parameters are easy to set, no epsilon like in vanilla DBSCAN

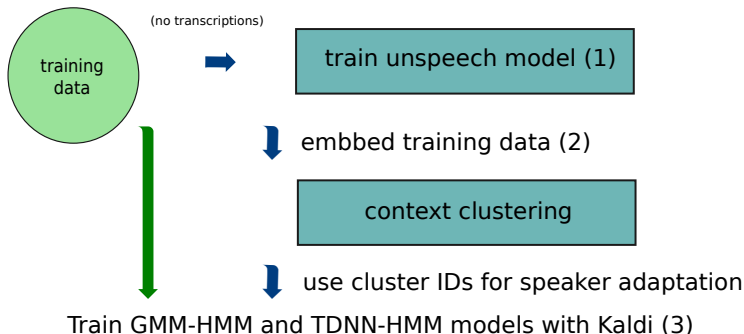
²L. McInnes, J. Healy, and S. Astels, HDBSCAN*: Hierarchical density based clustering,” The Journal of Open Source Software, vol. 2, no. 11, p. 205, 2017.

Context clustering - NMI

Table: Comparing Normalized Mutual Information (NMI) on clustered utterances from TED-LIUM using i-vectors and (normalized) Unspeech embeddings with speaker labels from the corpus. "-sp" denotes embeddings trained with speed-perturbed training data.

| Embedding | Num. clusters | | | Outliers | | | NMI | | |
|----------------|---------------|-----|------|----------|-----|------|---------------|---------------|---------------|
| | train | dev | test | train | dev | test | train | dev | test |
| TED-LIUM IDs | 1273 (1492) | 14 | 13 | 3 | 4 | 2 | 1.0 | 1.0 | 1.0 |
| i-vector | 1630 | 12 | 10 | 8699 | 1 | 2 | 0.9605 | 0.9804 | 0.9598 |
| i-vector-sp | 1623 | 12 | 10 | 9068 | 1 | 2 | 0.9592 | 0.9804 | 0.9598 |
| unspeech-32-sp | 1686 | 16 | 12 | 3235 | 22 | 32 | 0.9780 | 0.9536 | 0.9146 |
| unspeech-64 | 1690 | 16 | 11 | 5690 | 14 | 21 | 0.9636 | 0.9636 | 0.9493 |
| unspeech-64-sp | 1702 | 15 | 11 | 3705 | 23 | 25 | 0.9730 | 0.9633 | 0.9366 |

Context clustering for ASR



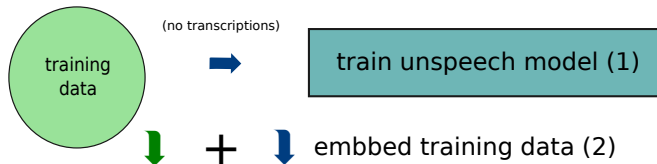
Context clustering for ASR

WER results

Table: WERs for different context IDs for speaker adaptation in TDNN-HMM ASR models. (One speaker per talk, one speaker per utterance, unspeech hdbscan IDs)

| Acoustic model | Spk. div. | Dev WER | Test WER |
|----------------|----------------|---------|-------------|
| GMM-HMM | per talk | 18.2 | 16.7 |
| TDNN-HMM | | 7.8 | 8.2 |
| GMM-HMM | per utt. | 18.7 | 19.2 |
| TDNN-HMM | | 7.9 | 9.0 |
| GMM-HMM | Unspeech 64 | 17.4 | 16.5 |
| TDNN-HMM | | 7.8 | 8.1 |

Unspeech contexts in TDNN-HMMs



TDNN-HMM models (3)

append context vectors to input

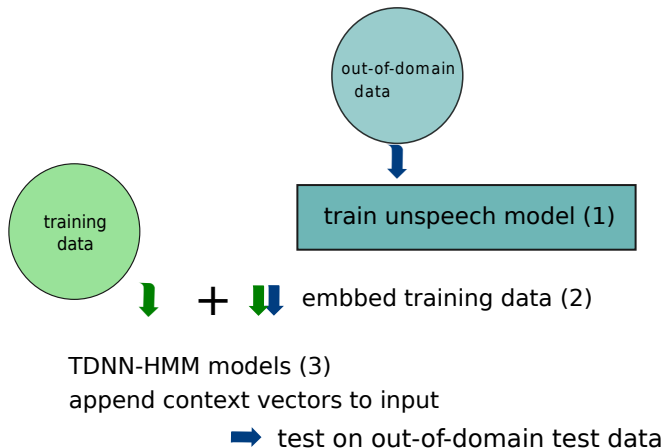
Note that the standard TDNN-HMMs recipes in Kaldi also use ivectors (speaker vectors) similarly.

