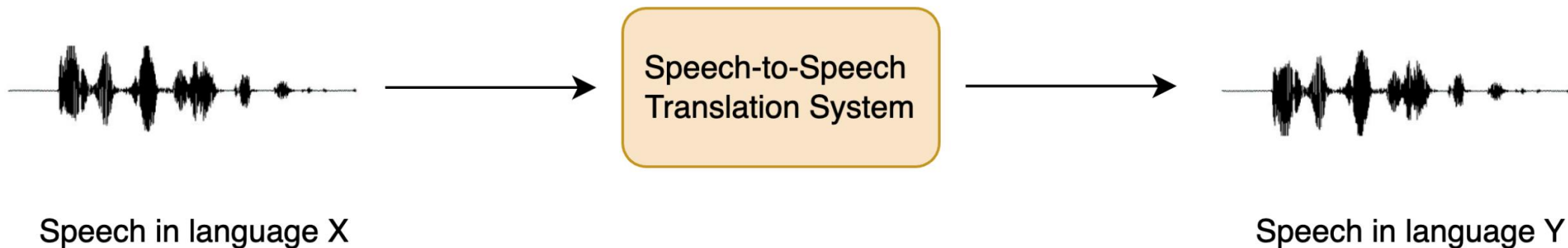


# DIFFNORM: SELF-SUPERVISED NORMALIZATION FOR NON-AUTOREGRESSIVE SPEECH-TO-SPEECH TRANSLATION

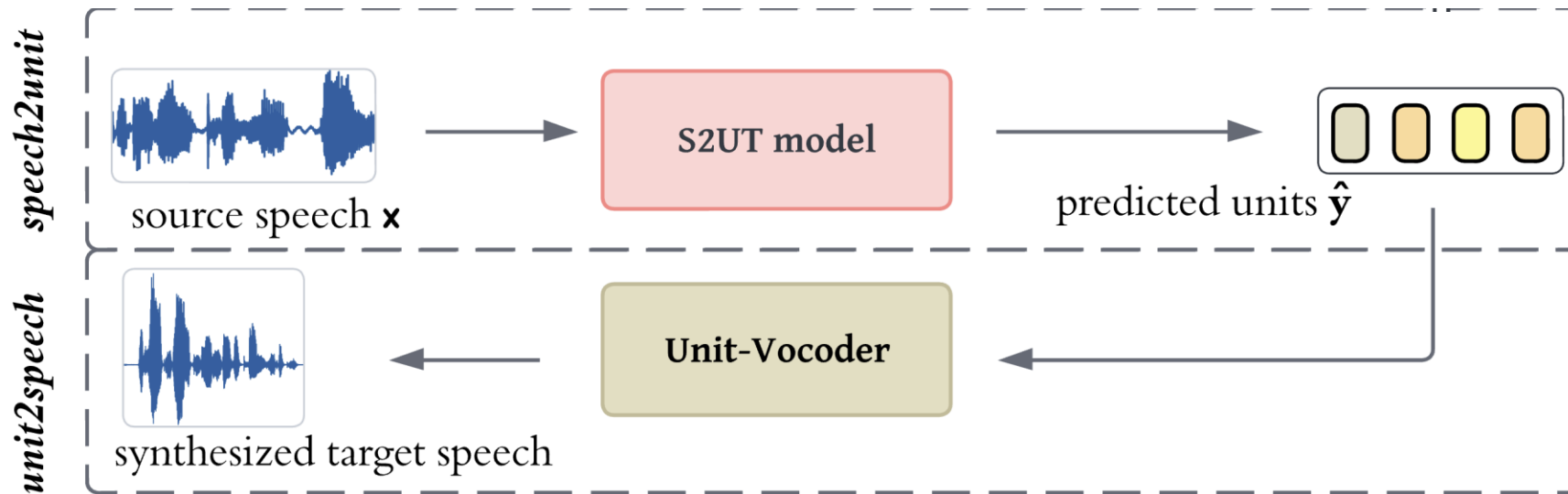
*Weiting Tan, Jingyu Zhang, Lingfeng Shen, Daniel Khashabi, Philipp Koehn*



