

CS11-737 Multilingual NLP

Speech Translation

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<https://lileicc.github.io/course/11737mnlp23fa/>



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Language Technologies Institute

HW2 Update

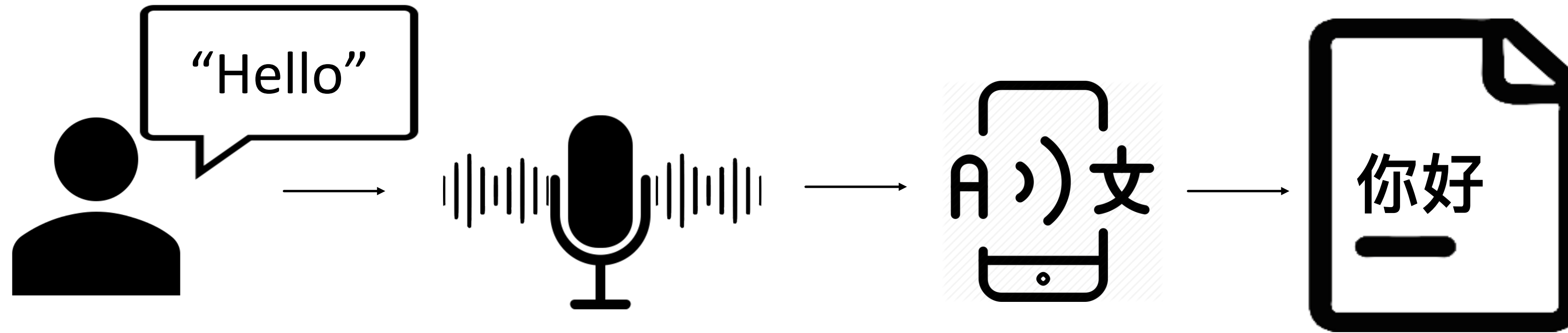
- ASR for Quechua will be optional
 - bonus 20pts if completed
 - additional data: http://festvox.org/cmu_wilderness/
 - http://lrec-conf.org/workshops/lrec2018/W14/pdf/4_W14.pdf

Mid-term Report

- Everything in proposal with adjustment,
 - project description
 - data
 - evaluation procedure/metric
 - a baseline model and baseline results
 - Clearly state individual team member contribution
 - Use ACL paper template in latex

Speech-to-Text Translation(ST)

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



Application Type

- (Non-streaming) ST e.g. video translation
- Streaming ST e.g. realtime conference translation

System

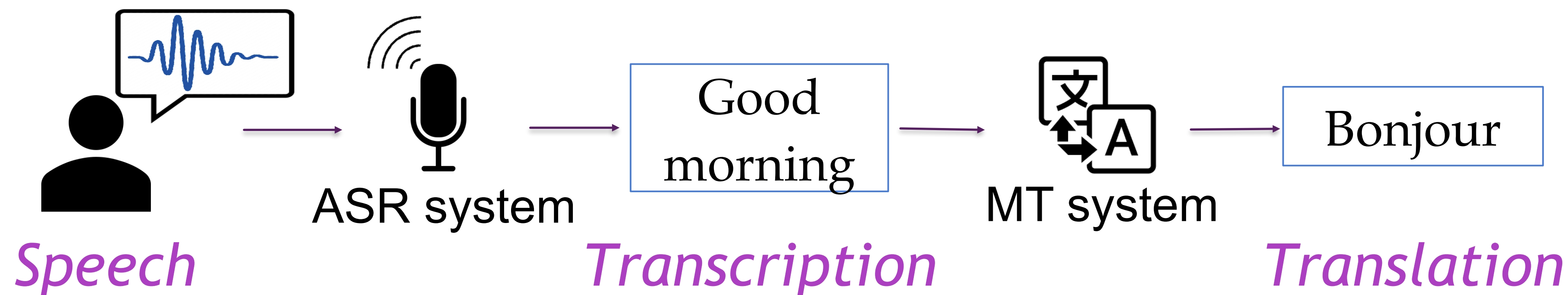
- Cascaded ST
- End-to-end ST

Cascaded ST System

- Challenges:

