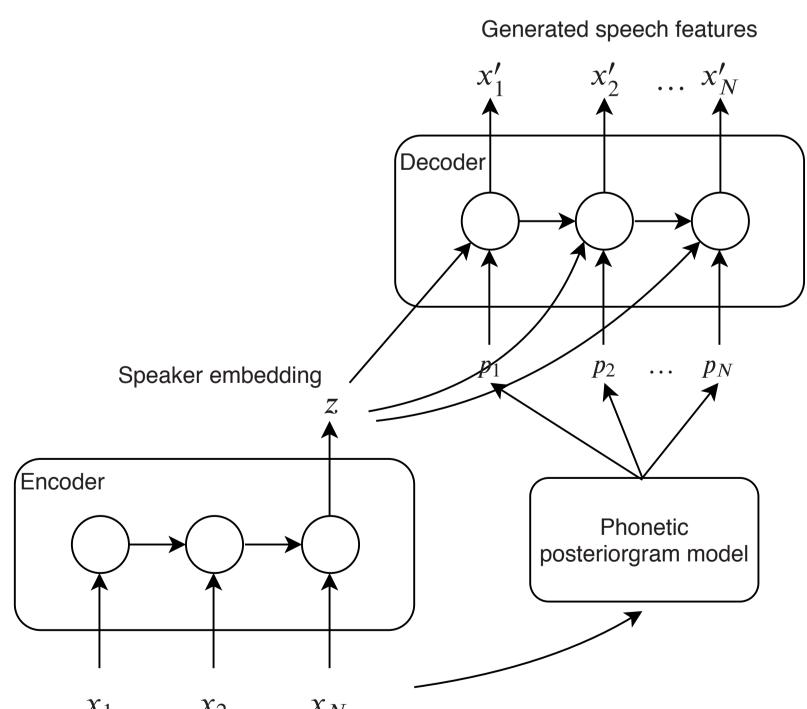
ONE-SHOT VOICE CONVERSION WITH DISENTANGLED REPRESENTATIONS BY LEVERAGING PHONETIC POSTERIORGRAMS

INTRODUCTION

- **Voice Conversion** (VC): converts a source speaker's speech to sound like a target speaker's voice.
- VC preserves target's speaker identity and source's phonetic context.
- One-shot VC methods typically disentangle speaker identity and phonetic context. Then speaker identity representation is modified while keeping phonetic context constant.
- Challenges: The models cannot fully disentangle these factors as shown in a previous study.
- **Proposal**: We posit that the senone posteriorgrams (PPG) from an already-trained ASR model can be used in lieu of learned phonetic context representations.
 - We focus on learning only the speaker representation.
 - We present a one-shot voice conversion technique by modifying the learned speaker identity representation.
 - Through experiments, we show that modification of these factors allows better disentanglement and hence transformation of voice.

MODEL

- Our proposed model consists of an encoder and a decoder (RNNs)
 - Encoder's input is MCEPs, outputting a speaker embedding vector: $z = E(\mathbf{X})$
 - Decoder takes the generated speaker embedding along with PPG sequence as input, and generates the acoustic features: $X' = D(\mathbf{P}, z)$



 x_1 x_2 ... x_N Input speech features

• We train the model by optimizing the training loss:

 $\ell(X, X') = \sum_{i=1}^{N} ||x_i - x'_i||_2^2$

- To perform VC:
 - Compute z^{src} and z^{trg} of the source and target utterances.
 - Compute average diff vector $\overline{z}^{diff} = \overline{z}^{trg} \overline{z}^{src}$
 - Add average diff vector to source $z^{converted} = z^{src} + \overline{z}^{diff}$

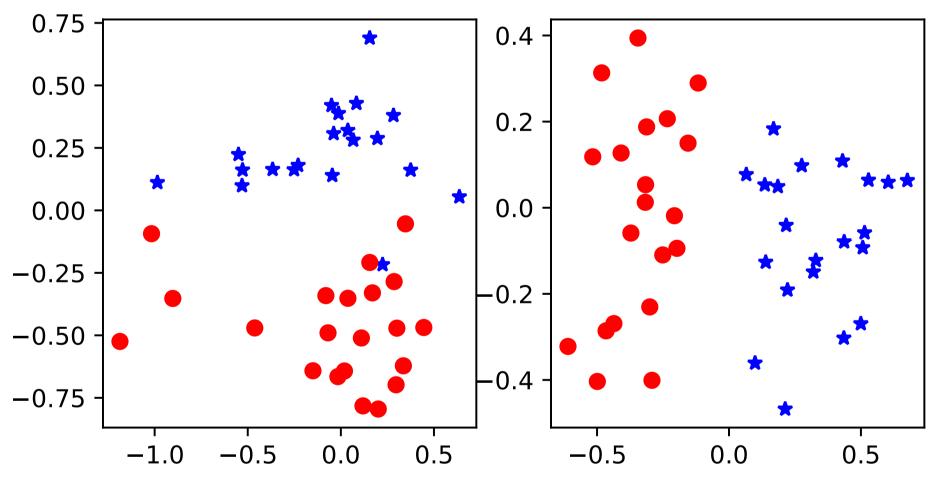
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EXPERIMENT

- We used the TIMIT corpus as the training data.
- To compute the phonetic posteriorgrams, we use Kaldi
- We use librispeech as speech corpus to train ASR.

