An Attention-based Neural Network on Multiple Speaker Diarization

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2022 IEEE International Conference on Artificial Intelligence Circuits and Systems
Virtual & Hybrid Conference

National Yang Ming Chiao Tung University

Outline

- Introduction
- Background
- Proposed Method
- Experimental Results
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Introduction

- Speaker Diarization
  - Who spoke when?
  - Multiple speakers in each speech utterance

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Background

- Evaluation metrics - Diarization Error Rate
  - Evaluation metrics in this work

$$DER = \frac{Miss\ Detection + False\ Alarm + Overlap + Confusion}{Reference}$$

  - Notice that some papers adopt simplified metrics

$$DER\_Simplified = \frac{Confusion}{Reference}$$

Background

- Permutation-Free Objectives with Cross Entropy Loss [1]
  - Find minimum loss between hypothesis and all the possible combinational of ground truth

Ground truth

A	0	0	1	1	1
B	1	1	0	0	0

Compute Loss

$$l_n = y_n \cdot \log x_n + (1 - y_n) \cdot \log(1 - x_n)$$

$$Loss\ of\ each\ batch = \frac{\sum_0^{n=N} l_n}{N}$$

x_n *is hypothesis* n, y_n *is ground truth* n

Hypothesis

A	1	1	0	0	0
B	0	0	1	1	1

Permutation

Possible hypothesis

A	1	1	0	0	0
B	0	0	1	1	1
A	0	0	1	1	1
B	1	1	0	0	0

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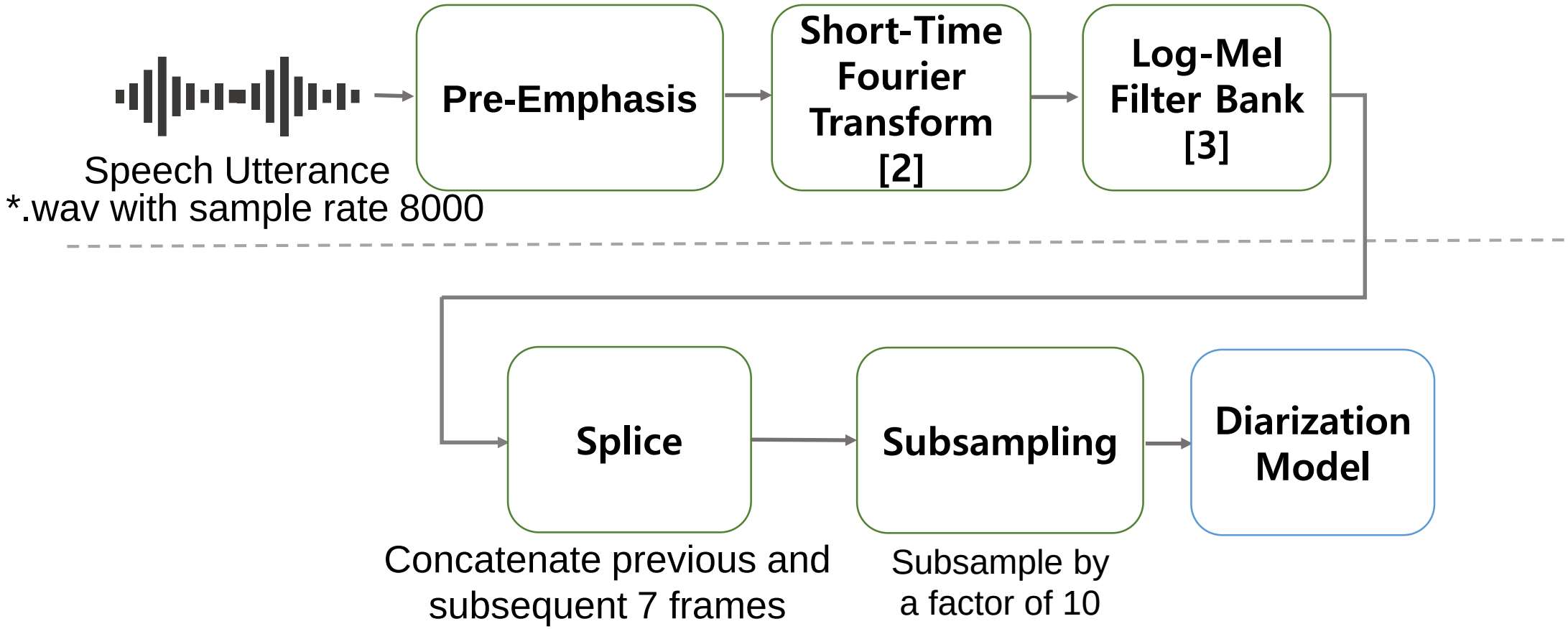
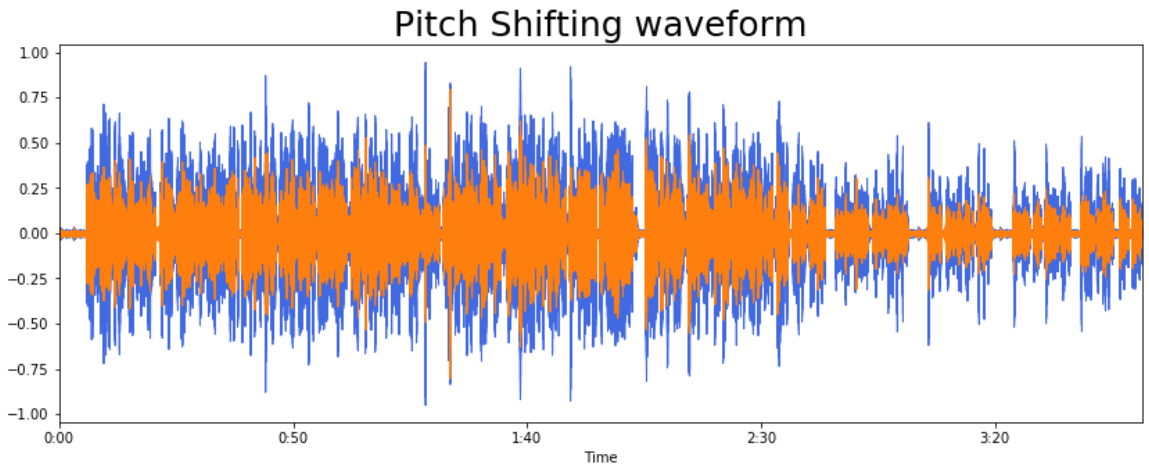
Proposed Method