1. Computationally inefficient

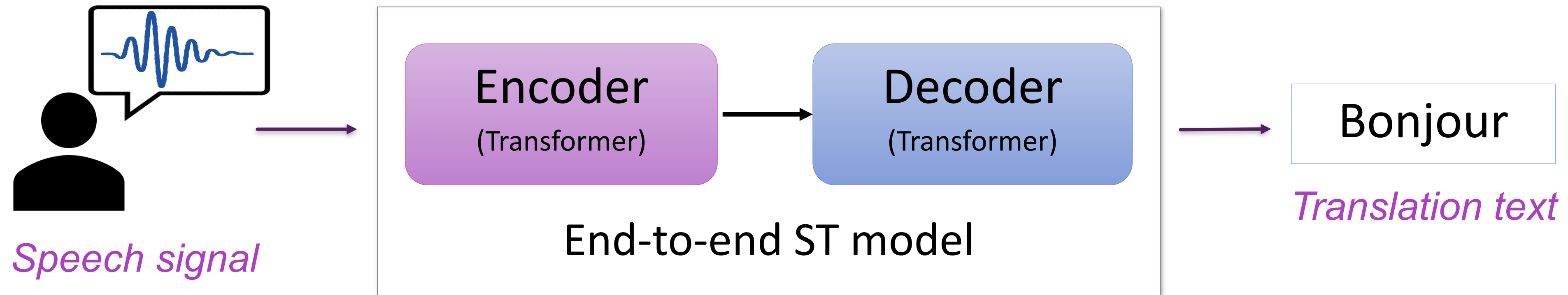
2. Error propagation: Wrong transcription → Wrong translation