# Speech-to-Speech Translation (S2ST)

## Two stages

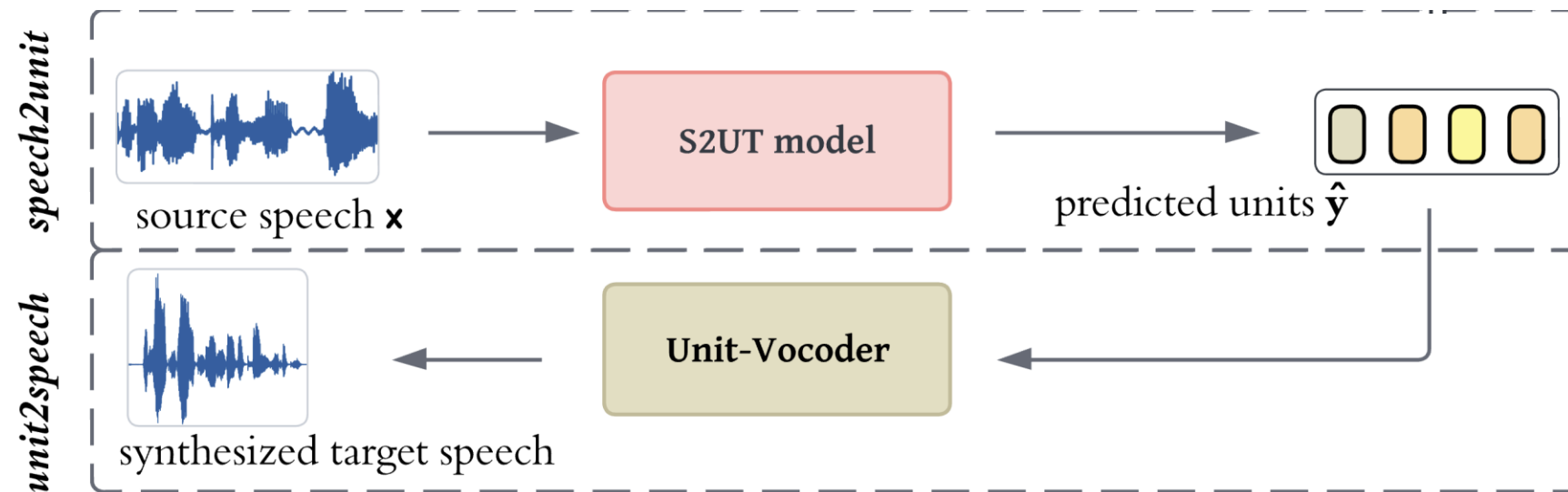
- S2UT (speech-to-unit translation): Convert source speech into target speech units
- Unit-Vocoder: Synthesize target speech from target speech units



# Non-autoregressive Speech-to-Speech Translation

## S2UT Model

- Transformer/Conformer-based
- Non-autoregressive Transformer (NAT): Masked-Predict Language Model



# Non-autoregressive Speech-to-Speech Translation

## S2UT Backbone: CMLM [1]

- Source encoded by Transformer/Conformer-Encoder
- Target units predicted by Transformer-Decoder non-autoregressively
  - Use Iterative Refinement during decoding
  - Tokens of all positions are predicted and tokens with TopK prob are kept for next iteration