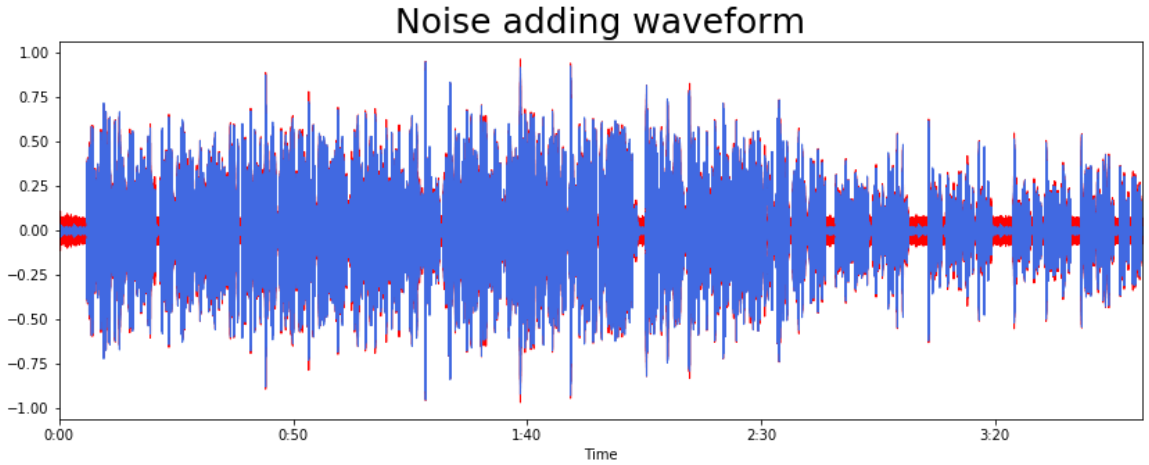
- Data Preprocessing



Proposed Method

- Data Augmentation
  - Pitch shifting
  - Noise adding
  - SpecAugment [4]