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

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

End-to-end ST Model

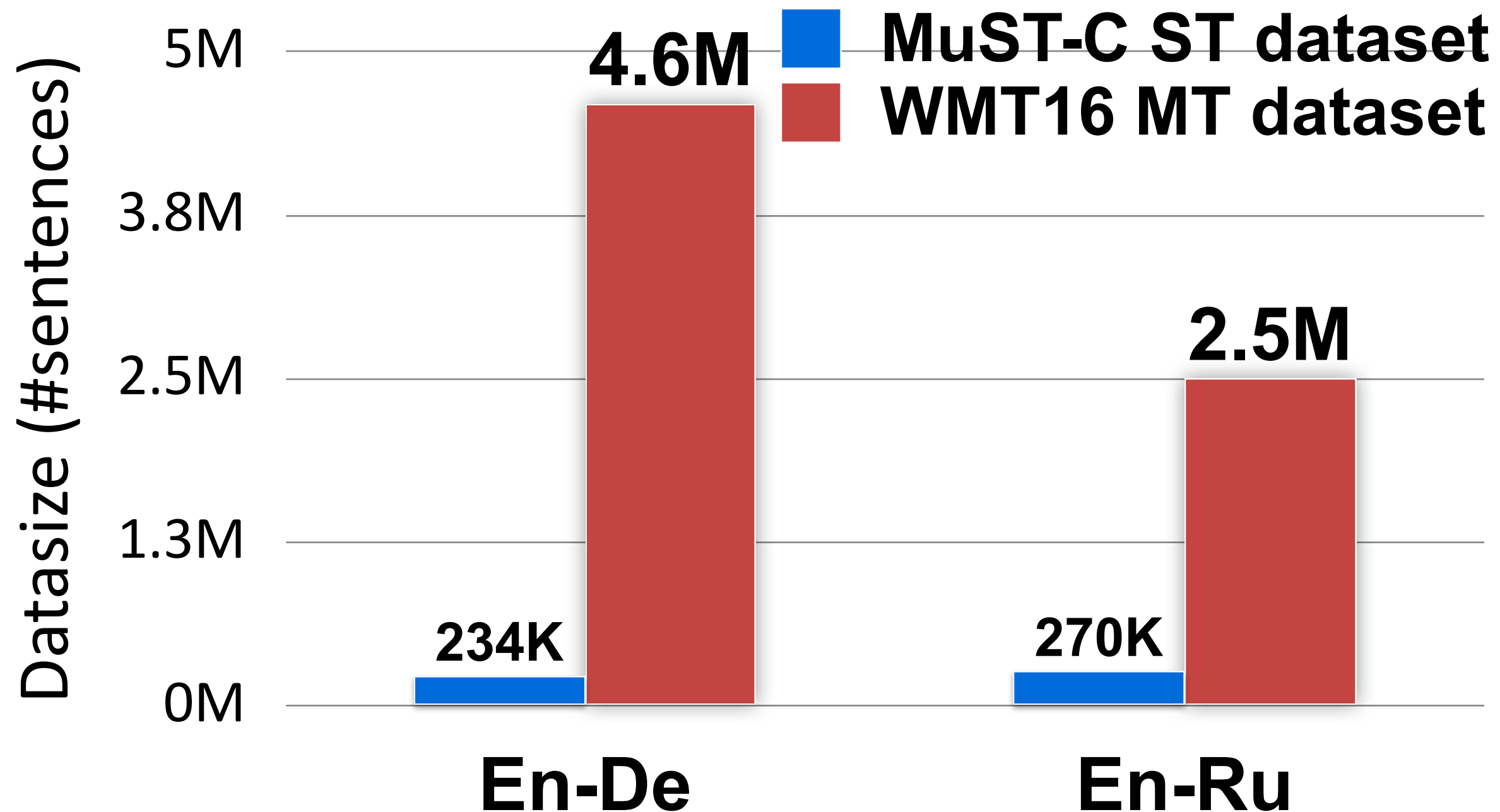


- Single model to produce text translation from speech
- Basic model: Encoder-Decoder architecture (e.g. Transformer)
- Advantage:
 - Reduced latency, simpler deployment
 - Avoid error propagation

