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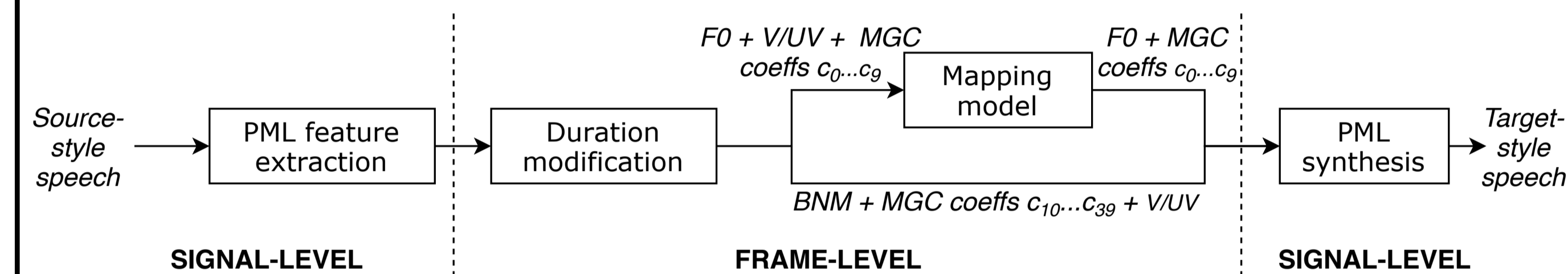
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## Goal of the study

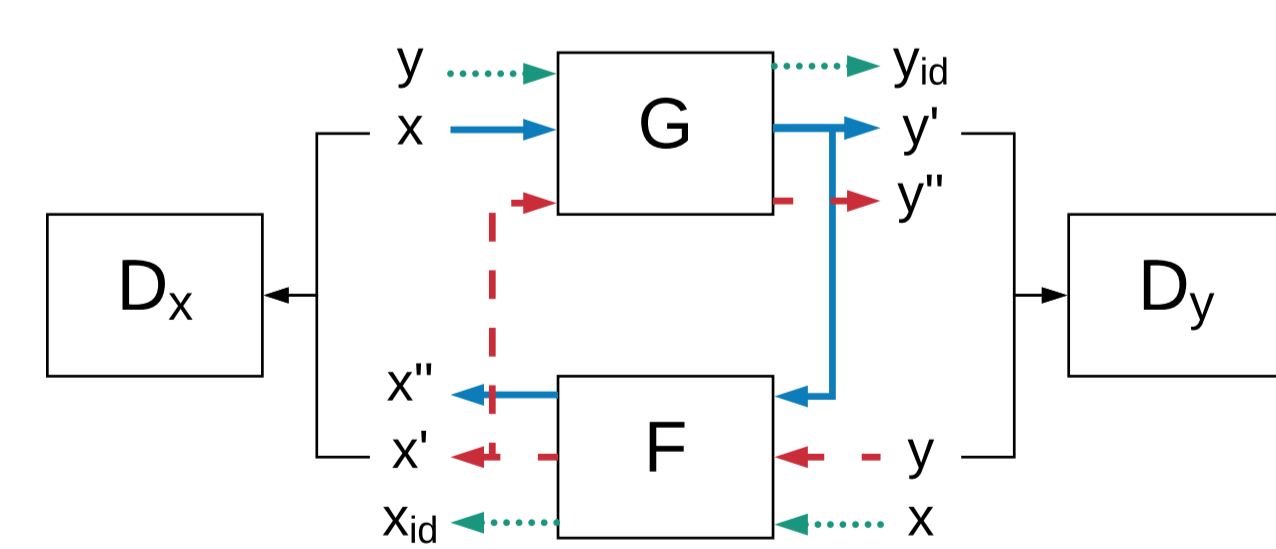
- Speaking style conversion (SSC) [1] is the technology of converting natural speech signals from one style to another.
- This study focuses on SSC for speech with varying vocal effort, focused on conversion between normal and Lombard
- We use CycleGANs [2] as a mapping model with PML vocoder features.
- The CycleGAN was compared in subjective listening tests with 2 other standard mapping methods used in conversion.