Unspeech contexts in TDNN-HMMs

Table: WER for TDNN-HMM chain models trained with Unspeech embeddings on TED-LIUM.

| Context vector | Dev WER | Test WER |
|-----------------------------|------------|------------|
| (1) none | 8.5 | 9.1 |
| (2) i-vector-sp-ted | 7.5 | 8.2 |
| (3) unspeech-64-sp-ted | 8.3 | 9.0 |
| (4) unspeech-64-sp-cv | 8.3 | 9.1 |
| (5) unspeech-64-sp-cv + (2) | 7.6 | 8.1 |
| (6) unspeech-64-tedx | 8.2 | 8.7 |
| (7) unspeech-128-tedx | 8.2 | 8.9 |

Unspeech contexts in TDNN-HMMs



Unspeech contexts in TDNN-HMMs

WER results out-of-domain test data

Table: Training on TED-LIUM and decoding on Common Voice V1.

| Context vector | Dev WER | Test WER |
|-----------------------------|-------------|-------------|
| (1) none | 29.6 | 28.5 |
| (2) i-vector-sp-ted | 29.0 | 28.2 |
| (3) unspeech-64-sp-cv | 27.9 | 26.9 |
| (4) unspeech-64-sp-cv + (2) | 28.2 | 27.4 |
| (5) unspeech-64-tedx | 28.8 | 27.5 |
| (6) unspeech-128-tedx | 28.7 | 28.0 |

Conclusion

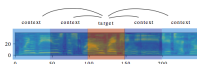
- We showed a simple unsupervised context embedding method, that can be trained on large amounts of unlabelled data
- Our context embeddings contain speaker characteristics
- Our method can be used for context clustering
- Context cluster ids can aid speaker adaptation in acoustic models when no speaker information is available
- Can help in domain adaptation, when the unspeech models are trained on unlabelled data of the target domain

- <http://unspeech.net> - Download model source code (Python3/Tensorflow), pretrained models and documentation



Unspeech

Unsupervised Speech Context Embeddings



Unspeech embeddings are based on unsupervised learning of context feature representations of spoken language. Variance and variability in recordings of speech and its representations are a common problem in automatic speech processing tasks. E.g. speaker, environment characteristics and the type of microphone will make large differences in typical speech representations (e.g. FBANK, MFCC), making (direct) similarity comparisons difficult. We can describe such factors of variance also as the context of an utterance; speech sounds that occur close in time share similar contexts. Unspeech allows you to learn embeddings of such contexts in an unsupervised way on raw speech data; speaker IDs, channel information or transcriptions are not needed.

Uses cases:

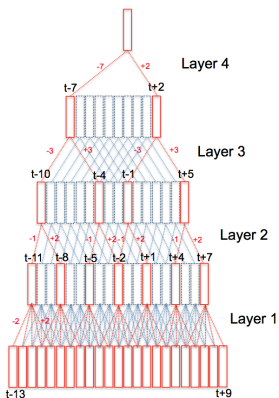
- Cluster a speech corpus in-domain, to help speaker adaption methods in HMM-GMM and (TD)NN-HMM acoustic models; without the need for speaker annotations or trained speaker embeddings.
- As a context embedding in acoustic models; provide additional information to acoustic models.

Questions?

- now
- after the session
- or mail me: milde@informatik.uni-hamburg.de

Extra slides

Unspeech contexts in TDNN-HMMs



Stationary hypothesis

Automatic Speaker Clustering (Jin et. al 1997):³

- "Our algorithm takes the advantage [...] that consecutive segments are more likely to come from the same speaker"
- "In practice, we regard speaker as a generic concept which really means speaker with channel and background condition"
- We call this generic concept "context"

³H. Jin, F. Kubala, and R. Schwartz, "Automatic speaker clustering," in Proceedings of the DARPA speech recognition workshop, 1997, pp. 108–111