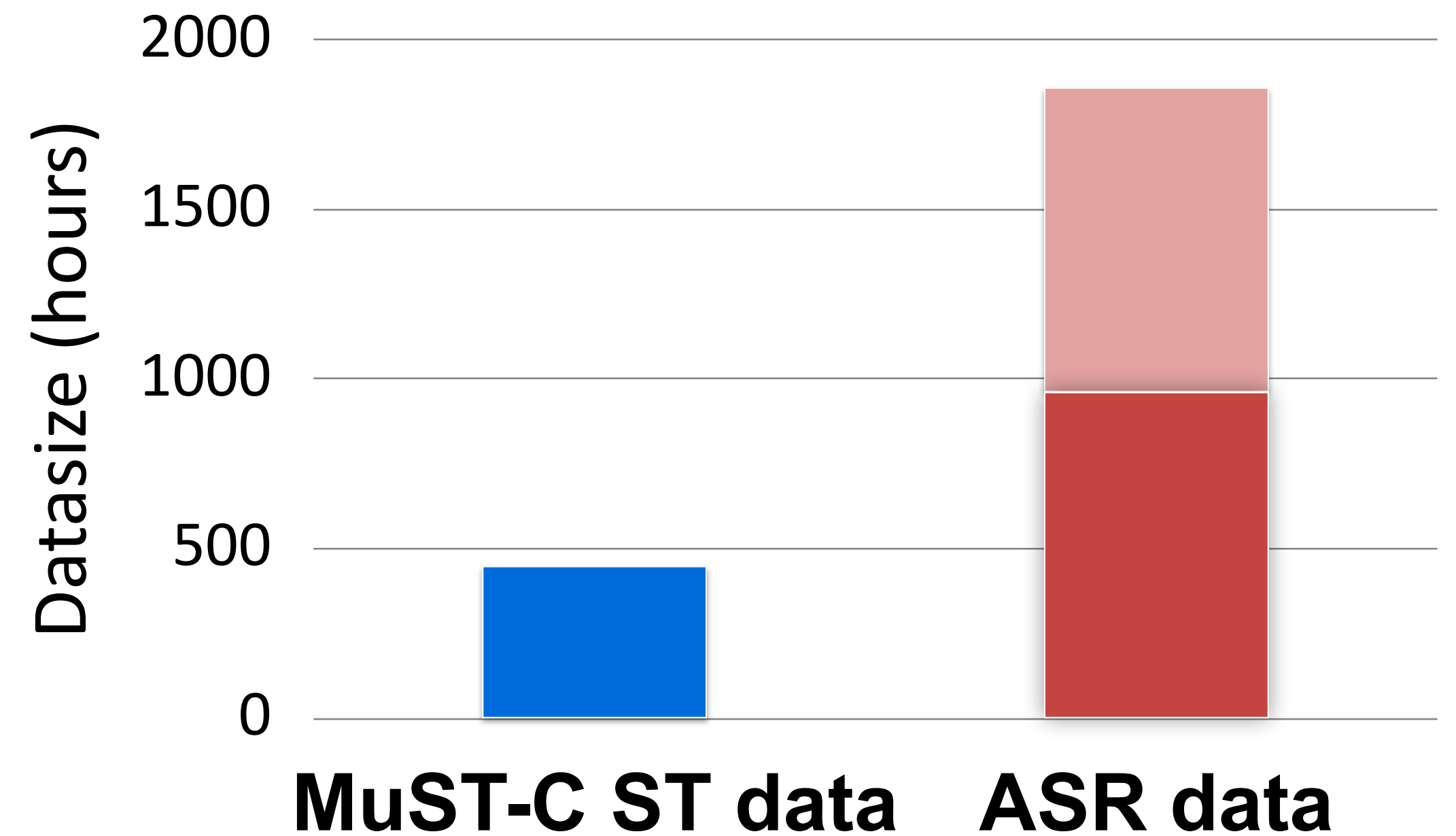
Challenge

- Data scarcity - lack of large parallel audio-translation corpus
- Modality Disparity between speech and text

Dataset size (Text)
ST vs MT



Dataset size
ST vs ASR

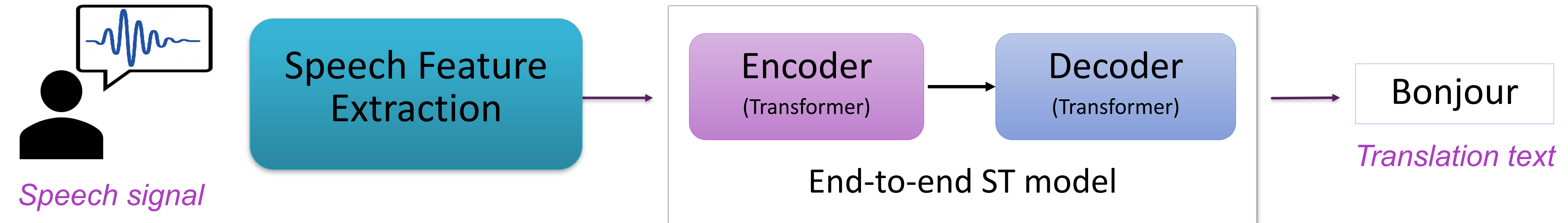


Challenge

- Modality Disparity between speech and text
 - Disfluencies
 - ▶ Hesitations: “uh”, “uhm”, “hmm”,
 - ▶ Discourse markers: “you know”, “I mean”,...
 - ▶ Repetitions: “It had, it had been a good day”
 - ▶ Corrections: “no, it cannot, I cannot go there”
 - Unlike (Text) MT, No punctuation
 - ▶ let s eat grandpa
 - ▶ Let’s eat, Grandpa !

