

# 语音翻译：从前沿研究到产品创新

## Speech Translation

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字节跳动人工智能实验室

2021/6/6

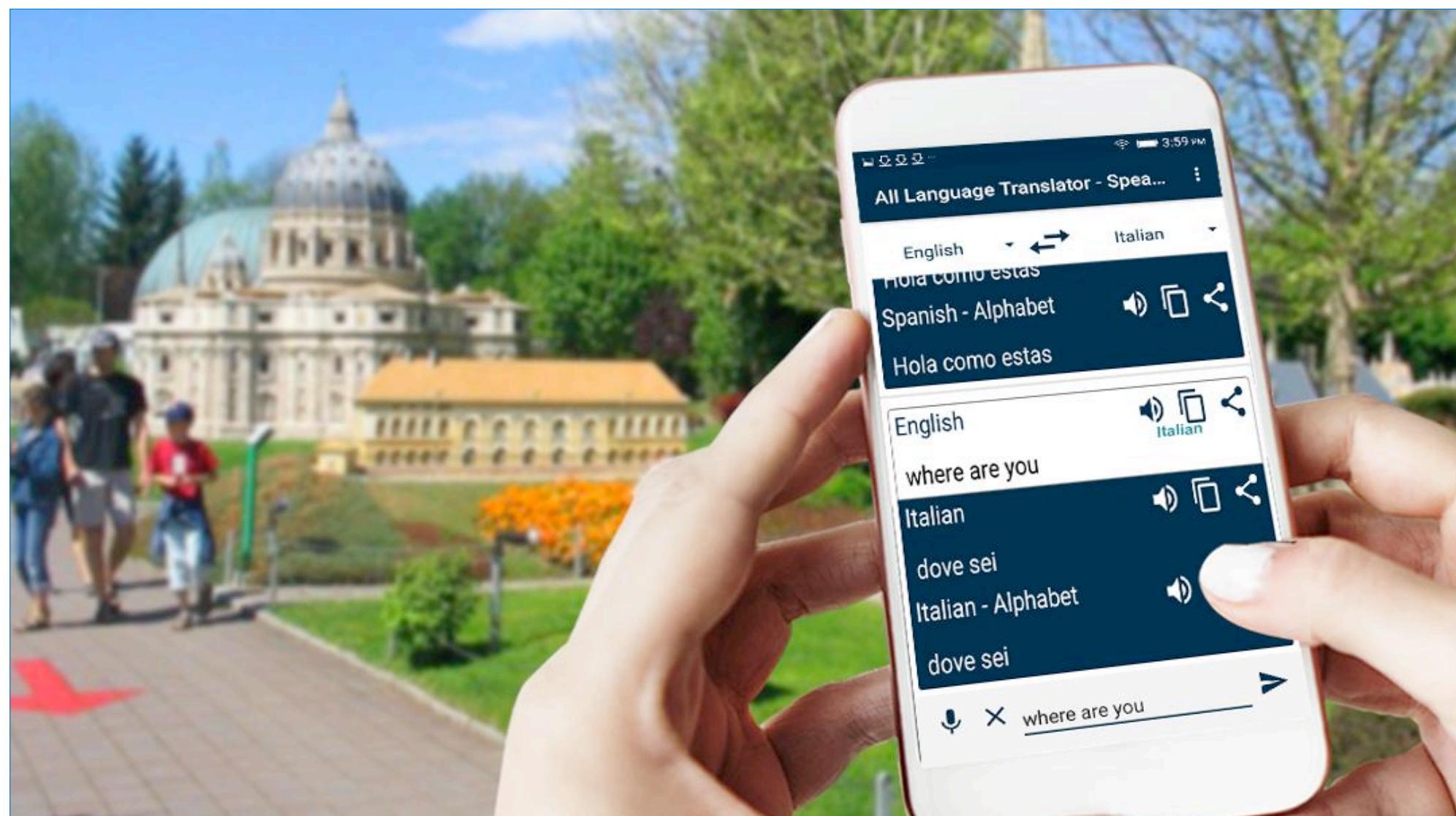
# Cross Language Barrier with Machine Translation



Foreign Media



Global Conferences



Tourism



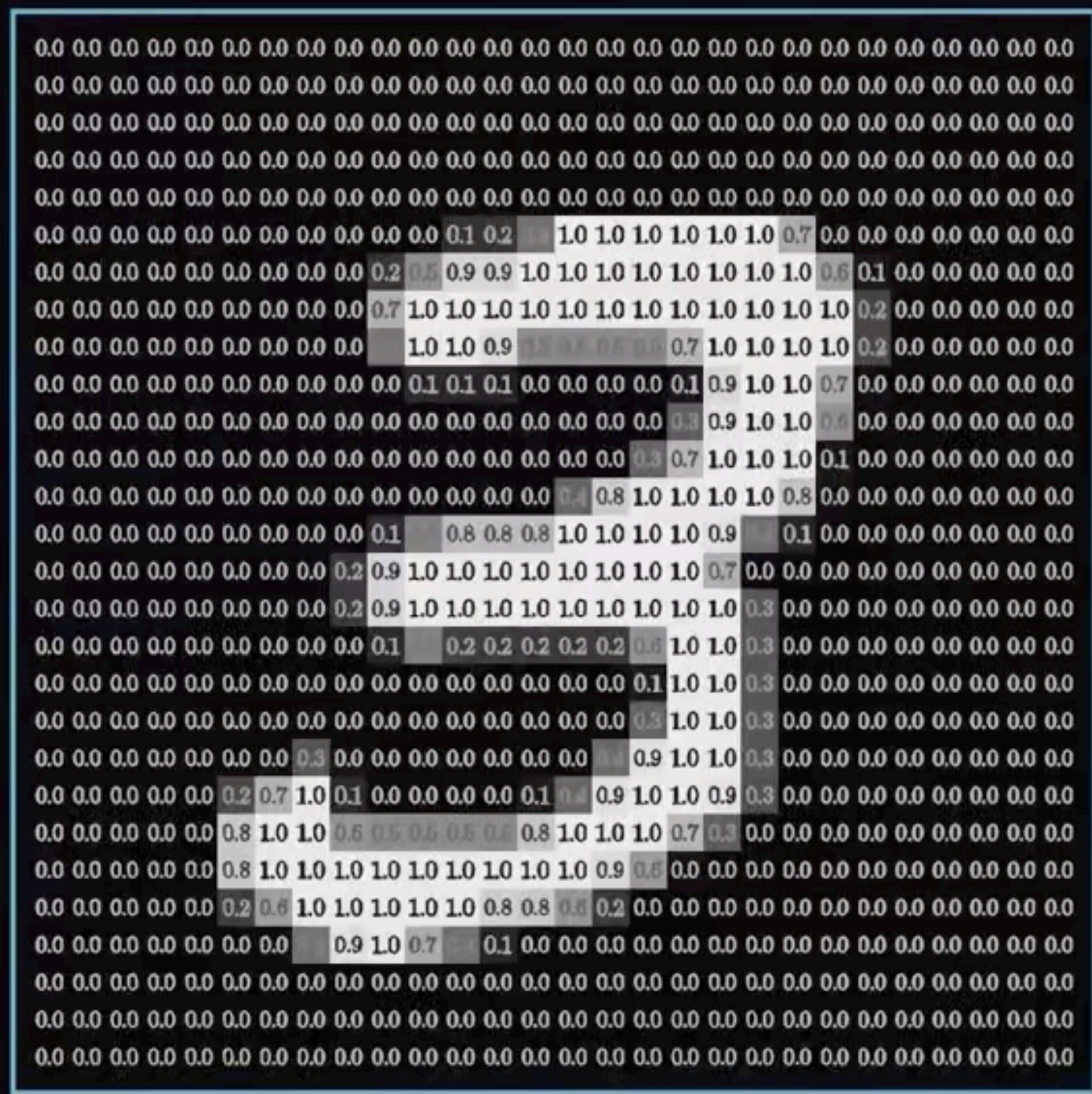
International Trade



火山翻译



# 西瓜视频



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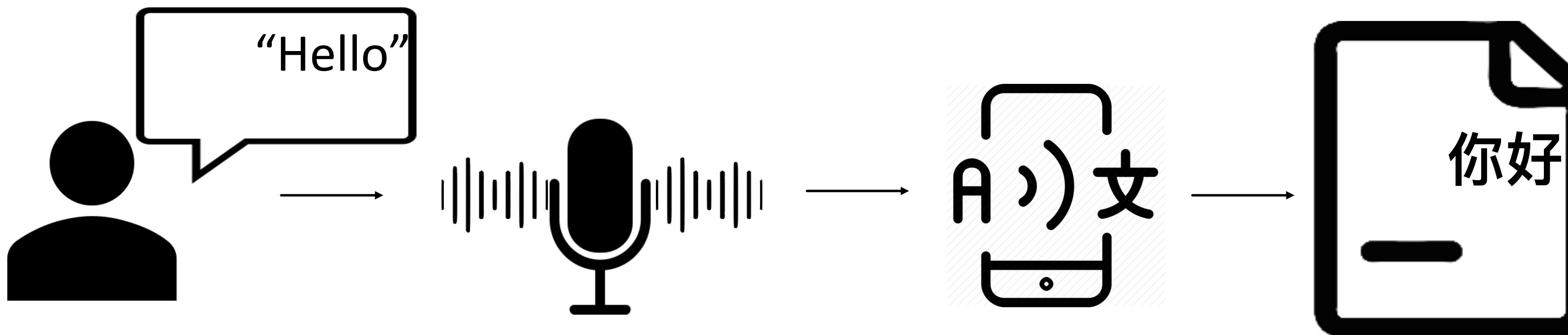
# Outline

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1. Overview: ST Problem and Challenge
2. What is a better model for ST?
3. Better training strategy for ST?
4. New ST-powered Products

# Speech-to-Text Translation(ST)

- source language **speech(audio)** → target lang **text**



## Application Type

- (Non-streaming) ST  
非流式语音翻译
- Streaming ST  
流式语音翻译

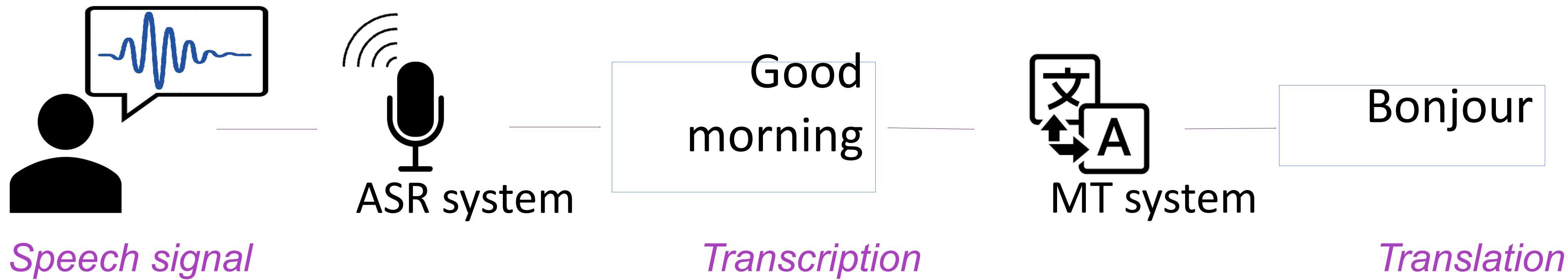
## System

- Cascaded ST  
级联语音翻译
- End-to-end ST  
端到端语音翻译

# Cascaded ST System

## - Challenges:

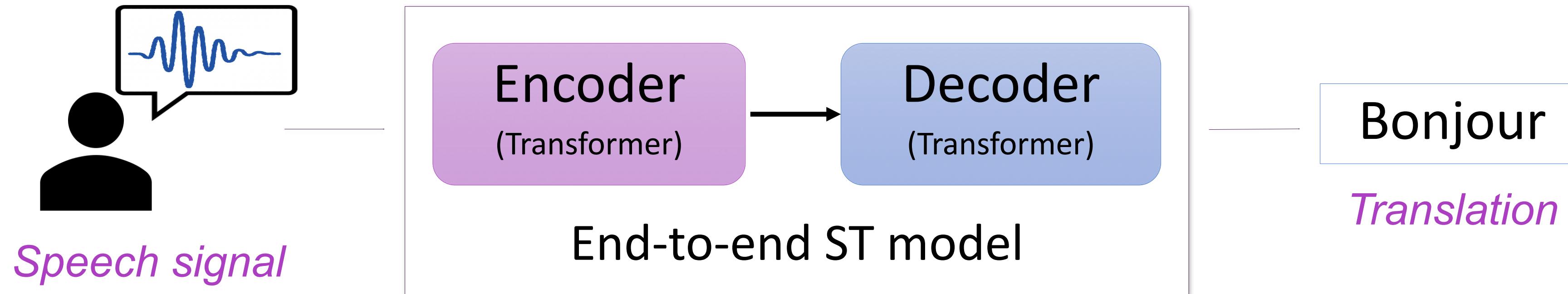
1. Computationally inefficient
2. Error propagation: Wrong transcription  Wrong translation