## Parametric SSC system

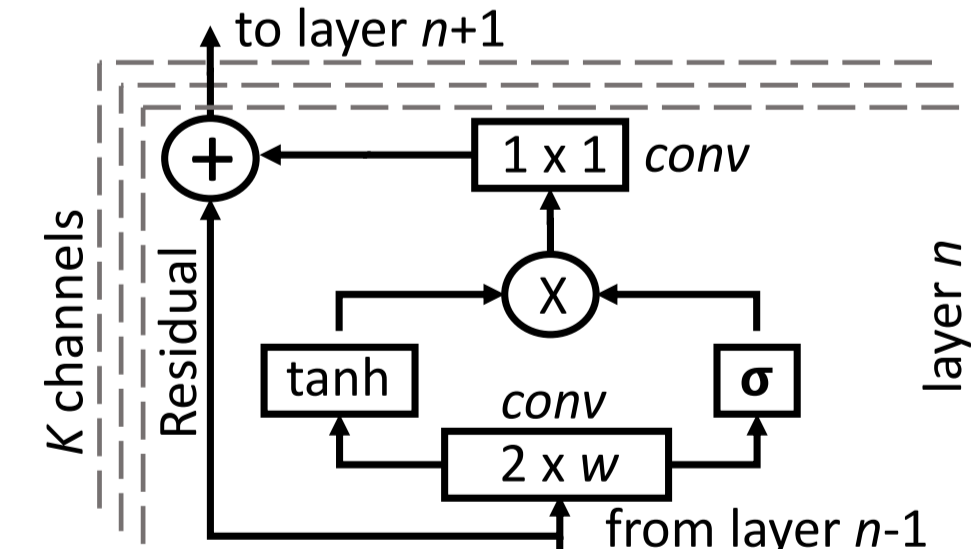


- Frame level features extracted from source signal using PML vocoder [3]: fundamental frequency, binary noise mask and spectral envelope.
- Duration modification based on characteristic voiced and unvoiced durations in each style.
- Features relevant for transformation between normal and Lombard speaking styles are transformed using a machine learning mapping model.
- Mapped features are converted to a speech waveform in the target style with the PML vocoder.

## CycleGAN



Mapping functions  $G$  and  $F$ , and discriminators  $D_X$  and  $D_Y$ . The forward cycle, backward cycle, and identity mapping indicated with red, blue, and green respectively.



Block diagram of layer  $n$  of the CNN used to model  $G$ ,  $F$ ,  $D_X$  and  $D_Y$ .

- A CycleGAN [2] is a non-parallel learning scheme that learns bi-directional deterministic mappings.
- Trained using adversarial learning – generative models trained as a solution to a minmax two-player game between two neural networks called as the generator and discriminator.
- We use Wasserstein distance metric (WGAN loss) with gradient penalty, along with an identity mapping loss.
- The CNN shown has 8 layers and 256 channels with 11-point convolutions (similar to [4]).

## Data

- Read and conversational speech recordings [5] from 20 Finnish speakers (10 female), in normal and Lombard styles.
- *Read* - each speaker read a text of 90 words (~ 1 minute).
- *Conversational* - realistic telephone conversations, where the subjects played the role of either a caller or a travel agent. Size is approximately the same as the read section.
- In order to elicit Lombard speech, background noise was played to the speakers' ears with headphones while they were being recorded.

## Compared Mapping Methods

### Parallel GMM

- A standard GMM is used with 8 components.
- DTW aligned features used to train frame-level models.

### INCA

- Non-parallel learning scheme [6] that iteratively looks for nearest neighbor feature pairs between the source and target while also iteratively updating the conversion model to progressively improve matching to the target style.
- Same 8-component GMM model as in the parallel training.
- Algorithm is run for 10 iterations.