Basic End-to-end ST Model

Basic ST model

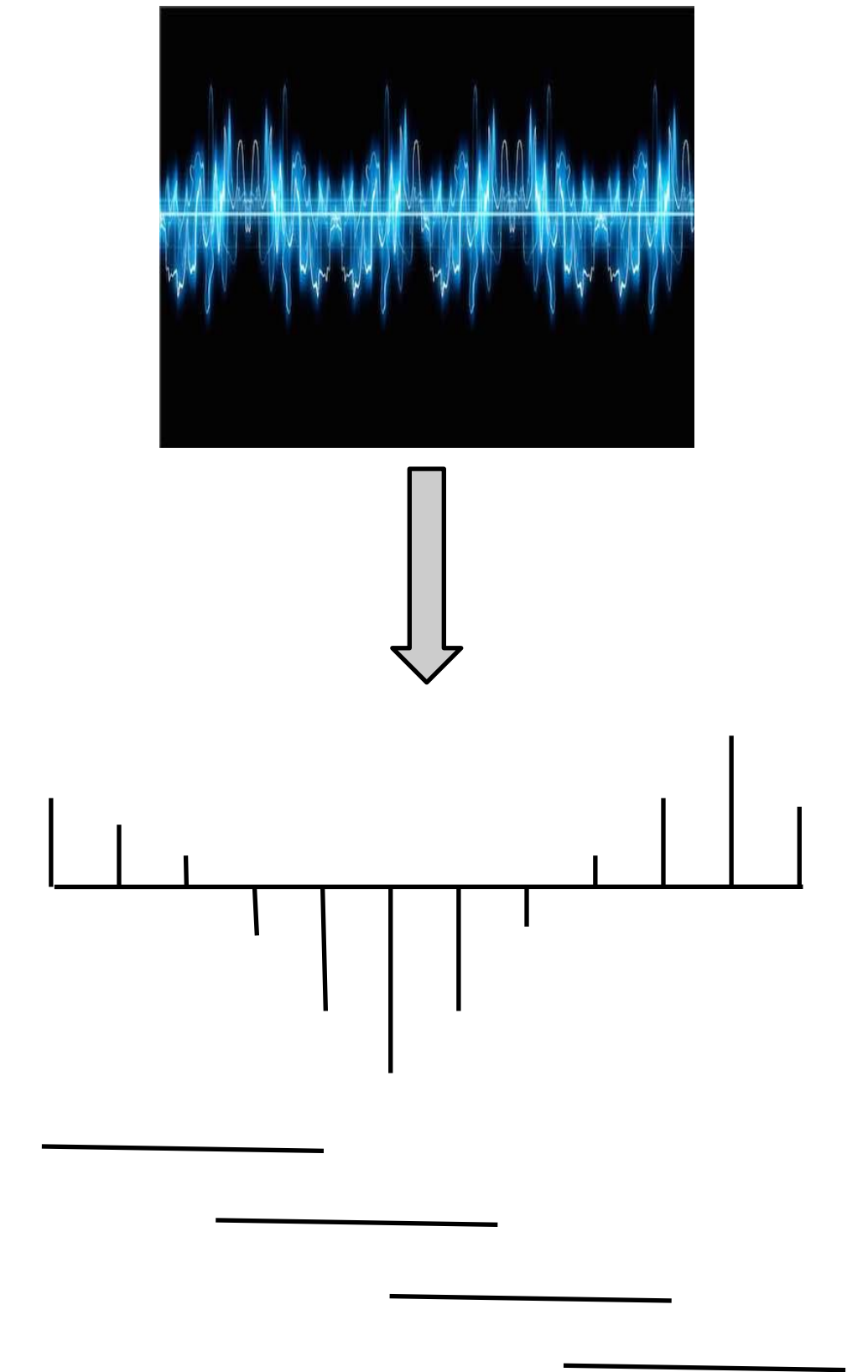


Main differences to text machine translation

Input: Audio signals are continuous and much longer!