**do at this** and see if it works for you  这样做，看看它是否对你有用

**duet this** and see if it works for you  二重奏一下，看看它是否对你有用

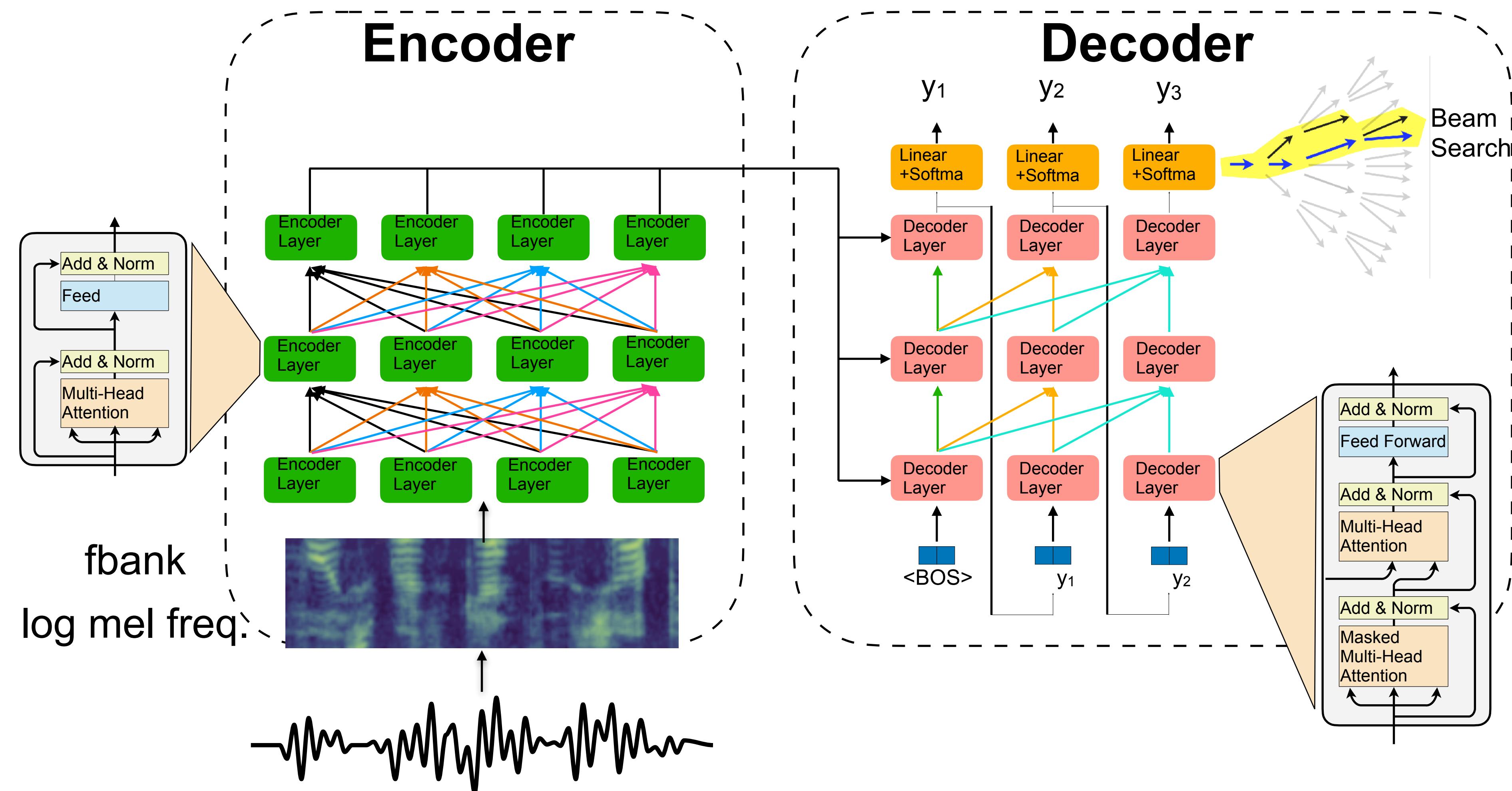
# End-to-end ST Model



- Single model to produce text translation from speech
- Popular model: Encoder-Decoder architecture (e.g. Transformer)
- Advantage:
  - Reduced latency, simpler deployment
  - Avoid error propagation

# Basic Speech Translation Architecture (Same as MT)

Transformer-based: N-layer encoder, M-layer decoder



# Challenge

- Data scarcity - lack of large parallel corpus
- Modality disparity between audio and text
- Require low latency for product serving

# Approaches for End-to-end ST

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- Model
  - Better Encoder: **LUT** [AAAI 2021a] **Chimera** [ACL 2021a]
  - Better Decoder: **COSTT** [AAAI 2021b]
- Training technique
  - Audio pre-training: Wave2Vec2.0 [Baevski et al 2021]
  - Progressive multi-task training: **XSTNet** [Interspeech 2021]
- Speed-up Inference (not in this talk)
  - Parallel Decoding: **GLAT** [ACL 2021b]
  - GPU optimization: **LightSeq** [NAACL2021]

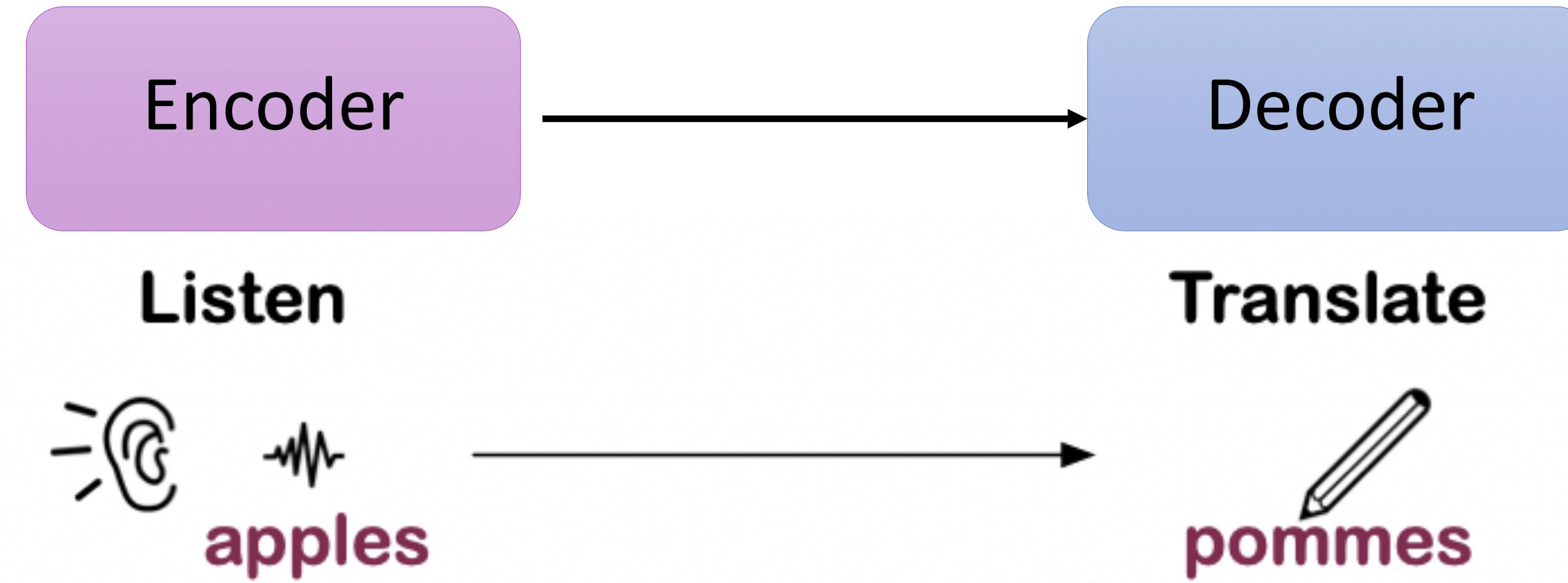


# Listen, Understand and Translate: Triple Supervision Decouples End-to-end Speech-to-text Translation

Qianqian Dong, Rong Ye, Mingxuan Wang, Hao Zhou, Shuang Xu, Bo Xu, Lei Li



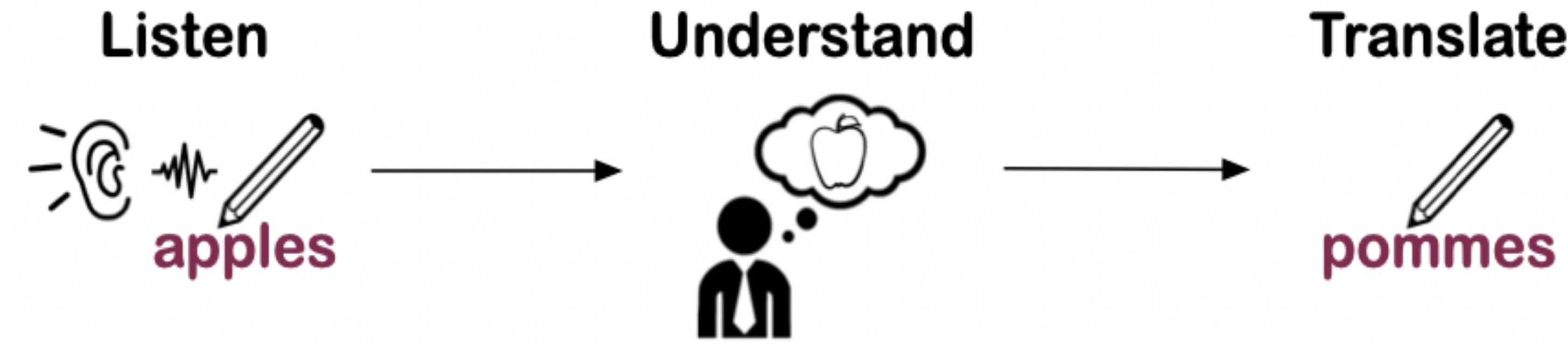
# Drawbacks of the Encoder-Decoder Structure



1. A **single** encoder is hard to capture the representation of audio for the translation.
2. Limited in utilizing the information of "*transcription*" in the training.