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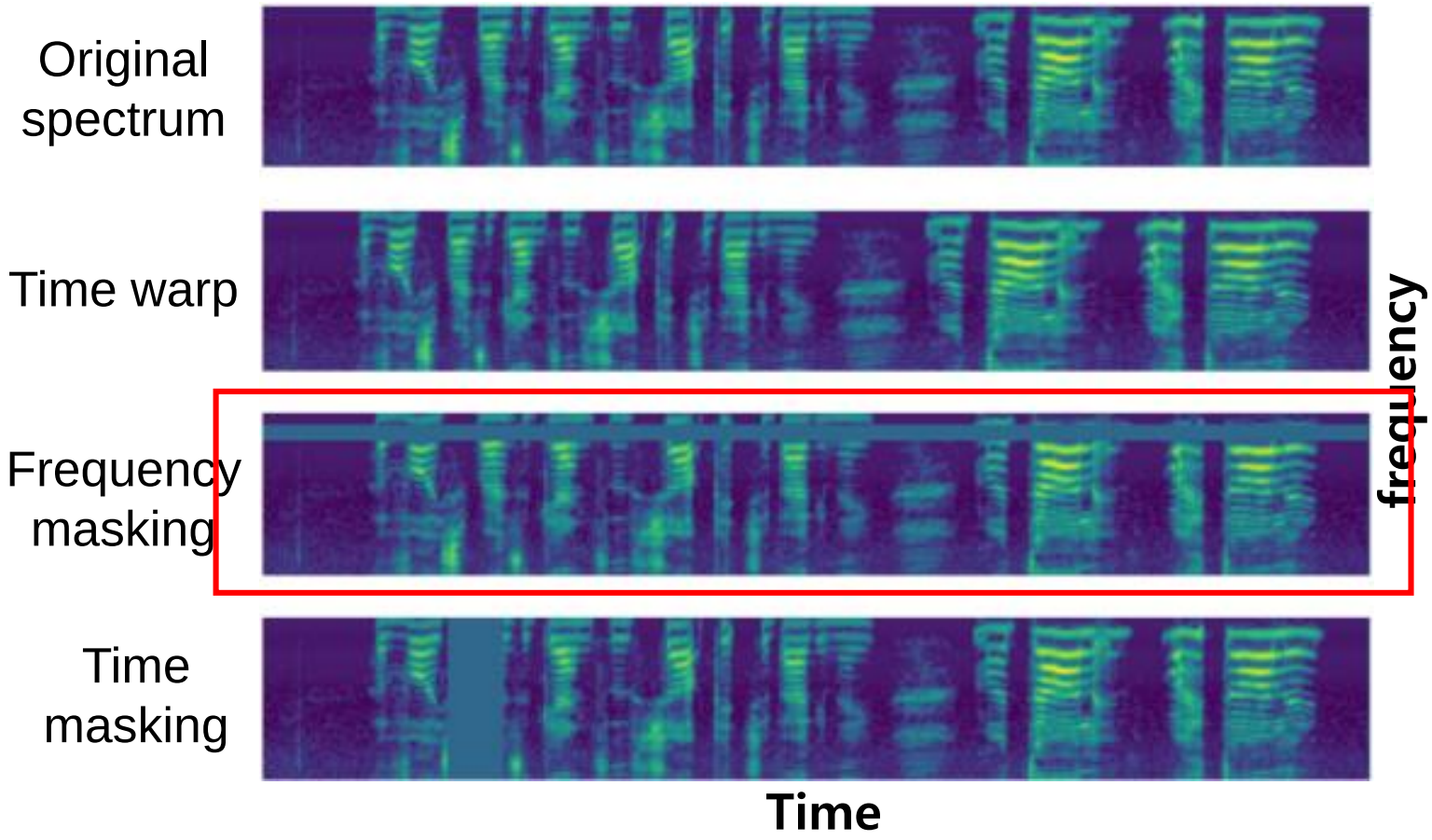