Audio Signal - Same as ASR

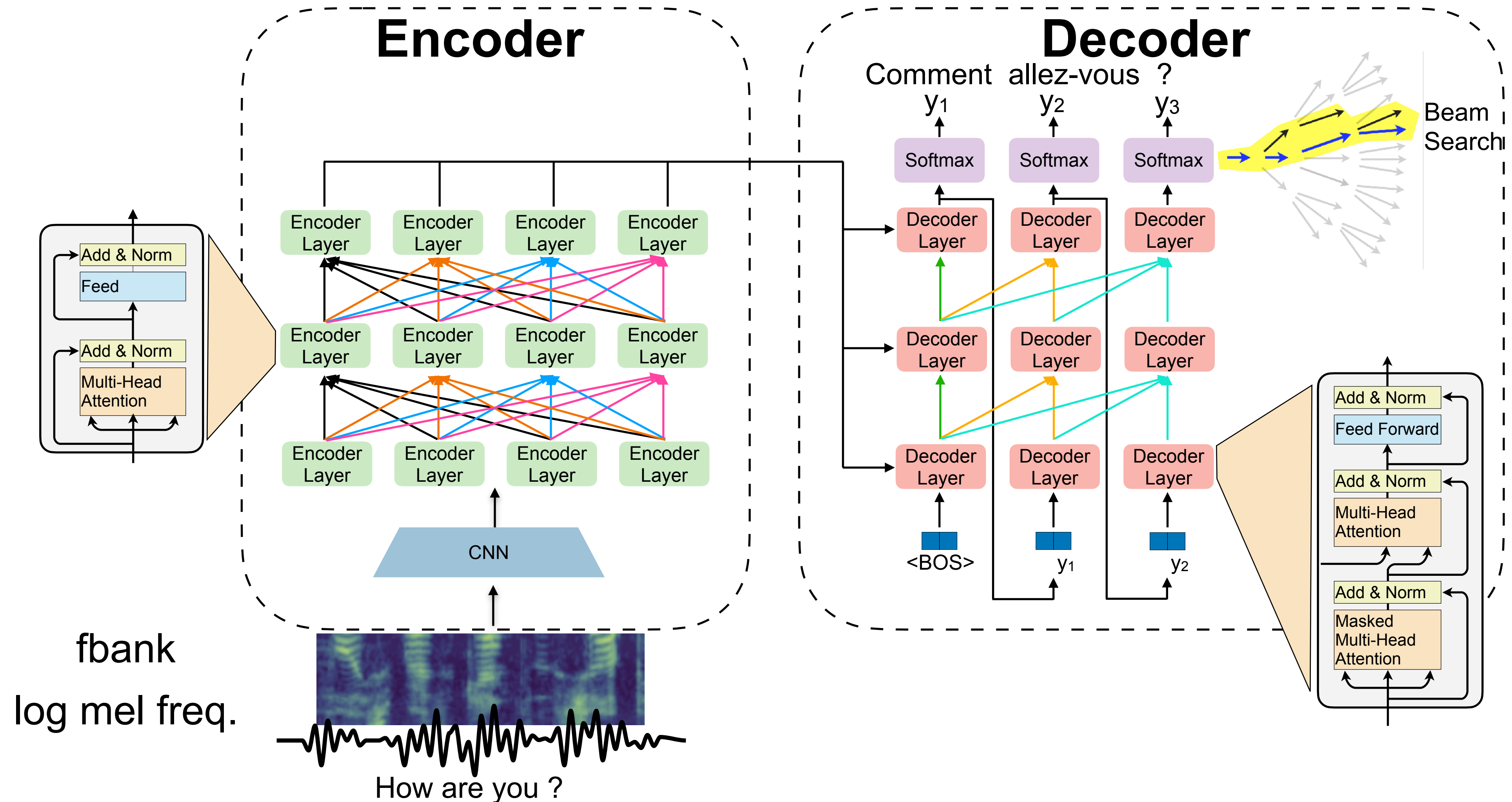
- Following best-practice from ASR
- Signal Sampling
 - Measure Amplitude of signal at time t
 - Typically 8kHz or 16 kHz
- Windowing — Frame
 - Split signal in different windows, called Frame
 - ▶ Length: ~ 20-30 ms (typically 25ms)
 - ▶ Stride: ~ 10 ms



Basic Speech Translation Model (Similar to MT)