# Motivation: Mimic human's behavior

Question: How human translate?

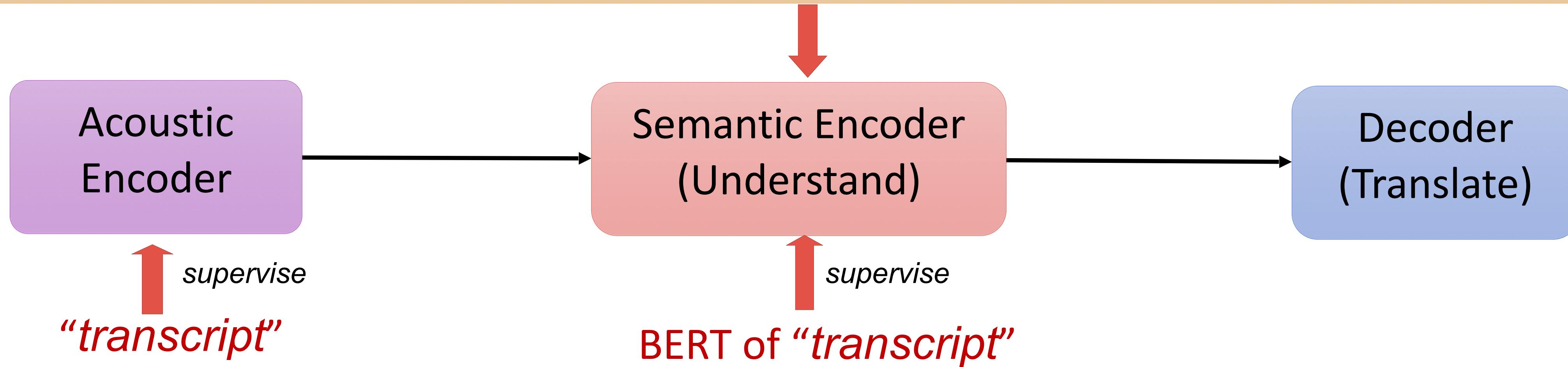


“Listen-Understand-Translate”(LUT) model based motivated by  
human’s behavior

# Motivation of Better Encoding

**Drawback 1:** A single encoder is not enough.

**Idea 1:** Introduce a **semantic encoder**



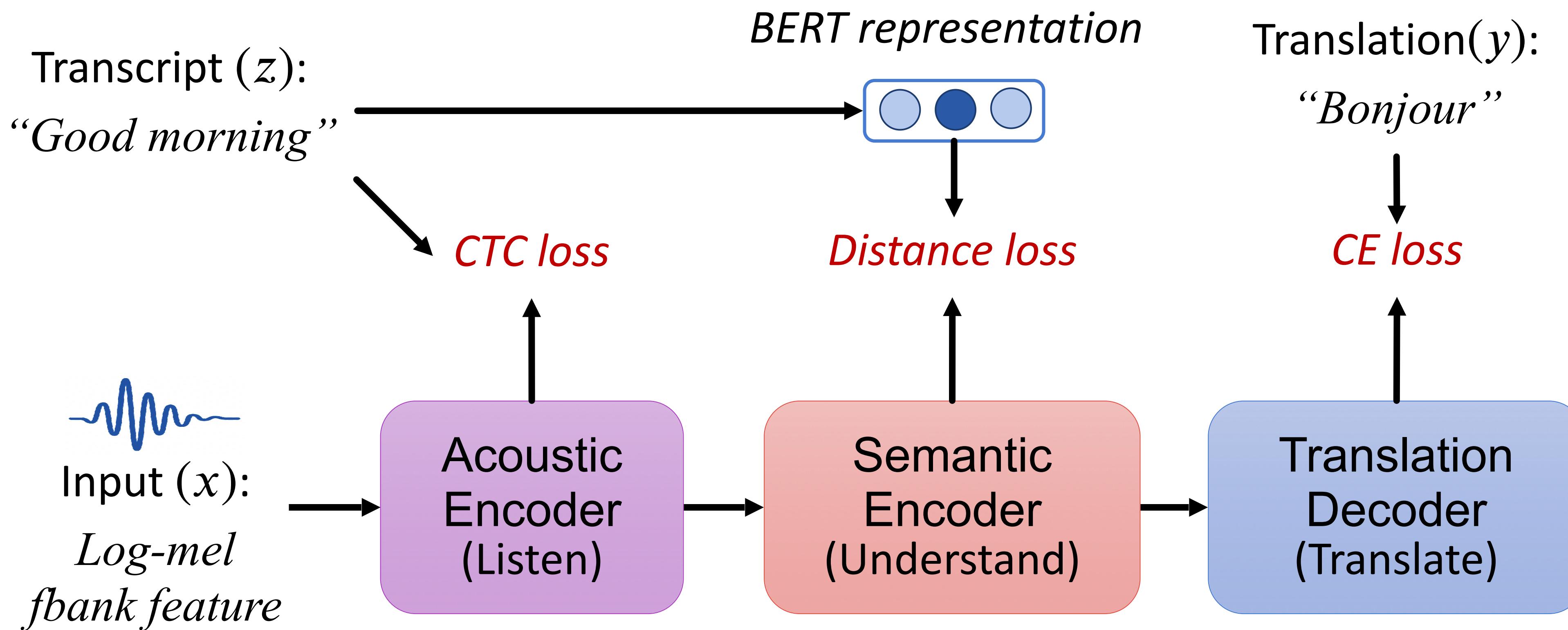
**Drawback 2:** Limit in using “transcript” info.

**Idea 2:** Utilizing the **pre-trained representation** (e.g. BERT) of the “transcript” to learn the semantic feature.

# LUT for End-to-end ST

Training data: triples of

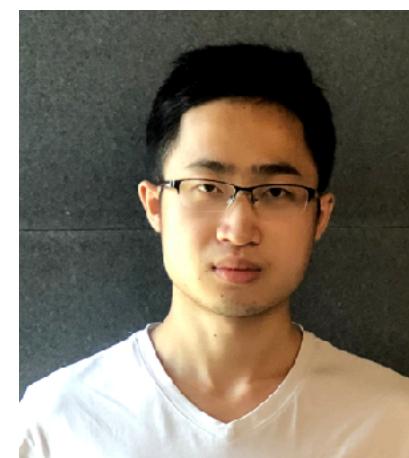
<speech, transcript\_text, translate\_text>