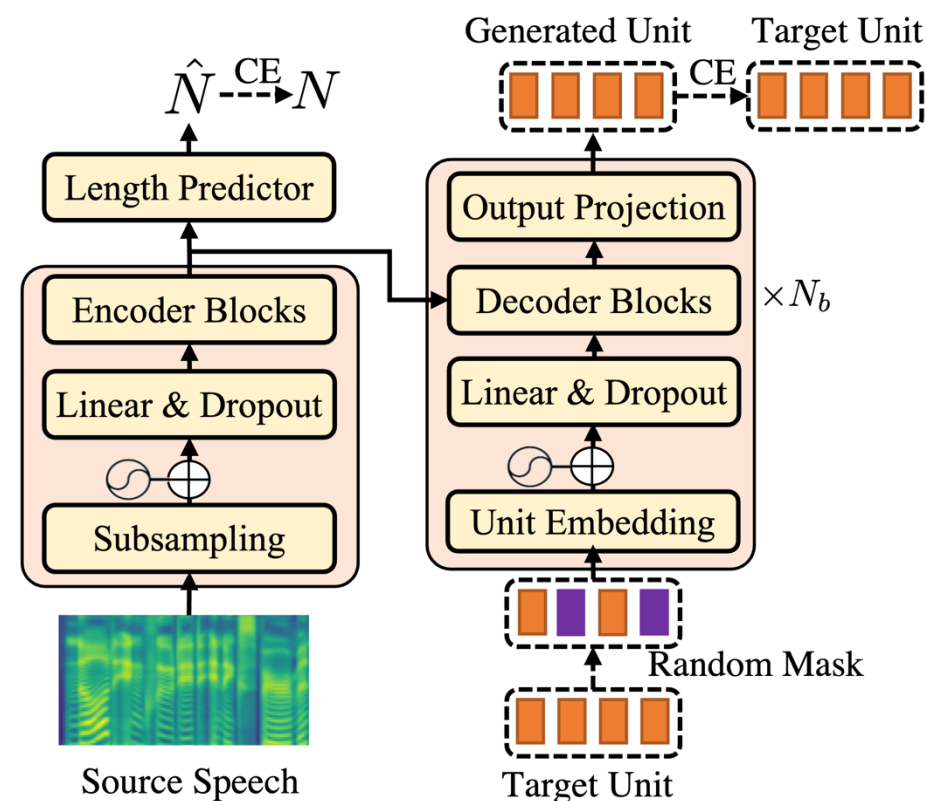


Figure from [2]

[1] Ghazvininejad et al., (2019). Mask-Predict: Parallel Decoding of Conditional Masked Language Models

[2] Huang et al., (2023). TRANSPEECH: SPEECH-TO-SPEECH TRANSLATION WITH BILATERAL PERTURBATION

# Challenge in Non-autoregressive S2ST

## Multi-modality Problem

- Acoustic: the same content can sound differently due to acoustic conditions
- Linguistic: Multiple correct translations exist for the same source speech