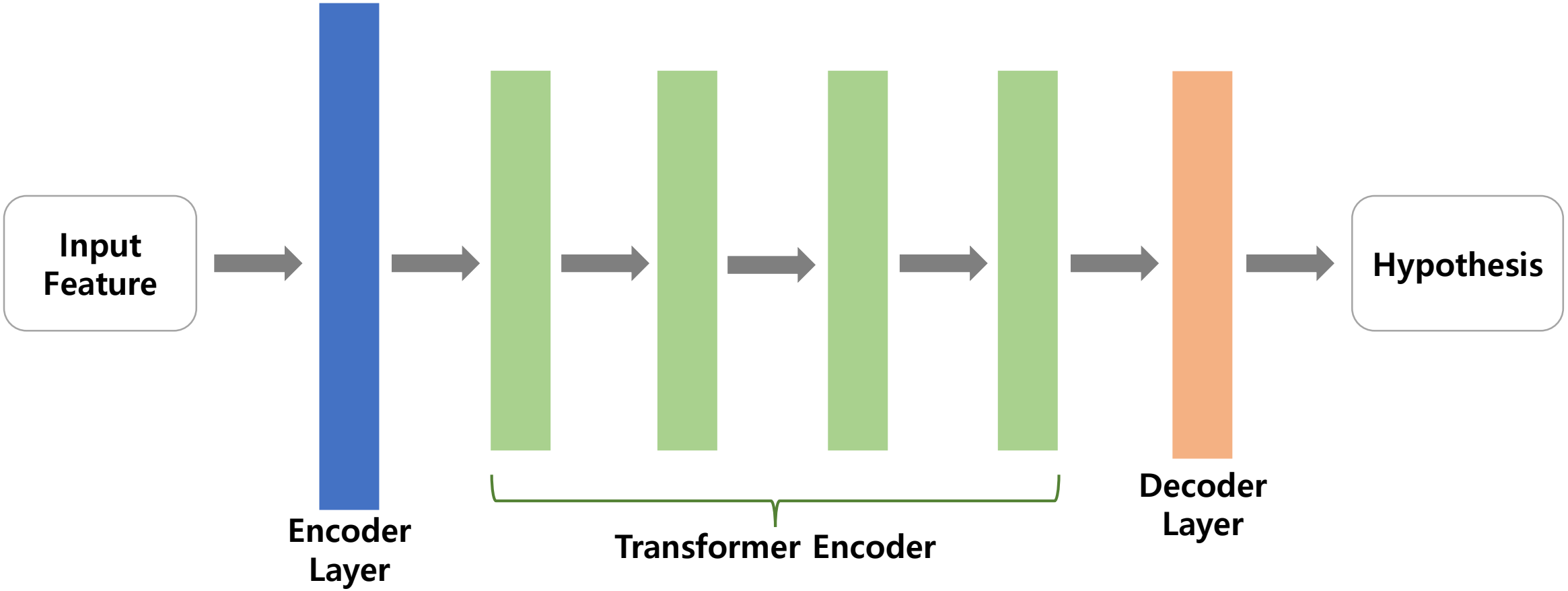


Proposed Method

- Model Structure [5]