# Learning Shared Semantic Space for Speech-to-Text Translation

Chi Han, Mingxuan Wang, Heng Ji, Lei Li

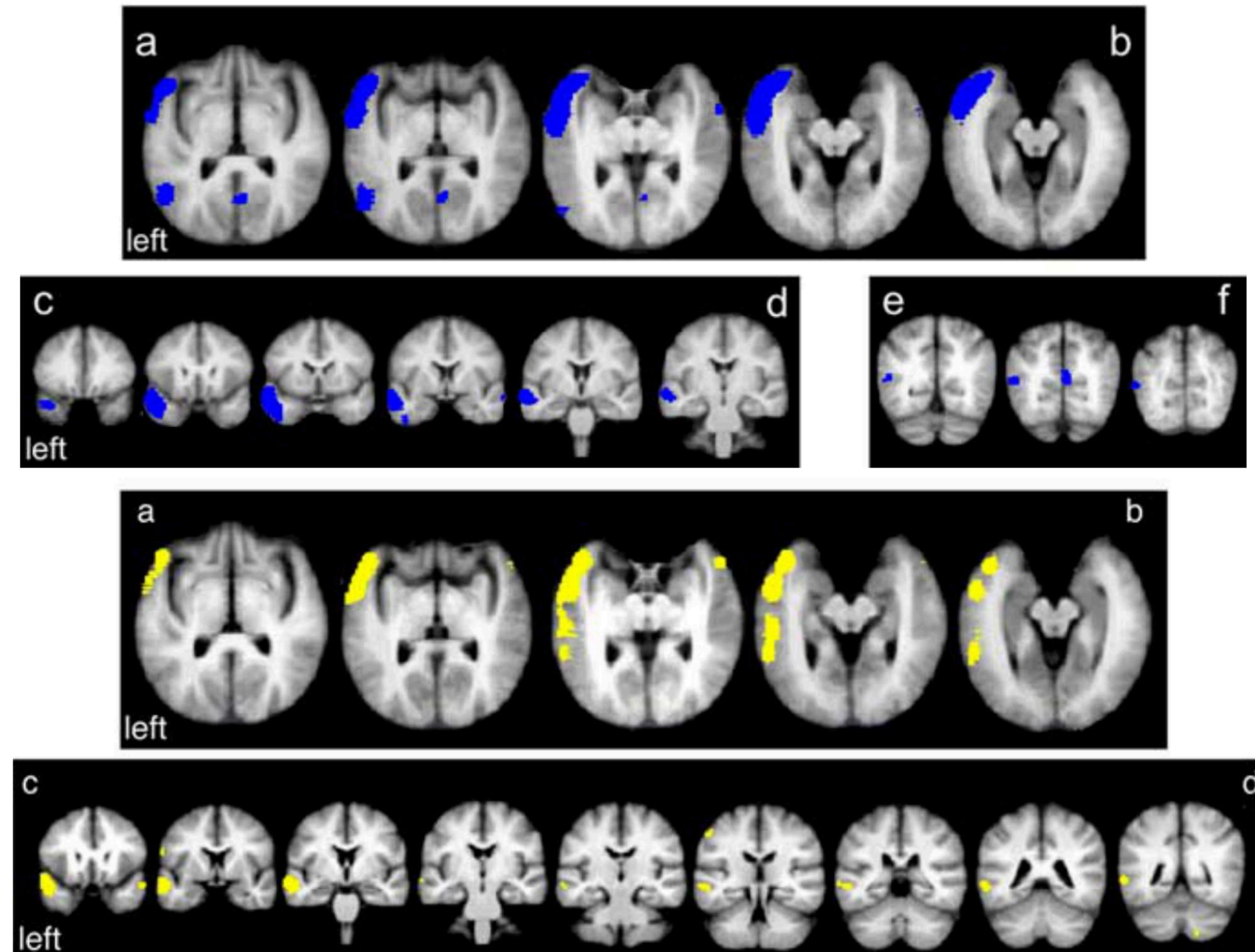


Paper: <https://arxiv.org/abs/2105.03095>

Code: <https://github.com/Glaciohound/Chimera-ST>

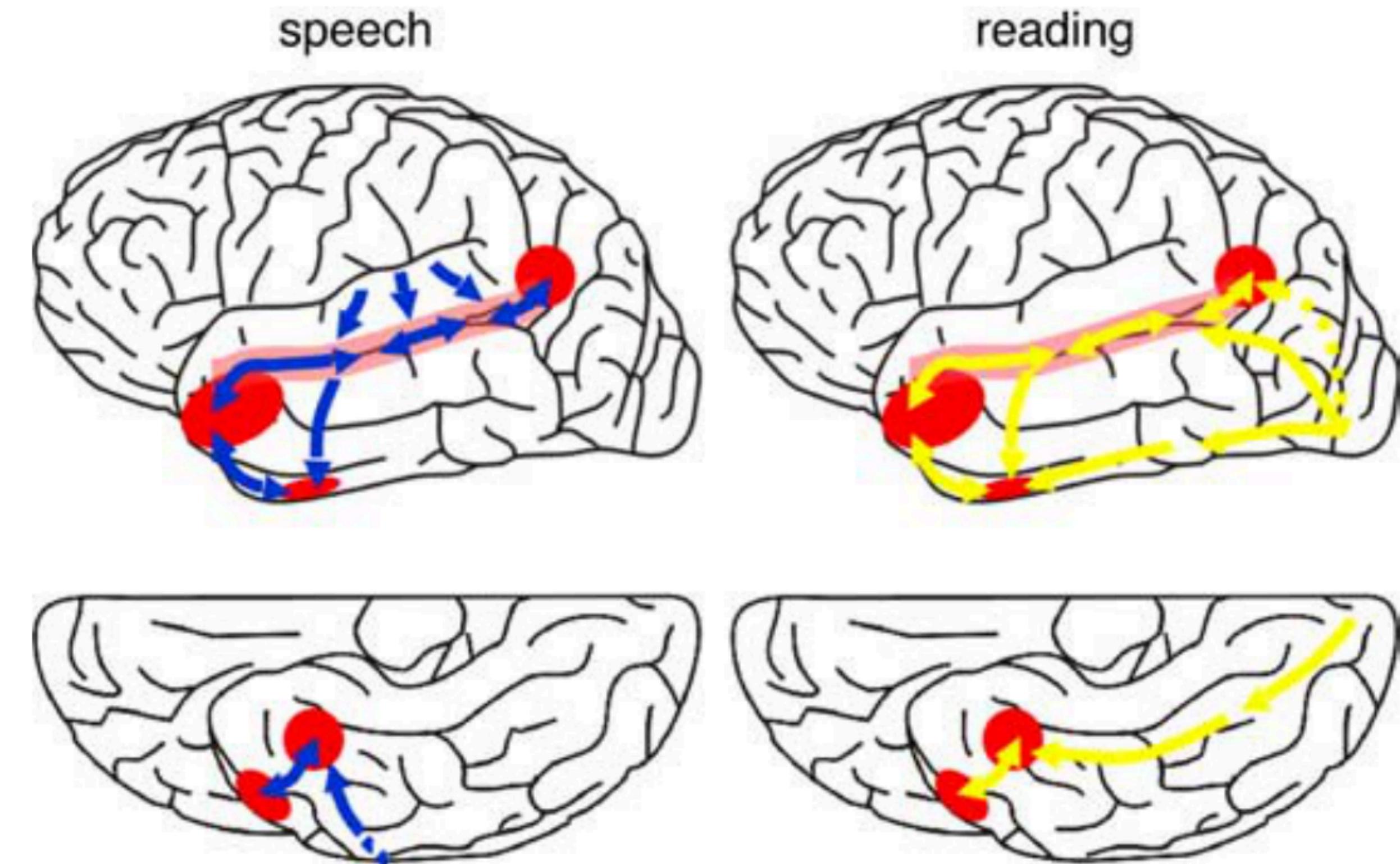
# Insights from Cognitive Neuroscience

Speech and text interfere with each other in brain<sup>[1]</sup>



activation map

Convergence sites of speech (blue) and text (yellow)



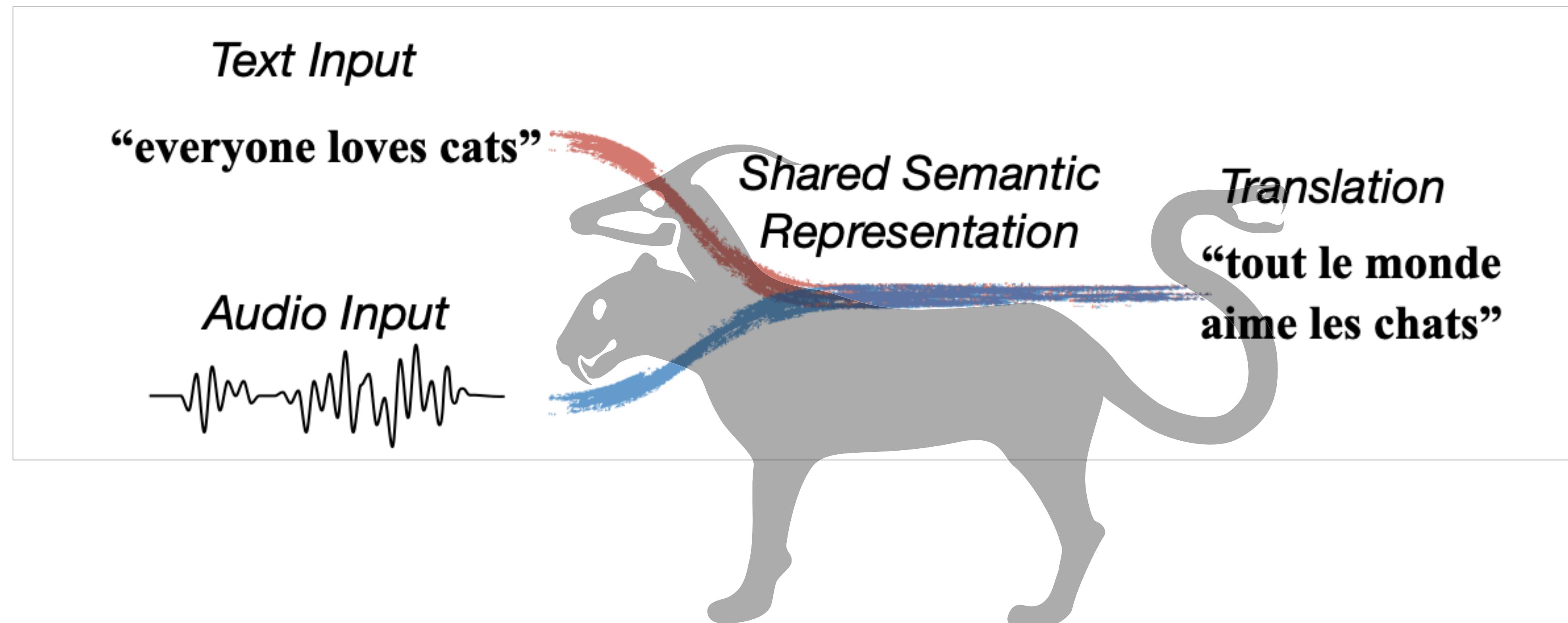
[1] Van Atteveldt, Nienke, et al. "Integration of letters and speech sounds in the human brain." *Neuron* 43.2 (2004): 271-282.

[2] Spitsyna, Galina, et al. "Converging language streams in the human temporal lobe." *Journal of Neuroscience* 26.28 (2006): 7328-7336.

# Idea: Bridging the Speech-Text modality gap with Shared Semantic Representation

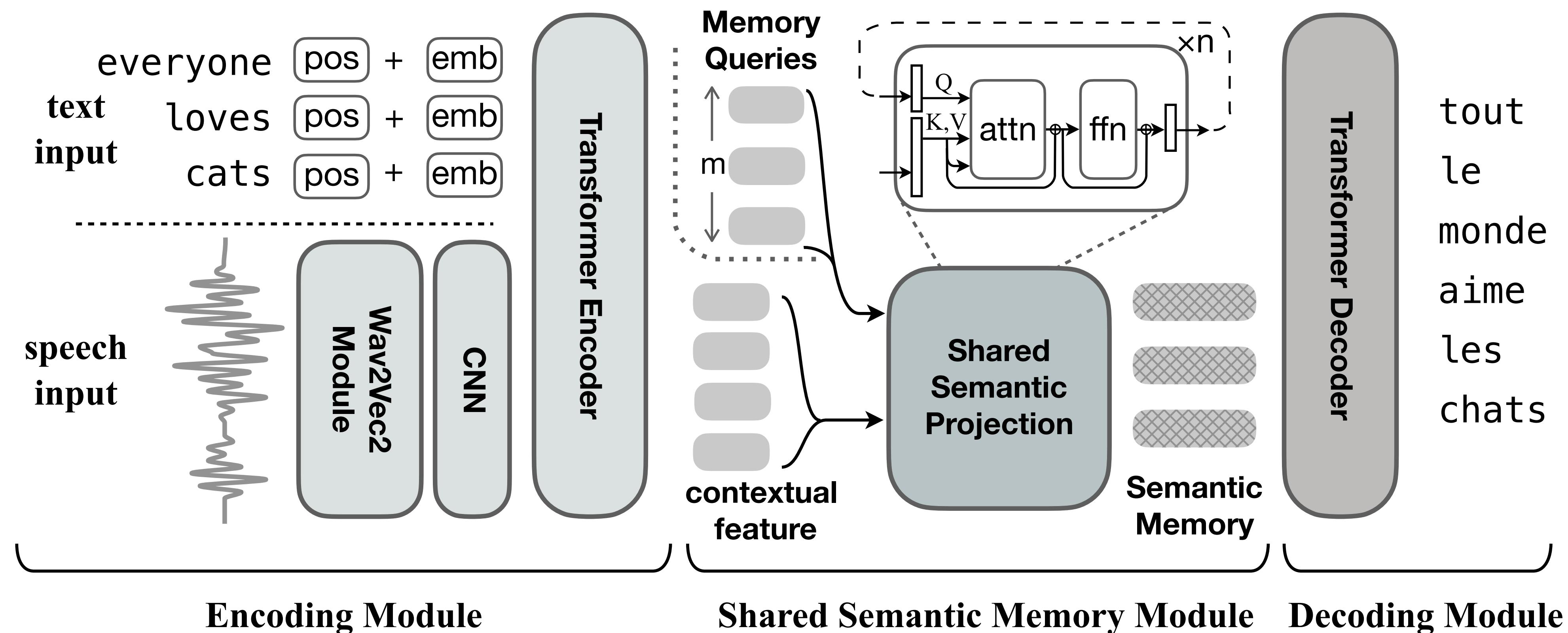
ST triple data:

<speech, transcript\_text, translate\_text>



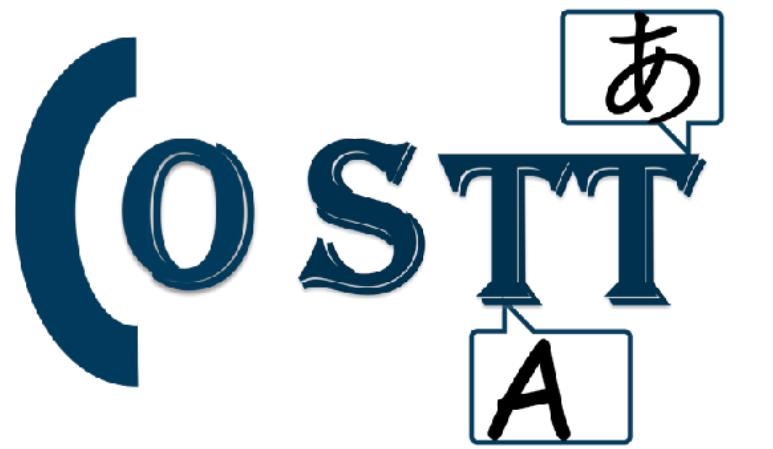
# Chimera Model for ST

Training with auxiliary objectives: ST + MT + Contrastive loss  
Benefit: able to **exploit large external MT data**