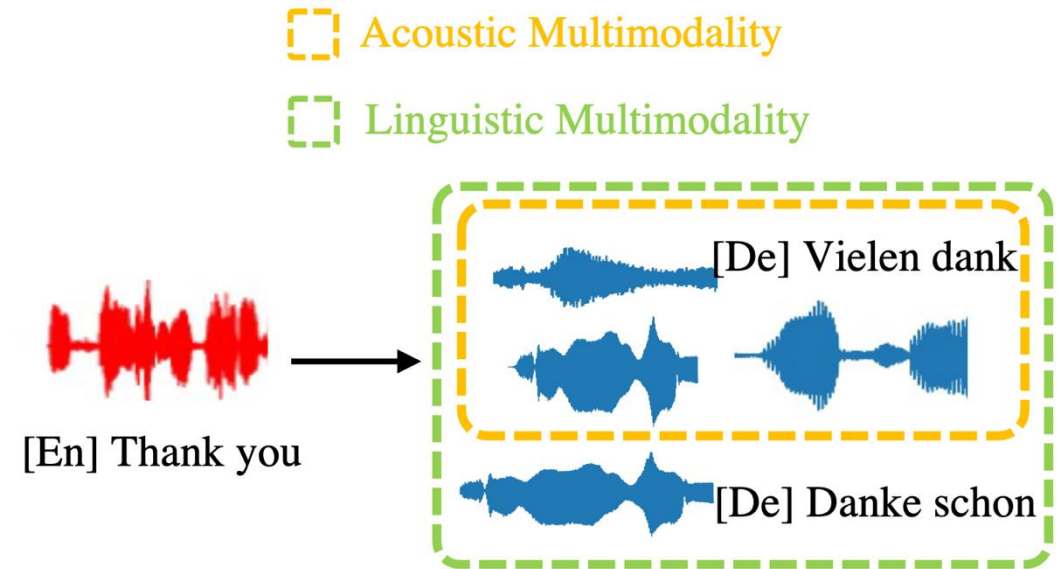
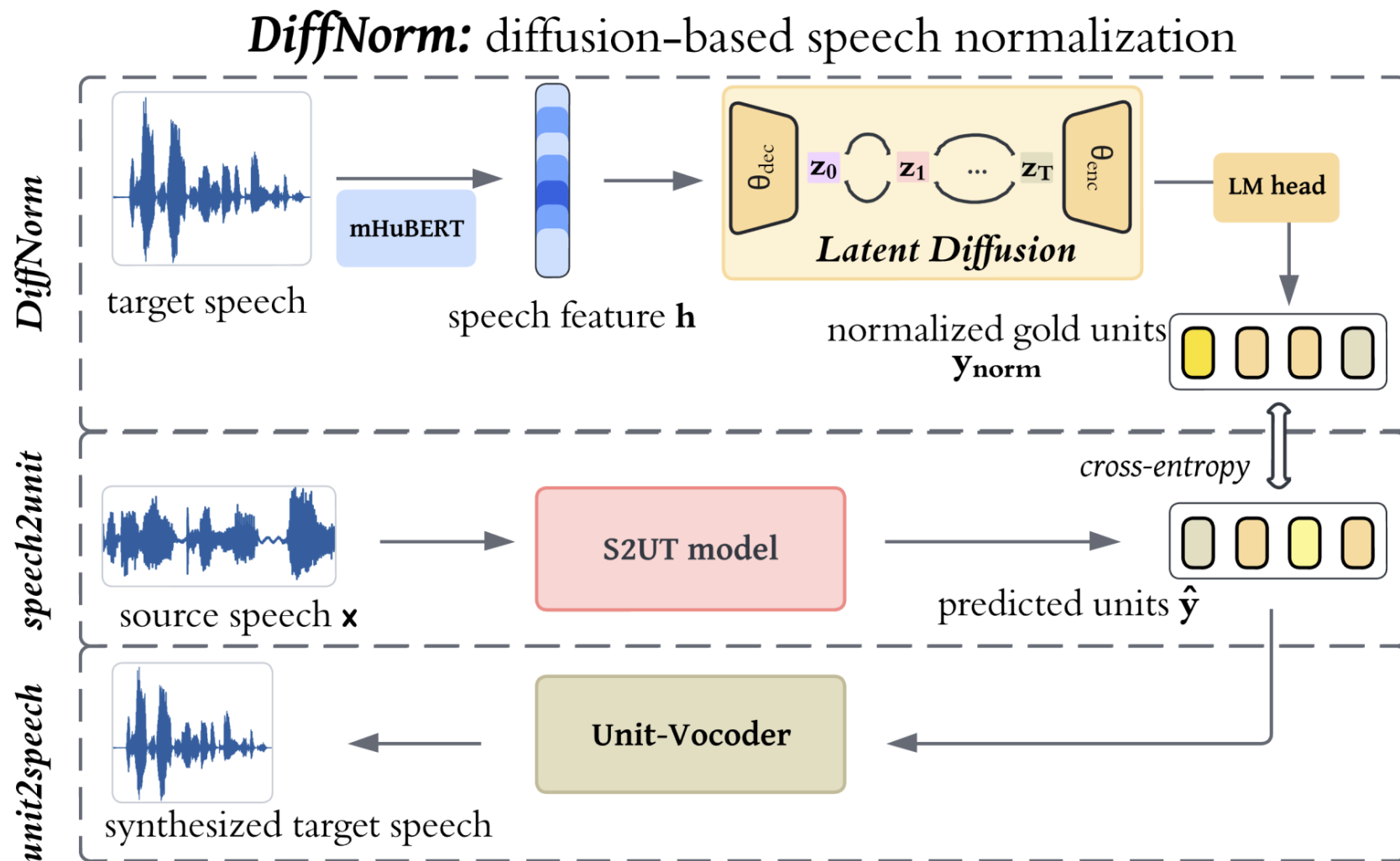


Figure from [1]