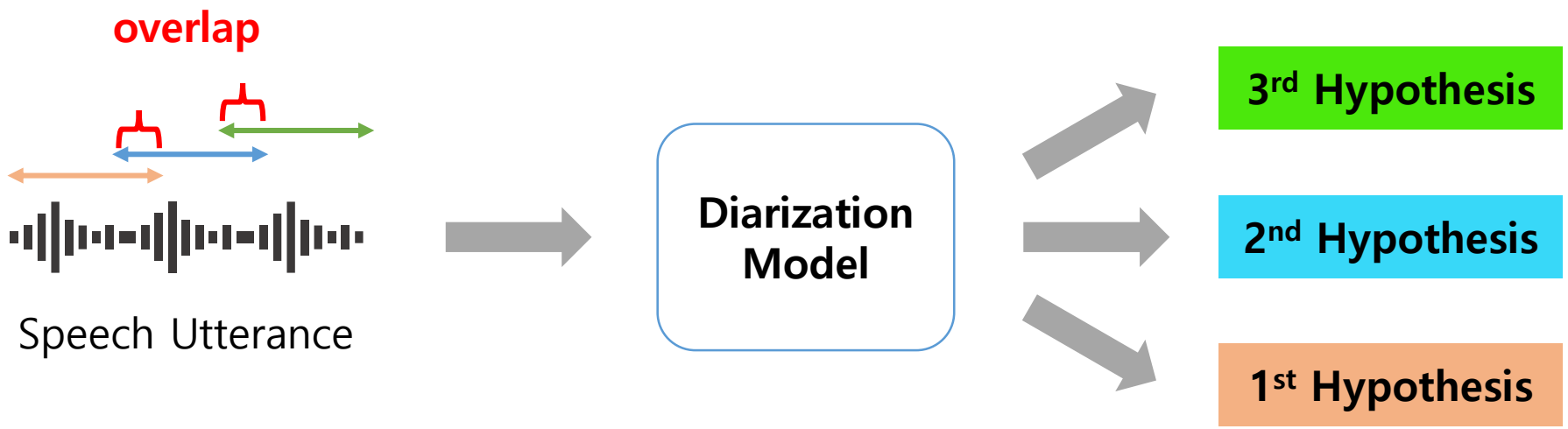


Proposed Method

- Data Postprocessing

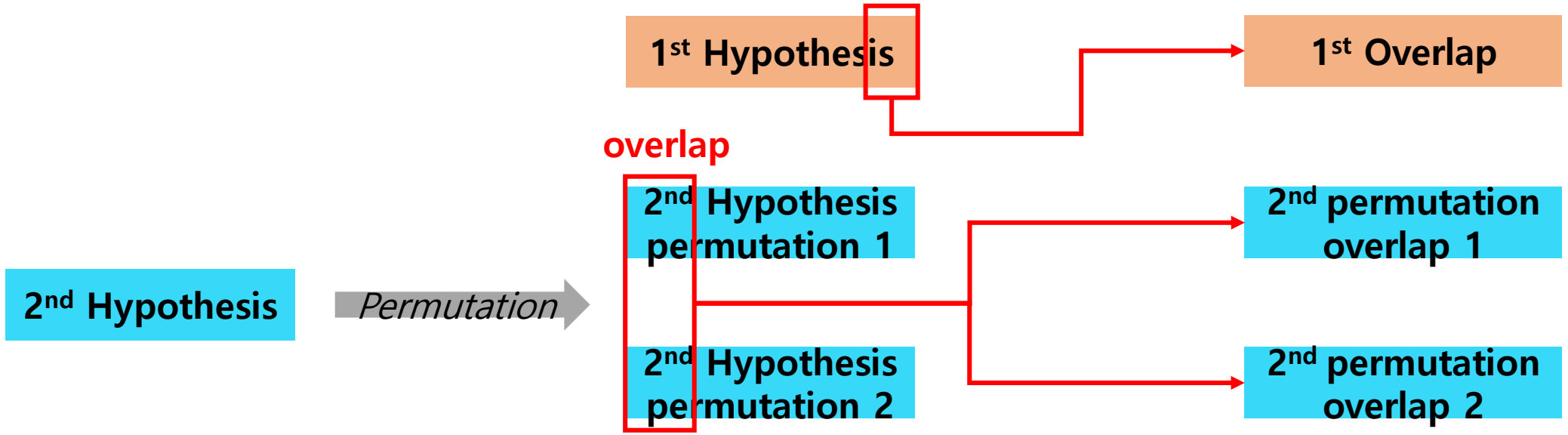
Proposed Method

- Utterance Reconstruction (1/4)
  - Split the speech utterance before inference due to length limited. Then co ncatenate the result by computing the loss of overlapping part.