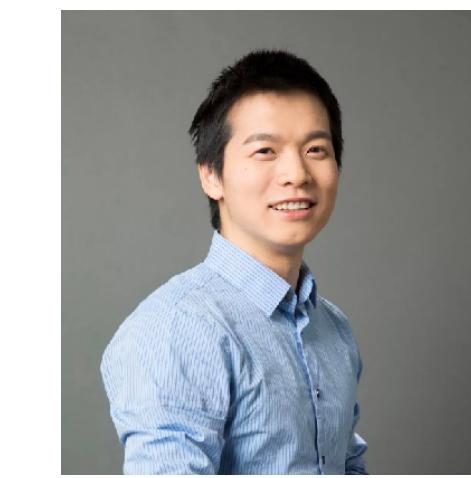
# Chimera achieves the best (so far) BLEU on all languages in MuST-C

Model	External Data			MuST-C EN-X							
	Speech	ASR	MT	EN-DE	EN-FR	EN-RU	EN-ES	EN-IT	EN-RO	EN-PT	EN-NL
FairSeq ST <sup>†</sup>	×	×	×	22.7	32.9	15.3	27.2	22.7	21.9	28.1	27.3
EspNet ST <sup>‡</sup>	×	×	×	22.9	32.8	15.8	28.0	23.8	21.9	28.0	27.4
AFS *	×	×	×	22.4	31.6	14.7	26.9	23.0	21.0	26.3	24.9
Dual-Decoder <sup>◊</sup>	×	×	×	23.6	33.5	15.2	28.1	24.2	22.9	<b>30.0</b>	27.6
STATST <sup>#</sup>	×	×	×	23.1	-	-	-	-	-	-	-
MAML <sup>ᵇ</sup>	×	×	✓	22.1	34.1	-	-	-	-	-	-
Self-Training <sup>°</sup>	✓	✓	×	25.2	34.5	-	-	-	-	-	-
W2V2-Transformer *	✓	×	×	22.3	34.3	15.8	28.7	24.2	22.4	29.3	28.2
Chimera Mem-16	✓	×	✓	25.6	35.0	16.7	30.2	24.0	23.2	29.7	28.5
Chimera	✓	×	✓	<b>27.1</b> *	<b>35.6</b>	<b>17.4</b>	<b>30.6</b>	<b>25.0</b>	<b>24.0</b>	<b>30.2</b>	<b>29.2</b>



# Consecutive Decoding for Speech-to-text Translation

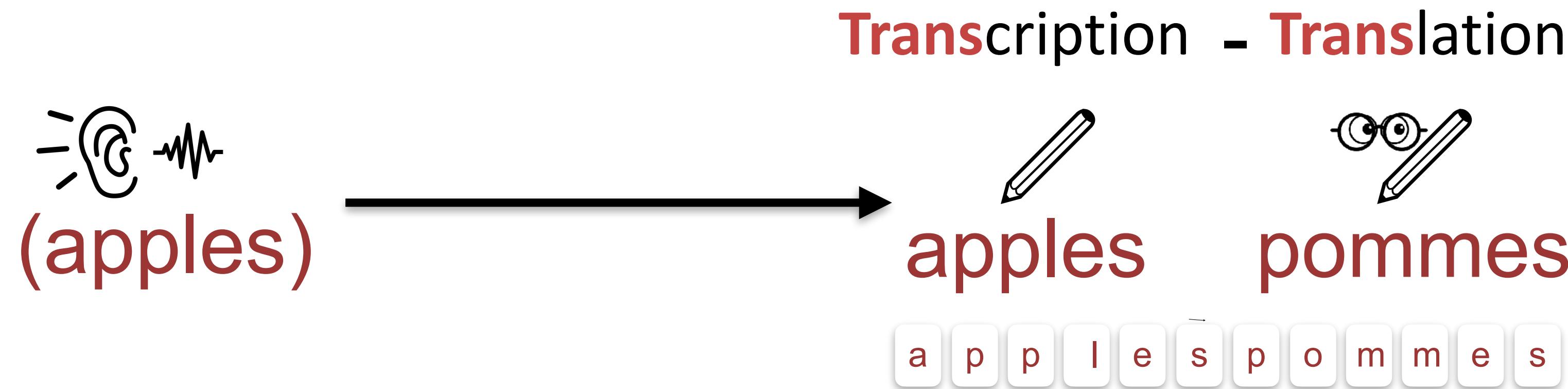
Qianqian Dong, Mingxuan Wang, Hao Zhou, Shuang Xu, Bo Xu, Lei Li



# Goal: Seamless Trans-trans



Question: How to help the model take notes like human interpreter?

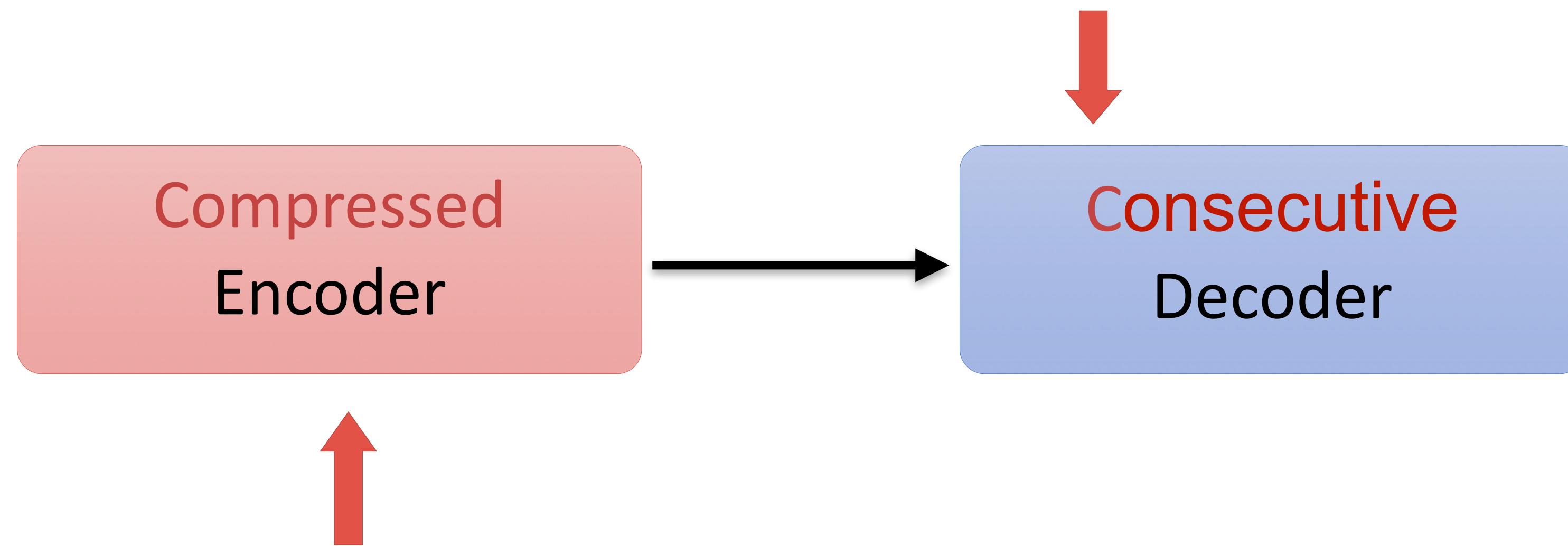


We design “COnsecutive Transcription and Translation”(COSTT) based on interpreter’s noting behavior to help the model memory.

# Motivation of Better Decoding

**Problem1:** How to give the decoder hints?

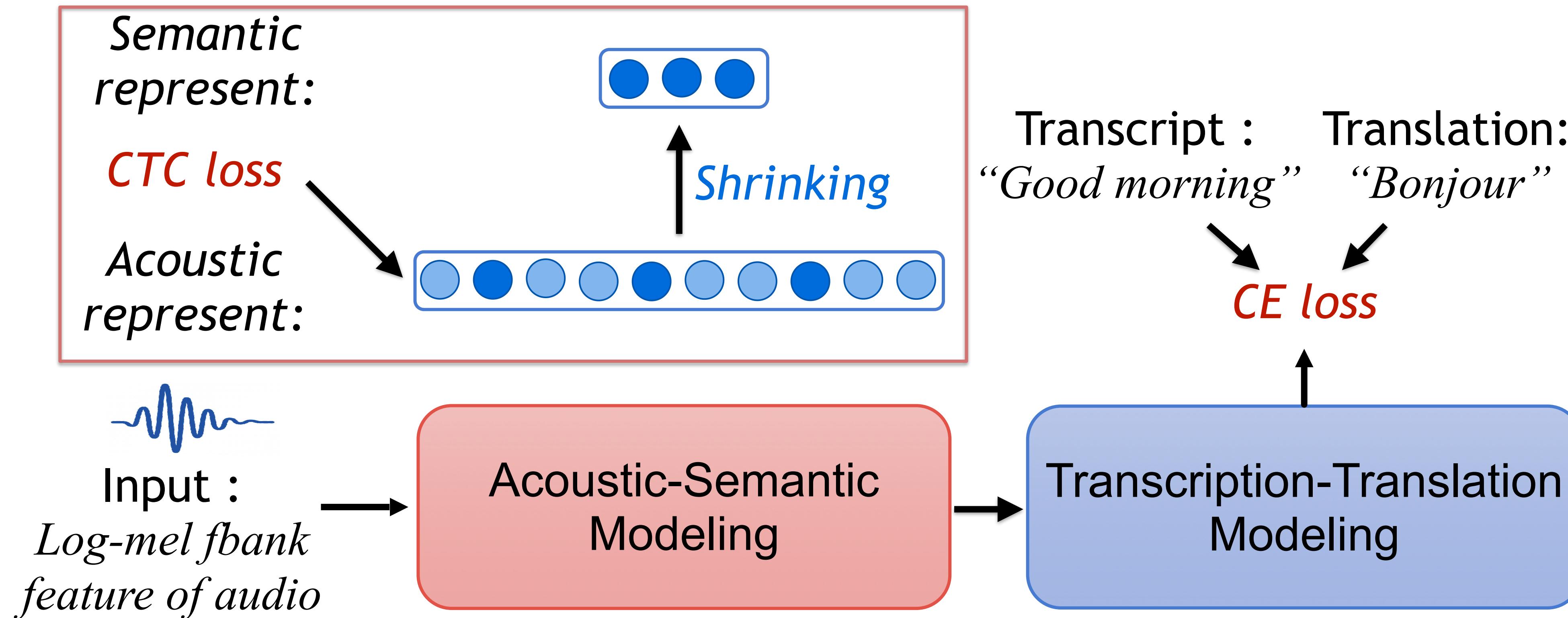
**Idea 1:** Introduce a **consecutive decoder** for trans-trans.