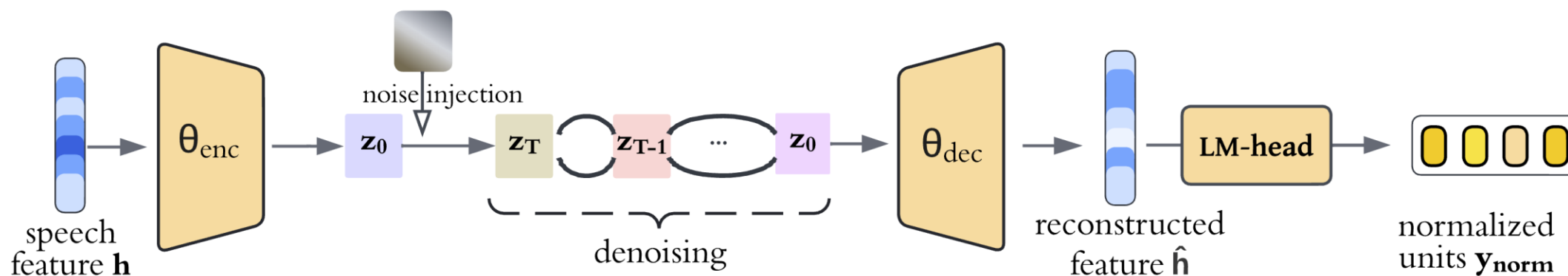
# Strategy: Speech Normalization with Diffusion



# Strategy: Speech Normalization with Diffusion

## Construct Normalized Speech Units:

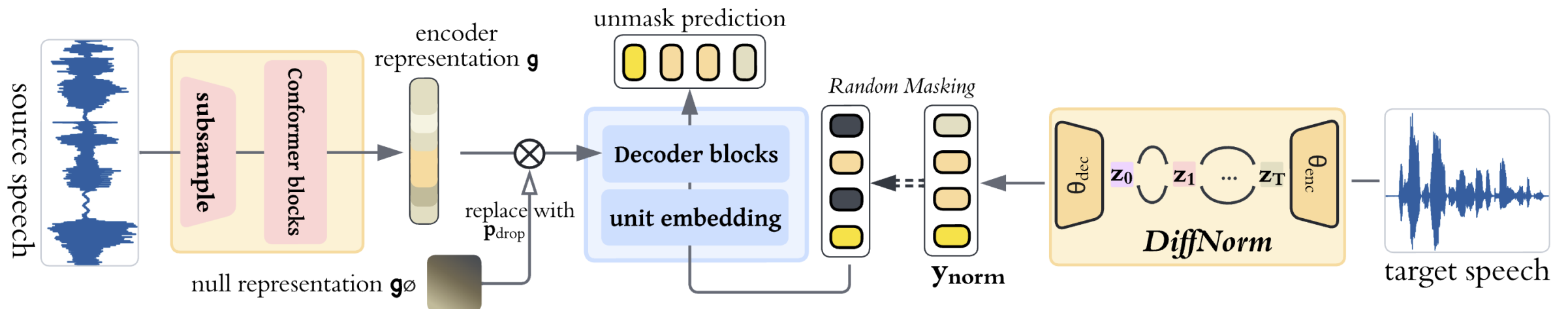
- Train VAE model on target speech feature
- Train Diffusion Model on VAE latents
- Units Construction:
  - Choose a start time  $T$  to inject noise into the clean latents ( $z_0 \rightarrow z_T$ )
  - Denoise with pre-trained Diffusion Model and reconstruct feature
  - Predict normalized speech units with reconstructed feature



# CMLM with DiffNorm Units

## Training with classifier-guidance (adapted from Diffusion to NAT)

- Randomly replace source representation with null representation
- Improve decoder's iterative decoding quality, especially for long-sequences