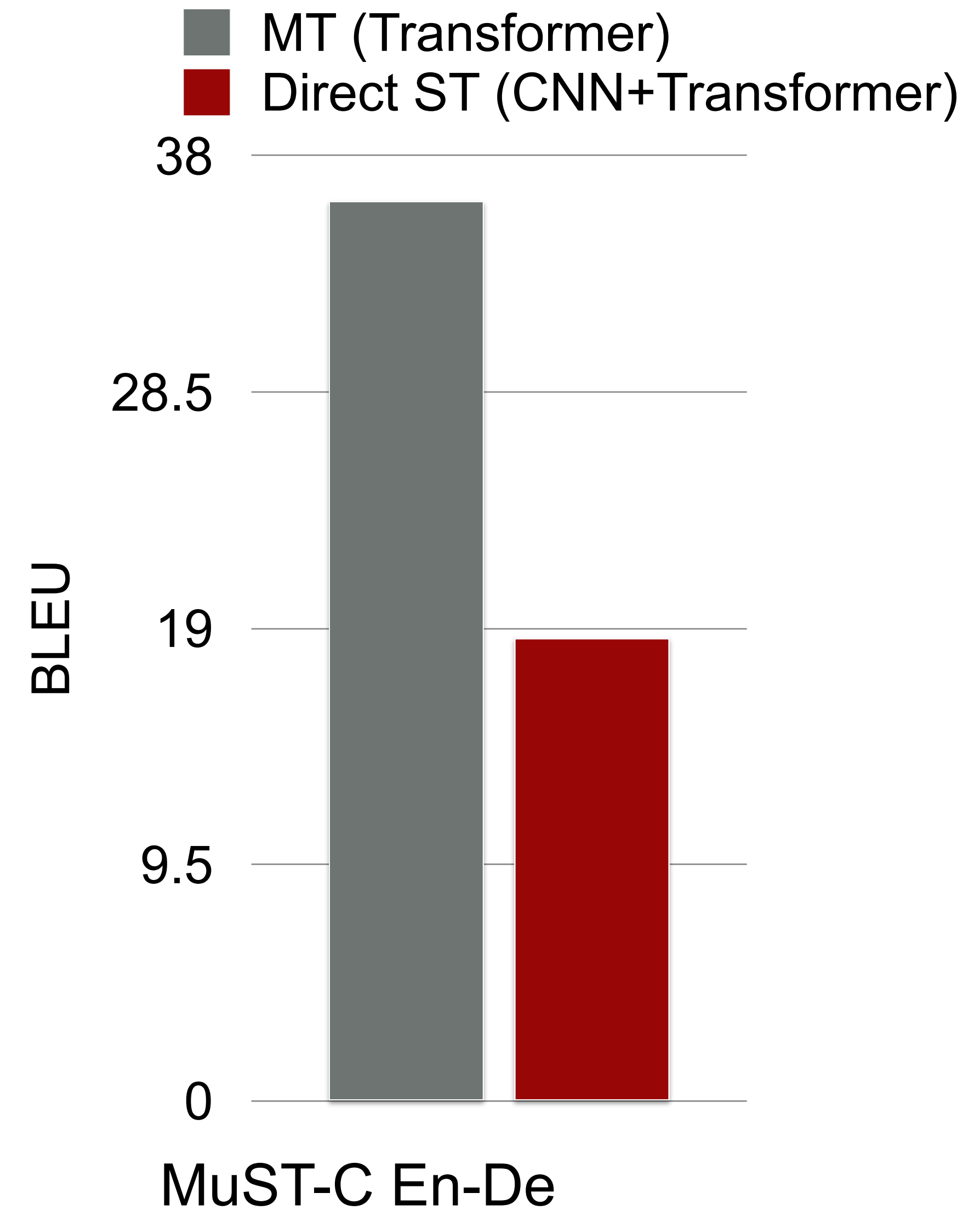
Transformer-based: N-layer convolution + attention encoder, M-layer decoder

Training data: <audio seq., translation text>



Speech Translation model lags behind MT

- Performance on MuST-C En-De:
 - ST 18.6
 - MT 36.2 (taking correct transcript as input)