**Problem2:** Long acoustic sequence is challenging for the encoder!

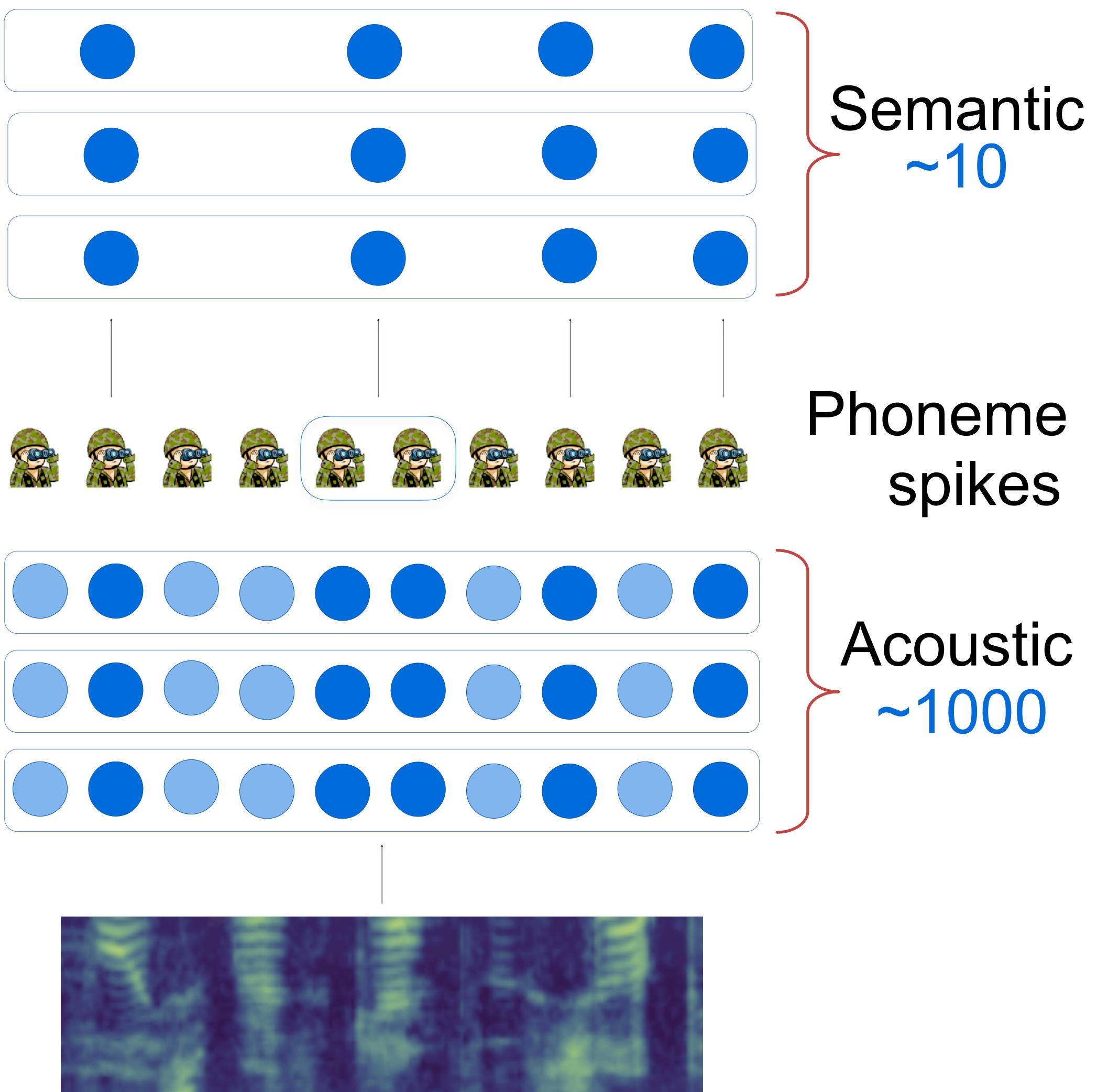
**Idea 2:** Introduce a **compressed encoder** to relief the model memory.

# COSTT for ST



# Advantages of COSTT

- Unified training with both transcript and translation text
- Reduced encoder output size with CTC-guided shrinking
- Able to pre-train the decoder with external MT parallel data



# End-to-end Speech Translation via Cross-modal Progressive Training

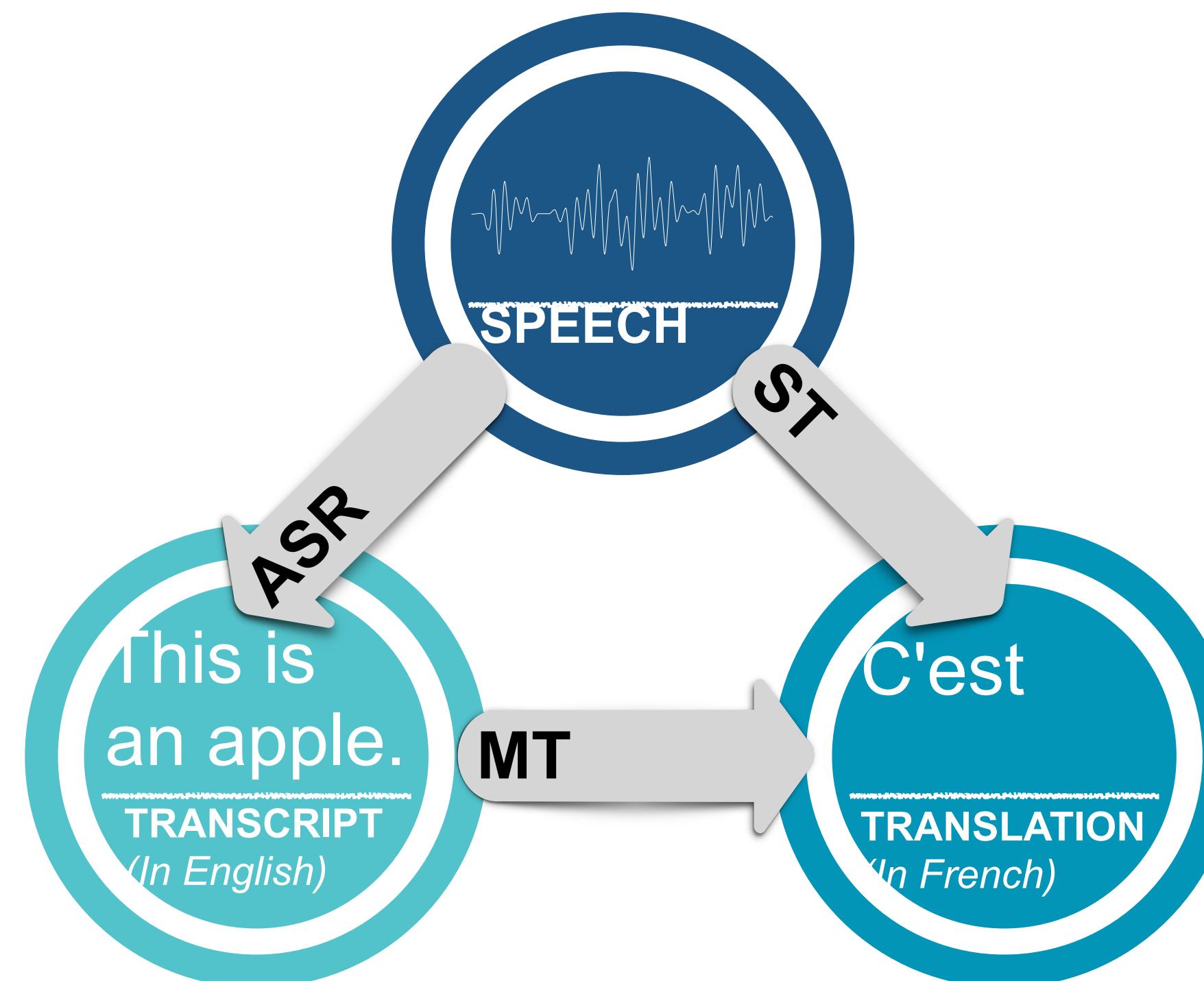
Rong Ye, Mingxuan Wang, Lei Li



- Link: <https://arxiv.org/abs/2104.10380>

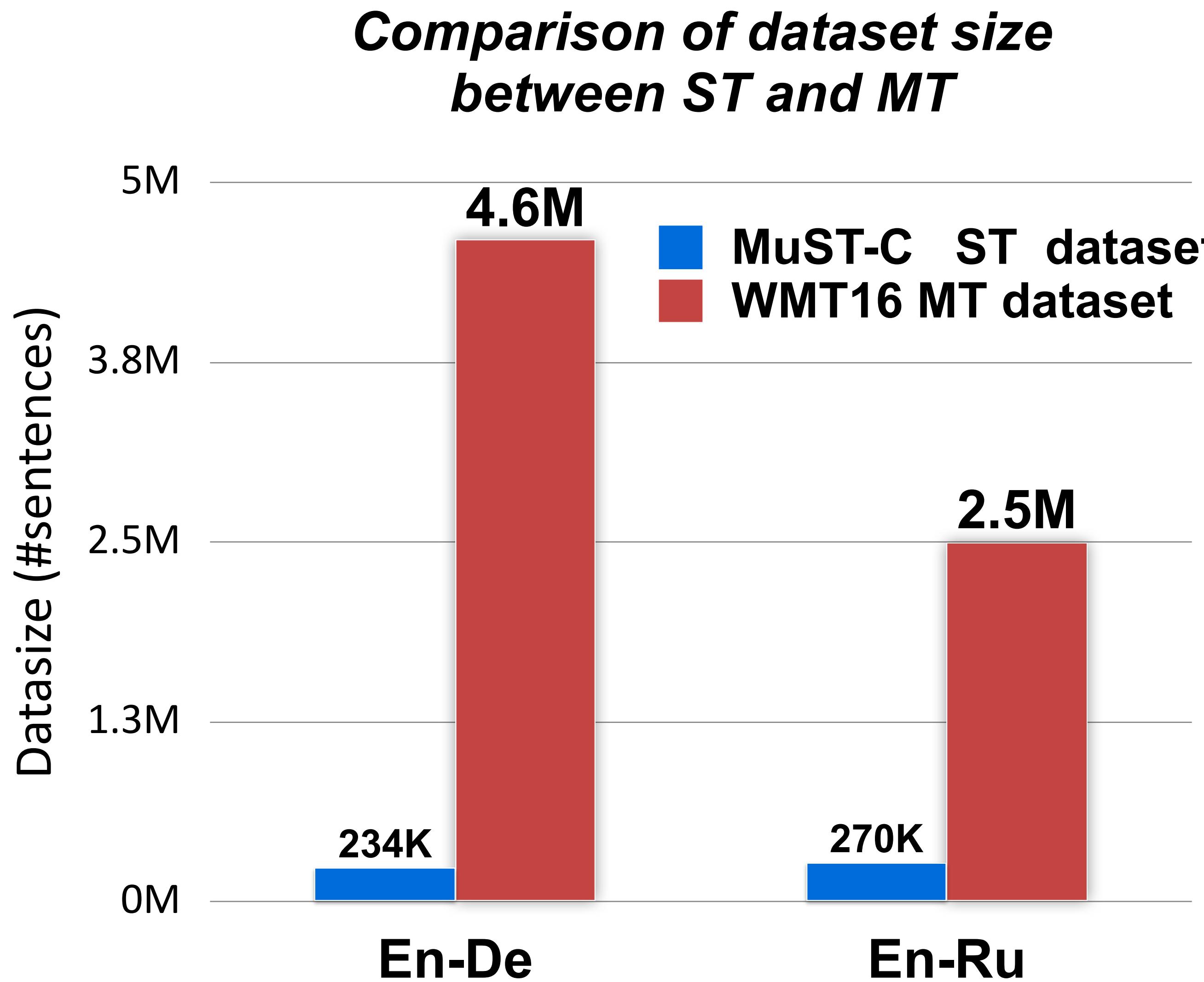
# Idea 1: Multi-task Training

Goal: To fully utilize the existing  
*<Speech, Transcript, Translation>* supervision.