	Encoder
recurrent layer	GRU-1024, Dropout
output layer	$FC-D_z$, ReLU
Decoder	
dense block	input PPGs, FC-1024
combine layer	dense output + speake
dense block	FC-1024, ReLU, Dro
recurrent layer	GRU-1024, Dropout
output layer	$FC-D_x$

The network architectures of our encoder and decoder models.



Visualization of speaker embedding: Blue dots are male speakers and red dots are females. FHVAE (left) vs. Proposed (right).

- We use Factorized Hierarchal Variational Autoencoder (FHVAE)
- [1] as baseline.
- We observe that
- The proposed model's computed speaker embeddings for different speakers fall further apart compared to FHVAE.
- Also they are more evenly distributed compared to VAE embeddings which tend to be more densely distributed.
- The gender clusters have a better separation margin.
- This subjectively depicts a more robust speaker embedding quality.
- The voice conversion samples are available at:
- https://shamidreza.github.io/is19samples

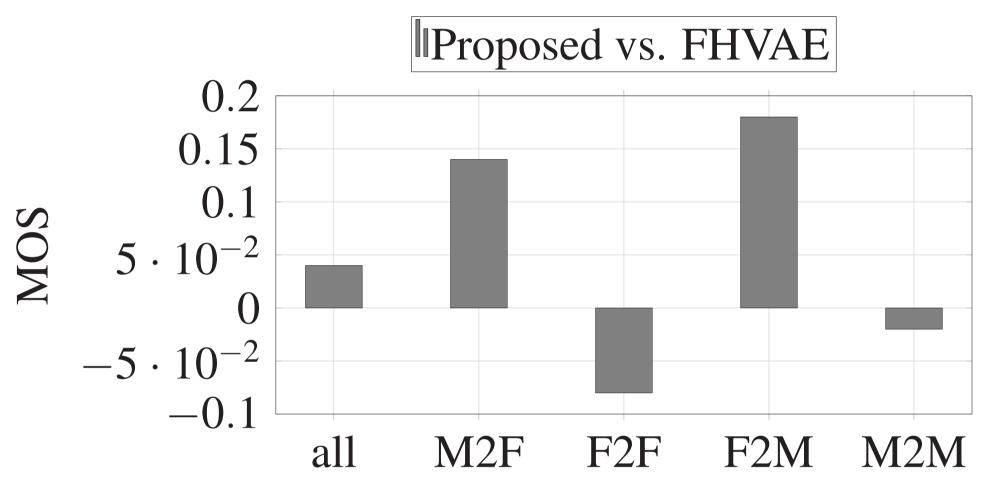
[1] Seyed Hamidreza Mohammadi and Taehwan Kim, Investigation of using disentangled and interpretable representations for one-shot cross-lingual voice conversion." Interspeech 2018.

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4, ReLU, Dropout ker embedding zopout

EXPERIMENT: SPEECH QUALITY

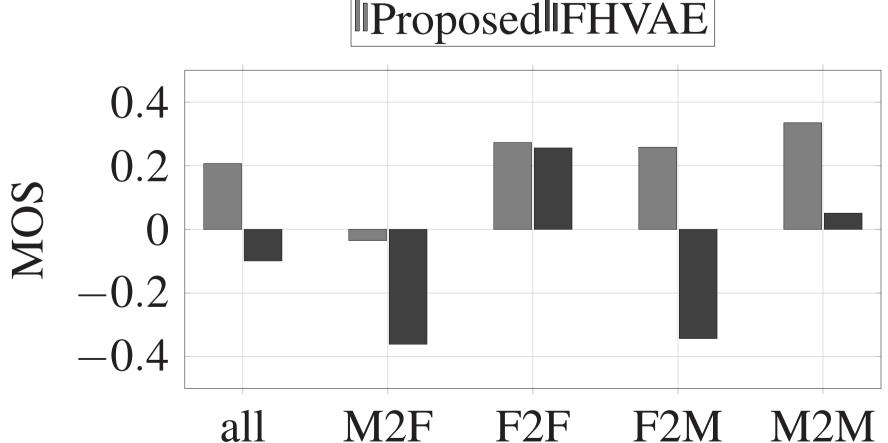
- by using +2 (much better) to -2 (much worse) score.
- condition.
- conditions.



Speech Quality average score with gender break-down. Positive scores favor proposed model. F2M preference score is statistically significant.

EXPERIMENT: SPEAKER SIMILARITY

- (definitely same) to -2 score (definitely different)
- -0.10 ± 0.12
- FHVAE in all comparison pairs



all Speaker Similarity average score with gender break-down. Positive scores are desirable. (confidence intervals for all is close to 0.11, and all score-pairs are statistically significant)



• We show the speech quality Comparative Mean Opinion Score (CMOS) in which 50 listeners score which sample quality is better • We found statistical significant preference scores for F2M

• We did not find statistical significant preference scores for other

• 50 listeners listen and rate A and B with score ranging from +2 • The results show proposed and FHVAE achieving 0.20 ± 0.11 and

• The proposed model performs statistically significantly better than

Proposed FHVAE