# Selected Experiment Results

<i>ID</i>	<i>System</i>	<i>Quality</i> $\uparrow$		<i>Inference Speed</i> $\uparrow$	
		En-Es	En-Fr	Speed	Speedup
<b>Autoregressive</b>					
1	Transformer <sup>†</sup> [30]	10.07	15.28	870	1.00 $\times$
2	Norm Transformer <sup>†</sup> [31]	12.98	15.93	870	1.00 $\times$
3	Conformer <sup>†</sup>	13.75	17.07	895	1.02 $\times$
<b>Non-autoregressive Model</b>					
4	CMLM	12.58	15.62	4651	5.34 $\times$
5	CMLM + BiP <sup>†</sup> [20]	12.62	16.97		
<b>Our Improved Non-autoregressive Model</b>					
6	CMLM + DIFFNORM	18.96	17.27	4651	5.34 $\times$
7	CMLM + CG <sup>‡</sup>	17.06	16.89		
8	CMLM + DIFFNORM + CG <sup>‡</sup>	<b>19.49</b>	<b>17.54</b>		

Table 2: Comparison of speech-to-speech models evaluated by quality (ASR-BLEU) and speed (units/seconds). Results with <sup>†</sup> are taken from the prior work [20]. <sup>‡</sup> We use  $w = 0.5$  for CG. **Our NAT models achieve superior translation quality while maintaining their fast inference speed.**

# Selected Experiment Results

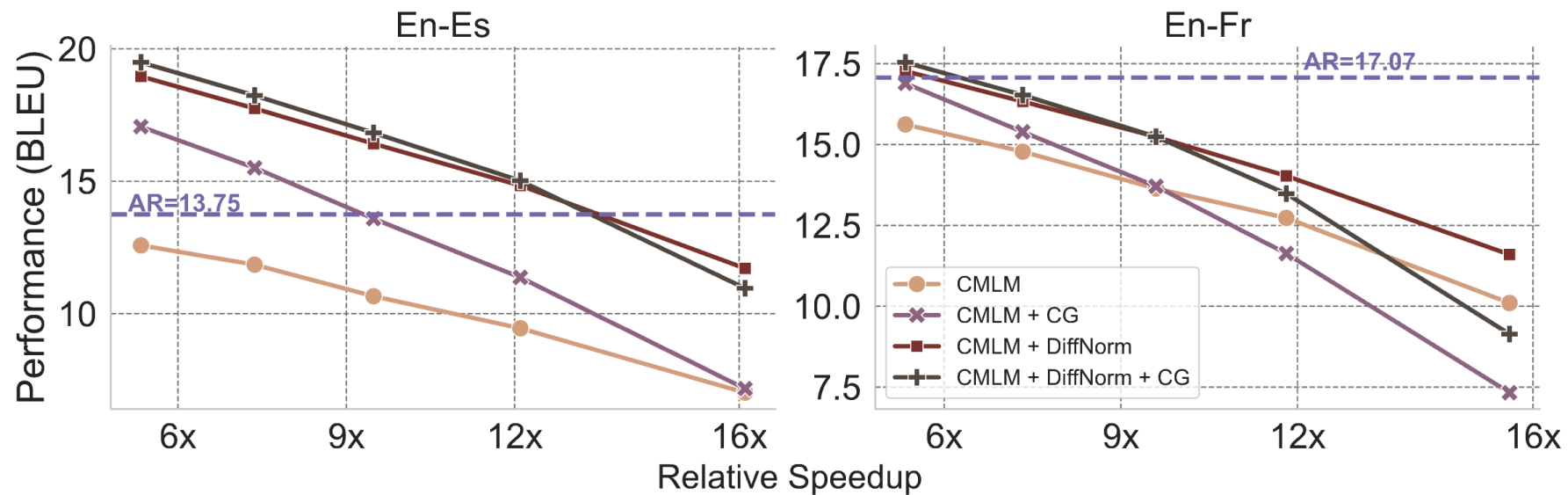


Figure 4: Trade-off between quality (ASR-BLEU) and latency for varying numbers of decoding iterations. Five markers correspond to {15, 10, 7, 5, 3} decoding iterations. Decreasing the number of iterations results in a decline in model performance, traded off for faster speedup. With DIFFNORM and CG, **our S2UT model achieves a better quality-latency trade-off** than CMLM and outperforms a strong autoregressive baseline with large speedups.

# THANK YOU!

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Please feel free to reach out to me with questions/suggestions at [wtan12@jhu.edu](mailto:wtan12@jhu.edu)