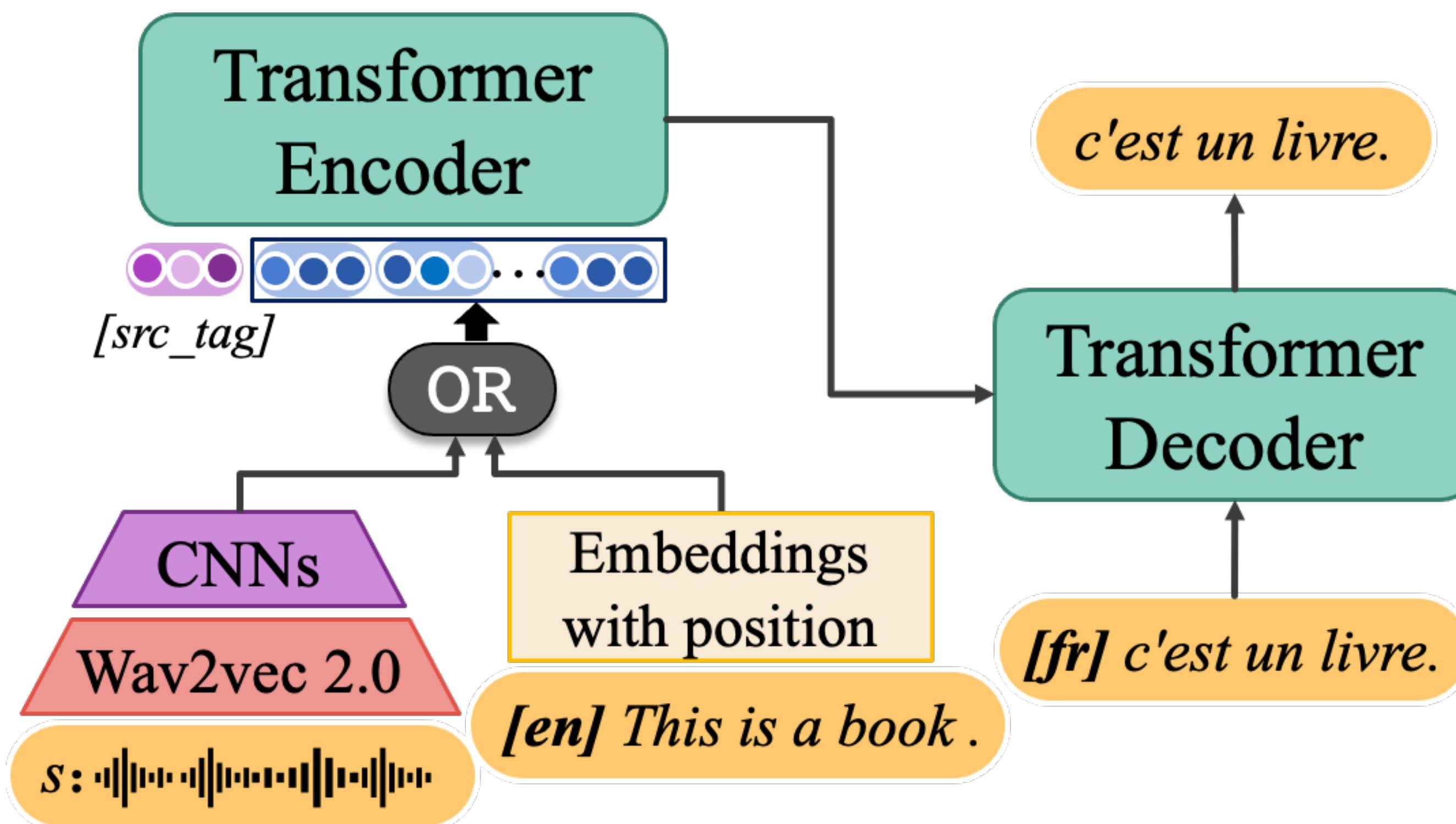
Decomposed  
into three sub-  
tasks with  
parallel  
supervision, ST,  
ASR and MT.

# Idea 2: Using large-scale MT data



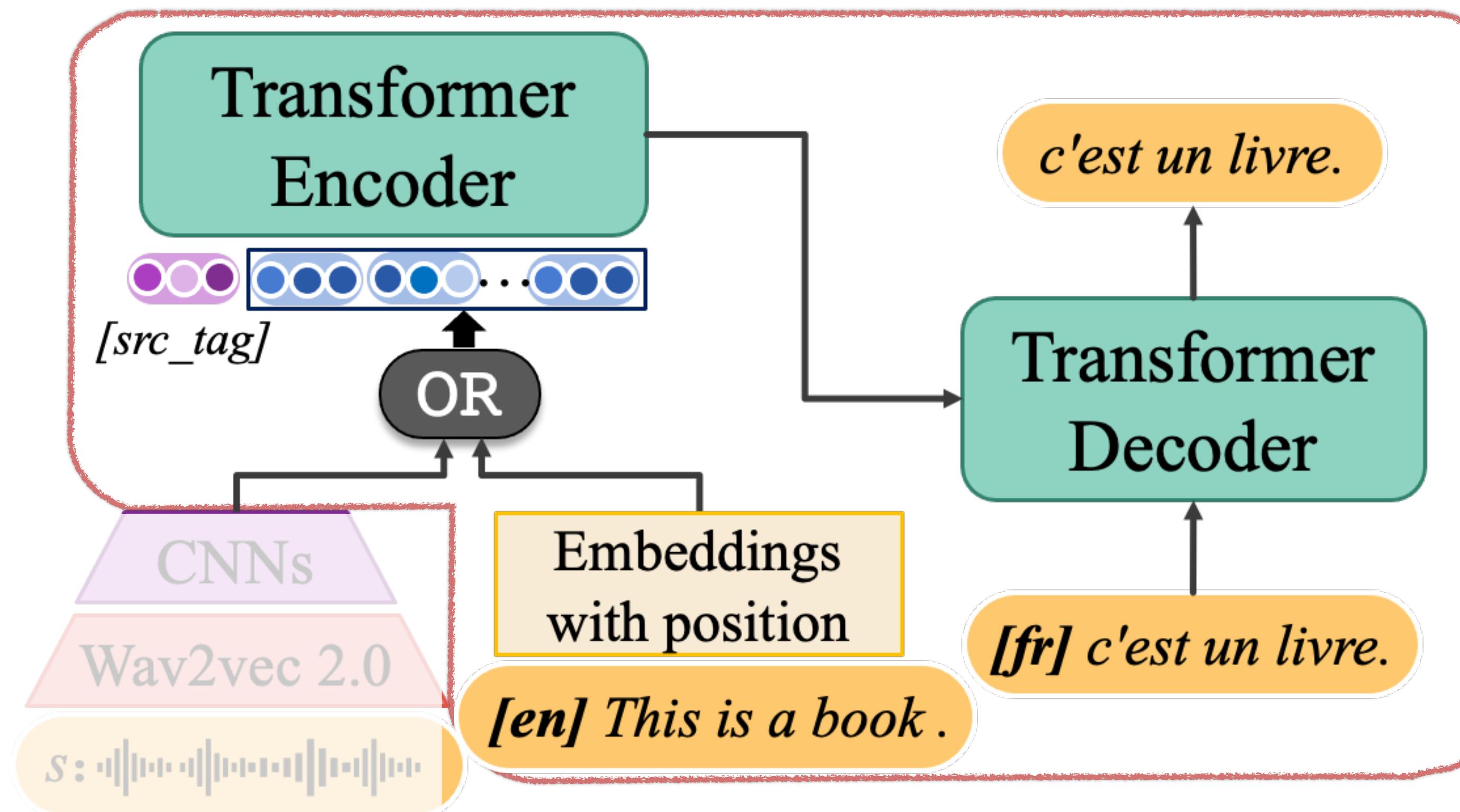
How to introduce MT data with much larger scale to improve ST performance?

# Cross Speech-Text Network (XSTNet)



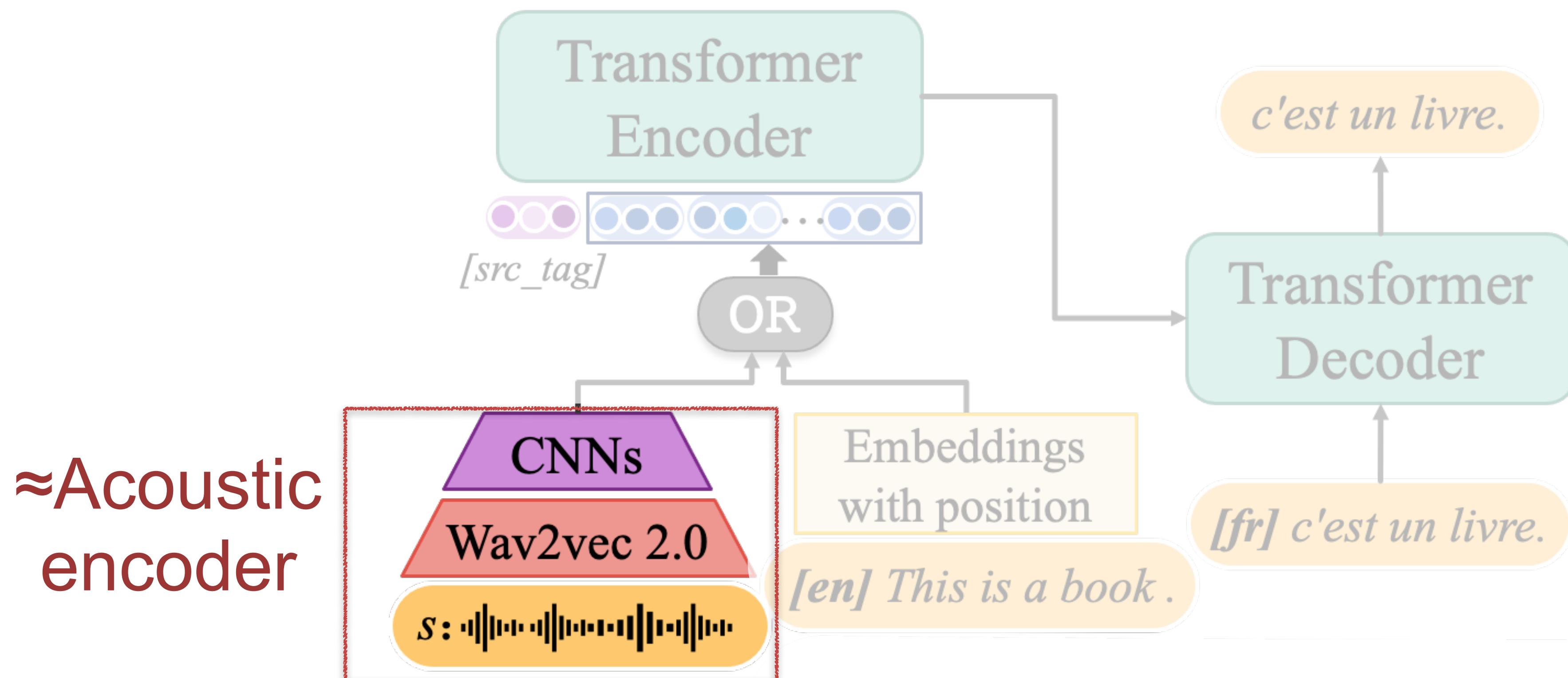
# Supports to train MT data

- Transformer MT model
- We can add more external MT data to train Transformer encoder & decoder



# Supports inputs of two modalities

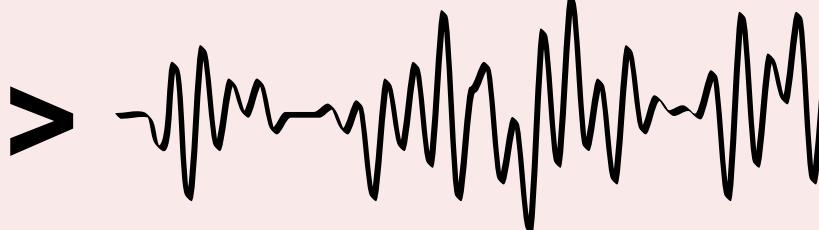
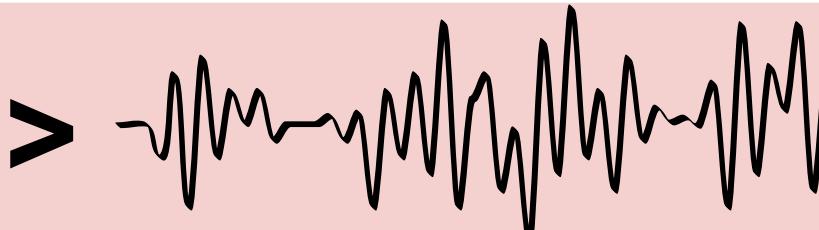
- Wav2vec2.0<sup>[1]</sup> as the acoustic encoder
- We add two convolution layers with 2-stride to shrink the length.