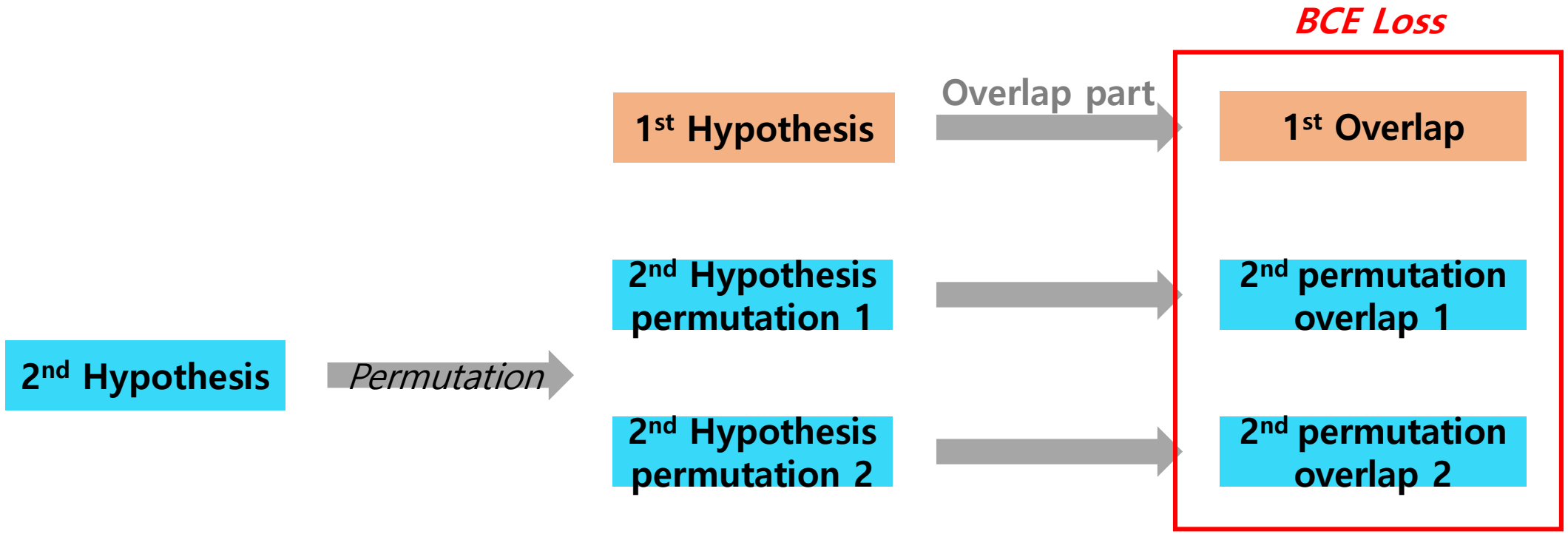
Proposed Method

- Utterance Reconstruction (2/4)
  - Split the speech utterance before inference due to length limited. Then co ncatenate the result by computing the loss of overlapping part.