## Subjective Evaluation

### Lombardness of mapped speech

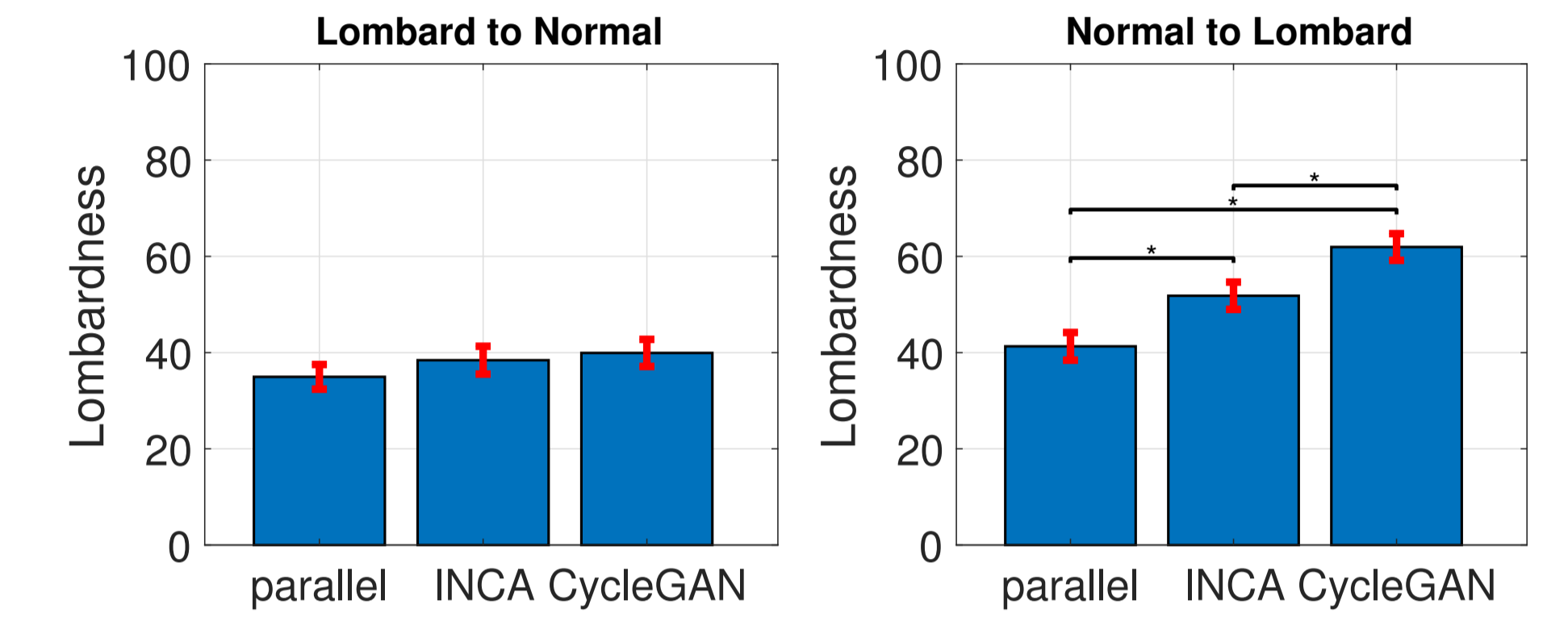
- Setup as a MUSHRA-like (Multiple Stimuli with Hidden Reference and Anchor) test.
- Aim: evaluate the Lombardness of the mapped utterances for the normal-to-Lombard and Lombard-to-normal mappings.
- Listeners rated Lombardness of mapped samples on a scale 0–100 based on known reference samples in normal and Lombard style.