[1] wav2vec 2.0: A framework for self-supervised learning of speech representations, 2020

# Language indicator strategy

- We use language indicators to distinguish different tasks.

Tasks	Source input	Target output
MT	<en> This is a book.	<fr> c'est un livre.
ASR	<audio> 	<en> This is a book.
ST	<audio> 	<fr> c'est un livre.

# Progressive Multi-task Training

# Large-scale MT pre-training

Using **external MT**  $D_{MT-ext}$



# Multi-task Finetune

Using **(1) external MT**  $D_{MT-ext}$

- (2)  $D_{ST}$  with  $\langle speech, translation \rangle$
- (3)  $D_{ASR}$  with  $\langle speech, transcript \rangle$

**Progressive:**  
*Don't stop  
training  $D_{MT-ext}$*

# XSTNet achieves State-of-the-art Performance

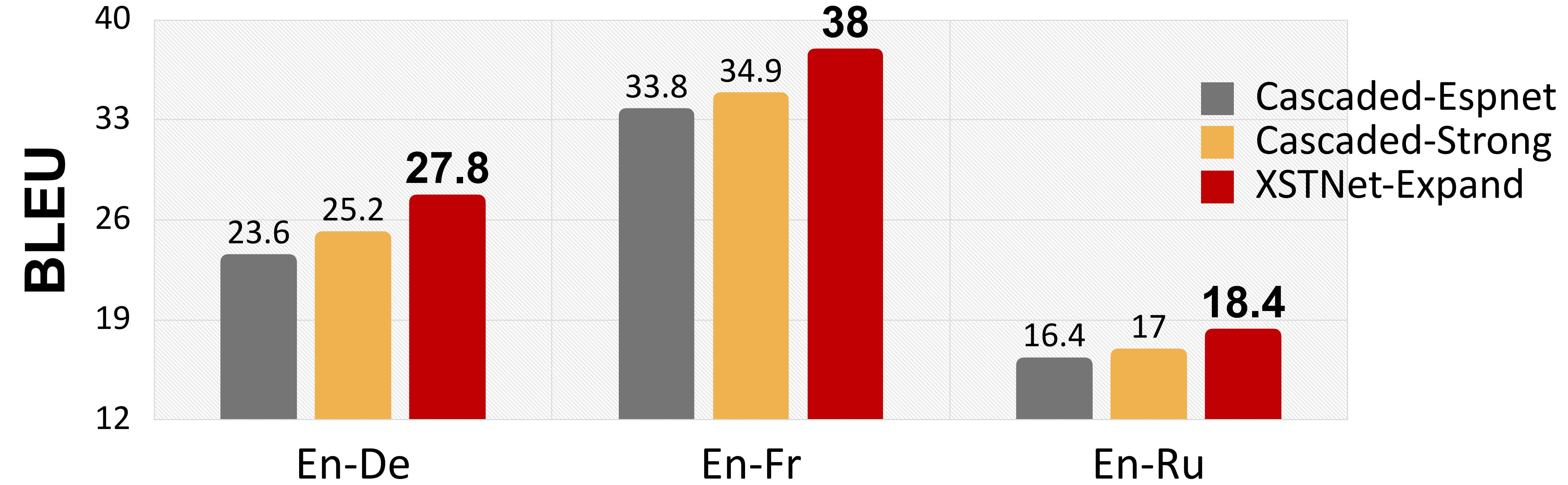
Models	External data	Pre-train tasks	En-De	En-Fr	En-Ru	Avg.
Transformer ST [13]	×	ASR	22.8	33.3	15.1	23.7
AFS [28]	×	×	22.4	31.6	14.7	22.9 (-0.8)
Dual-Decoder Transf. [15]	×	×	23.6	33.5	15.2	24.1 (+0.4)
STAST [29]	×	×	23.1	-	-	-
Tang et al. [2]	MT	ASR, MT	24.8	36.4	-	-
FAT-ST (Big) [6]	ASR, MT, mono-data <sup>†</sup>	FAT-MLM	25.5	-	-	-
W-Transf.	audio-only*	SSL*	23.6	34.6	14.4	24.2 (+0.5)
<b>XSTNet-Base</b>	audio-only*	SSL*	25.5	36.0	16.9	26.1 (+2.4)
<b>XSTNet-Expand</b>	MT, audio-only*	SSL*, MT	<b>27.8</b>	<b>38.0</b>	<b>18.4</b>	<b>27.8 (+4.1)</b>

Table 2: Performance (BLEU) on MuST-C En-De, En-Fr and En-Ru test sets. <sup>†</sup>: “Mono-data” means audio-only data from Librispeech, Libri-Light, as well as text-only data from Europarl/Wiki Text; \*: “Audio-only” data from Librispeech audio data is used in the pre-training of wav2vec2.0-base module, and “SSL” means the self-supervised learning from unlabeled audio data.

**XSTNet-Base:** Achieves the SOTA in the restricted setup

**XSTNet-Expand:** Goes better by using extra MT data

# XSTNet better than cascaded ST! a gain of 2.6 BLEU

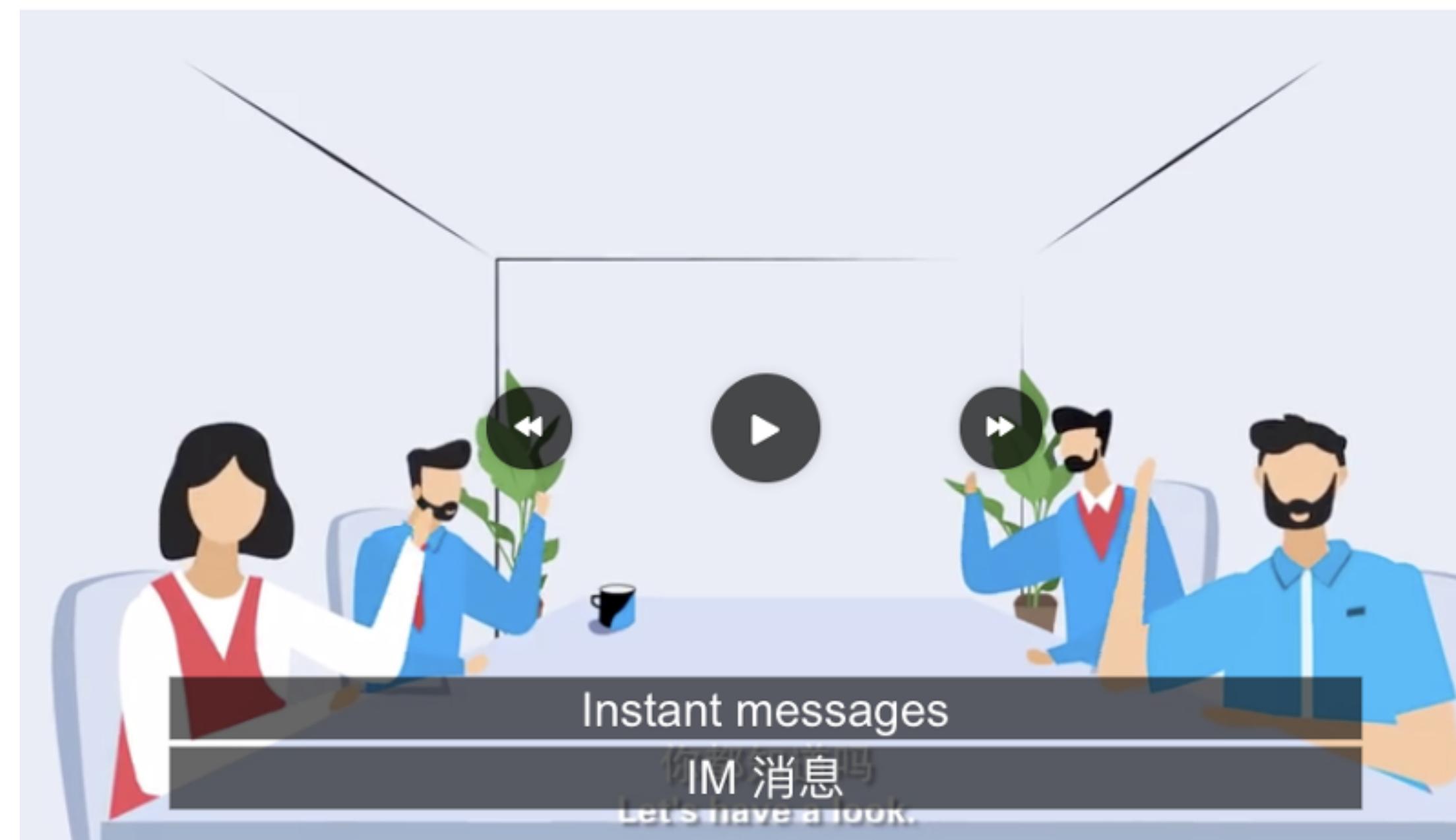


What is “Cascaded-Strong” system?

Strong ASR model		+ Large-scale MT data	Performance (En-De)
Cascaded - Strong	Model		WER=13.0 BLEU=31.7
ASR	W2V2+ Transformer	MuST-C $D_{ASR}$	
MT	Transformer-base	WMT + MuST-C $D_{MT}$	

# VolcTransStudio: Video Translation Platform

火山翻译  
VolcTransStudio



实时翻译, 自动提示 & 交互式修改



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# Summary

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- End-to-end Speech-to-Text works!
- Use external ASR, MT data, and audio/text for auxiliary signals
- Model
  - **LUT**: two-stage encoder, additional BERT KD [Dong et al AAAI 2021a]
  - **Chimera**: Shared semantic space encoder with fixed-size memory [Han et al ACL 2021]
  - **COSTT**: consecutive transcription-translation decoder [Dong et al AAAI 2021b]
- Training technique
  - Audio pre-training: Wave2Vec2.0[Baevski et al 2021]
  - External MT Pre-training
  - **XSTNet**: Progressive multi-task training [Ye et al Interspeech 2021]

# Thanks

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**WeurST** neural speech translation toolkit

<https://github.com/bytedance/neurst>



High performance sequence inference

<https://github.com/bytedance/lightseq>