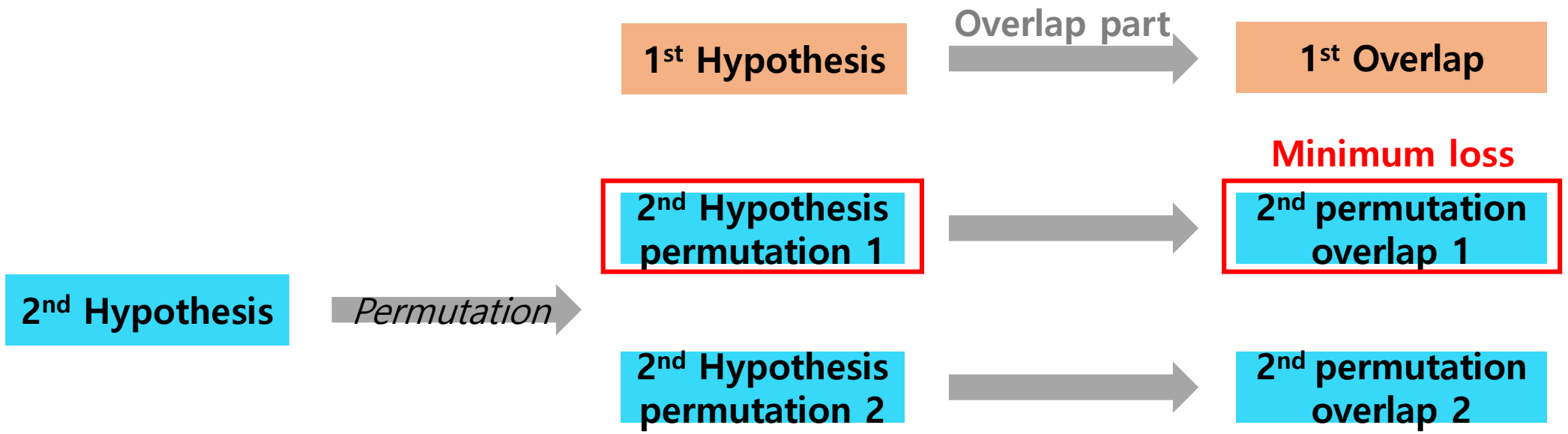
Proposed Method

- Utterance Reconstruction (3/4)
  - Split the speech utterance before inference due to length limited. Then co ncatenate the result by computing the loss of overlapping part.



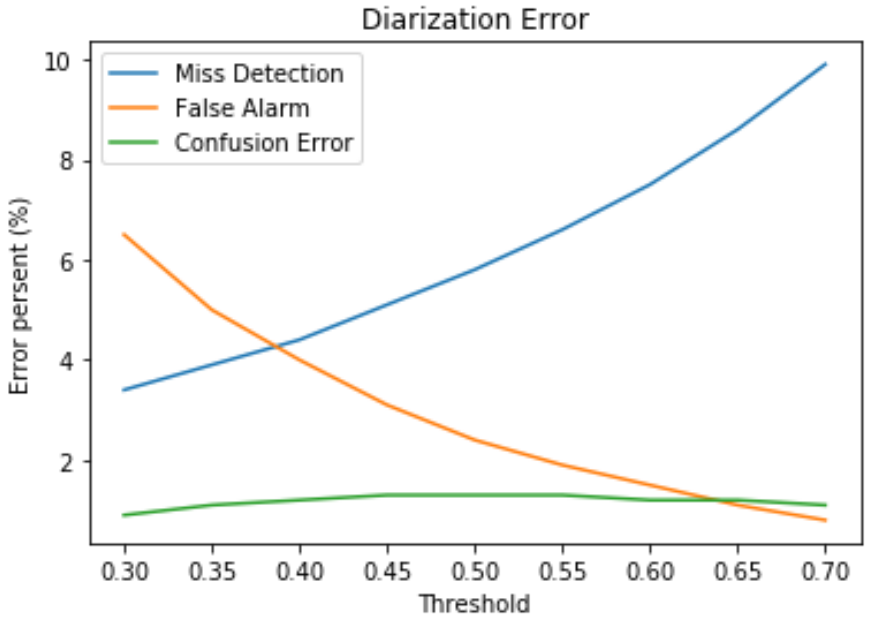
Proposed Method

- Utterance Reconstruction (4/4)
  - Split the speech utterance before inference due to length limited. Then co ncatenate the result by computing the loss of overlapping part.