### Quality of mapped speech

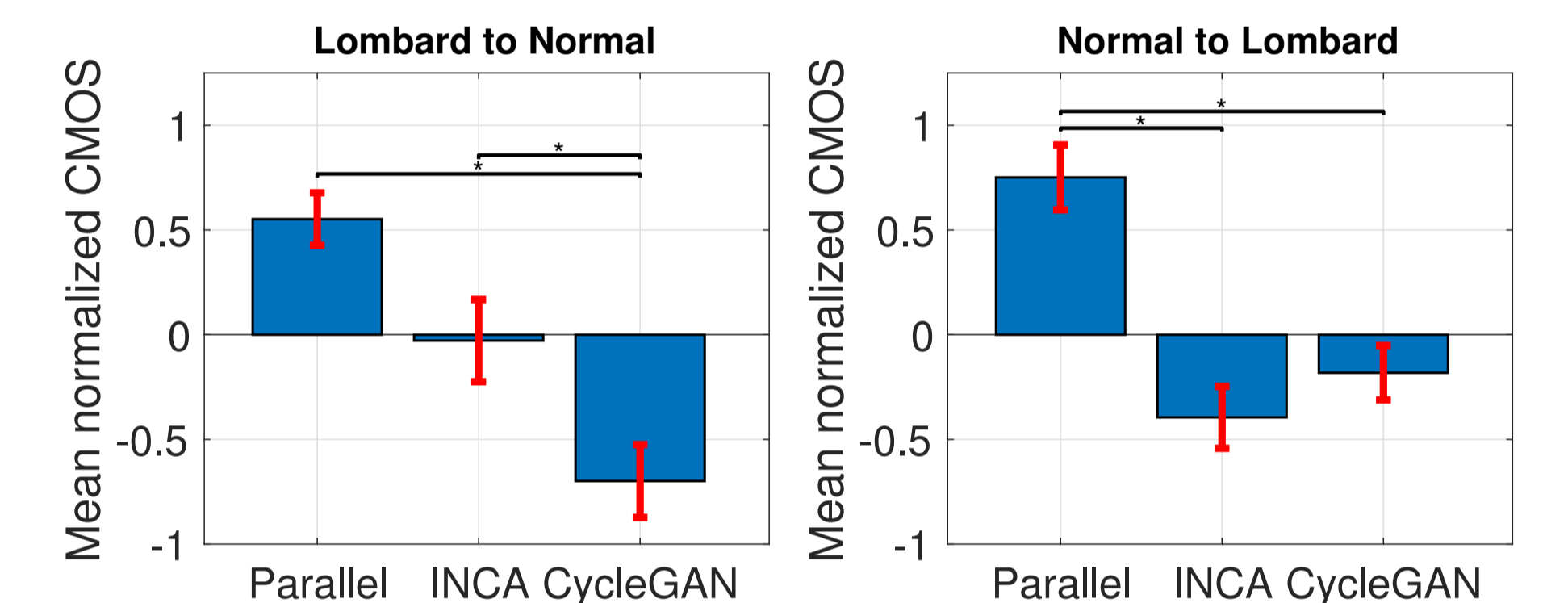
- Comparison category rating (CCR) test.
- Listeners were presented with pairs of speech utterances and asked to rate the perceived quality of the second utterance in comparison to the first one using a continuous rating scale from -3, much worse to 3, much better.
- Each utterance pair consisted of a mapped utterance and its corresponding natural Lombard utterance.
- Ratings converted to CMOS scores (smaller is better).

## Results

### Lombardness Test



### Quality Test



## Conclusions

- This work studied the use of non-parallel learning schemes to the task of vocal effort speaking style conversion, in this case between normal and Lombard speech.
- CycleGAN produces encouraging results compared to the baseline methods, producing the largest Lombard effect in normal-to-Lombard conversion while having indistinguishable quality from the INCA- based approach.
- In Lombard-to-normal conversion, CycleGAN achieves superior speech quality to the other methods.
- CycleGANs seems like a promising candidate for SSC problems, as they appear to provide a strong alternative for non-parallel training on problems where parallel data scarcity is a real challenge.
- The implementation of the CycleGAN is available on [https://github.com/shreyas253/CycleGAN\\_idCNN/](https://github.com/shreyas253/CycleGAN_idCNN/)

## References

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