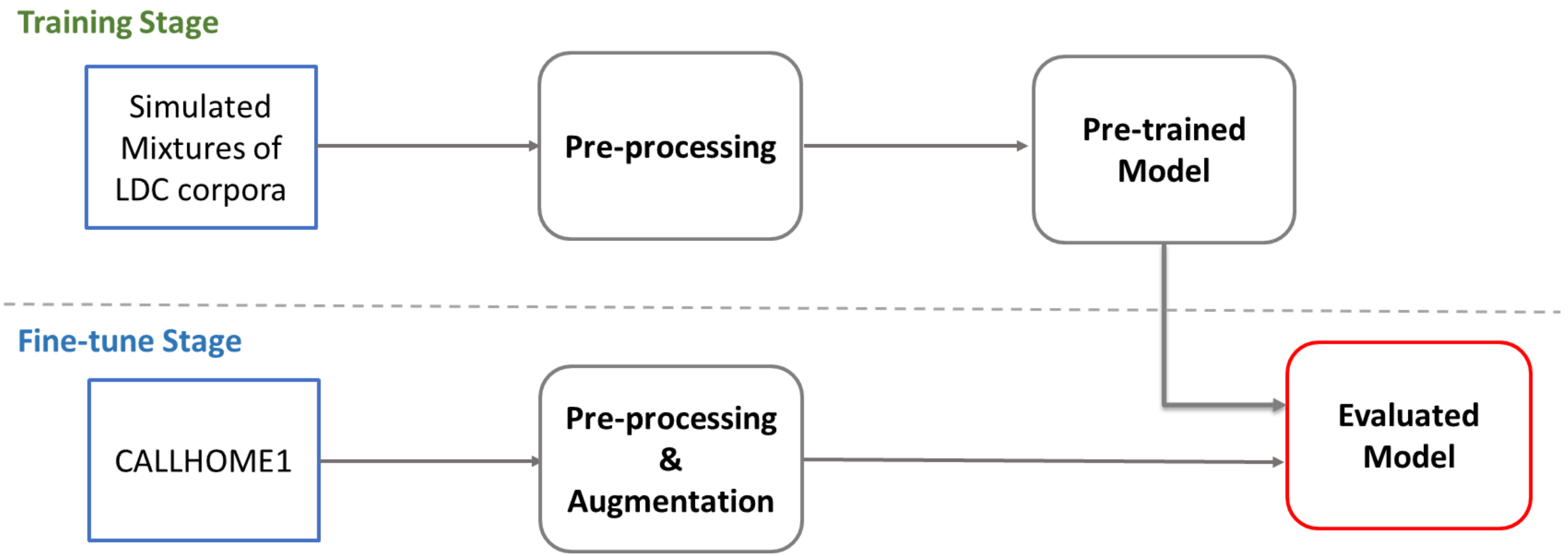
Proposed Method

- Threshold Choosing
  - Select different threshold with binary search.
    - The lower the threshold, the higher the false alarm.
    - The higher the threshold, the higher the miss detection.
    - The relationship between threshold and confusion error is small.



Proposed Method

- Training of Diarization System



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Experimental Results

- Experimental Data
  - Training set:
  - ➢Simulated Mixtures [6] of LDC Corpora [7]

  - Fine-tune set:
  - ➢Subset of 2000 NIST Speaker Recognition Evaluation (CALLHOME) [8]

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Experimental Results

- Experimental Results

Model	Pre-trained data	DER
Proposed convolution model	100000 simulated mixtures	9.47%
Proposed linear model		9.12%
SA-EEND [5]		9.54%
SA-EEND-EDA [9]	400000 simulated mixtures	8.07%

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