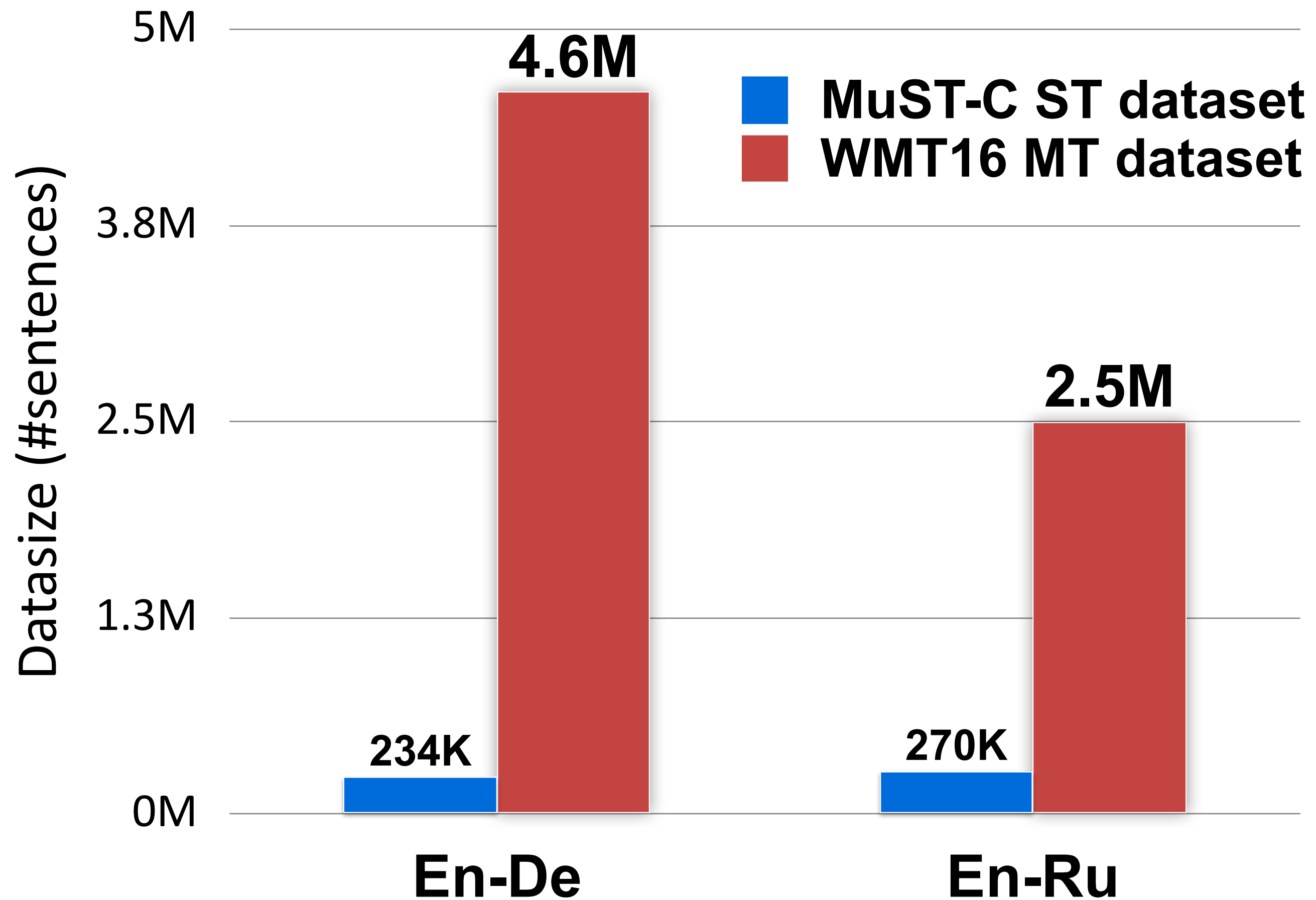


Approaches for Speech Translation

- Utilizing additional parallel text from MT corpus - MT pretraining
 - Decoder initialization from separately trained MT model
 - Single-modal(audio) Encoder-Decoder: COSTT[Dong et al, AAI 2021b]
- Using Additional ASR data - ASR Pre-training
 - Curriculum Pre-training [Wang et al, ACL 2020]
 - LUT [Dong et al, AAI 2021a]
- Using additional raw audio data
 - Wav2vec & Wav2Vec2.0 [Schneider et al. Interspeech 2019, Baevski et al NeurIPS2020]
 - Apply to ST [Wang et al, 2021, Zhao et al, ACL 2021, Wang et al, Interspeech 2021]
- Distilling knowledge from Pre-trained Language Model (BERT)
 - LUT [Dong et al, AAI 2021a]
- Learning Better Speech-text cross-modal representation for ST
 - TCEN-LSTM [Wang et al, AAI 2020]
 - Chimera [Han et al, ACL 2021a]
 - Wav2vec2.0 + mBart + Self-training [Li et al, ACL 2021b]
 - FAT-ST [Zheng et al, ICML 2021]
 - ConST [Ye et al, 2022]
 - WACO [Ouyang et al, 2023]
- Better Fine-tuning Strategy
 - XSTNet [Ye et al, Interspeech 2021]

Using external Parallel Text

Dataset size ST vs MT



How to use MT

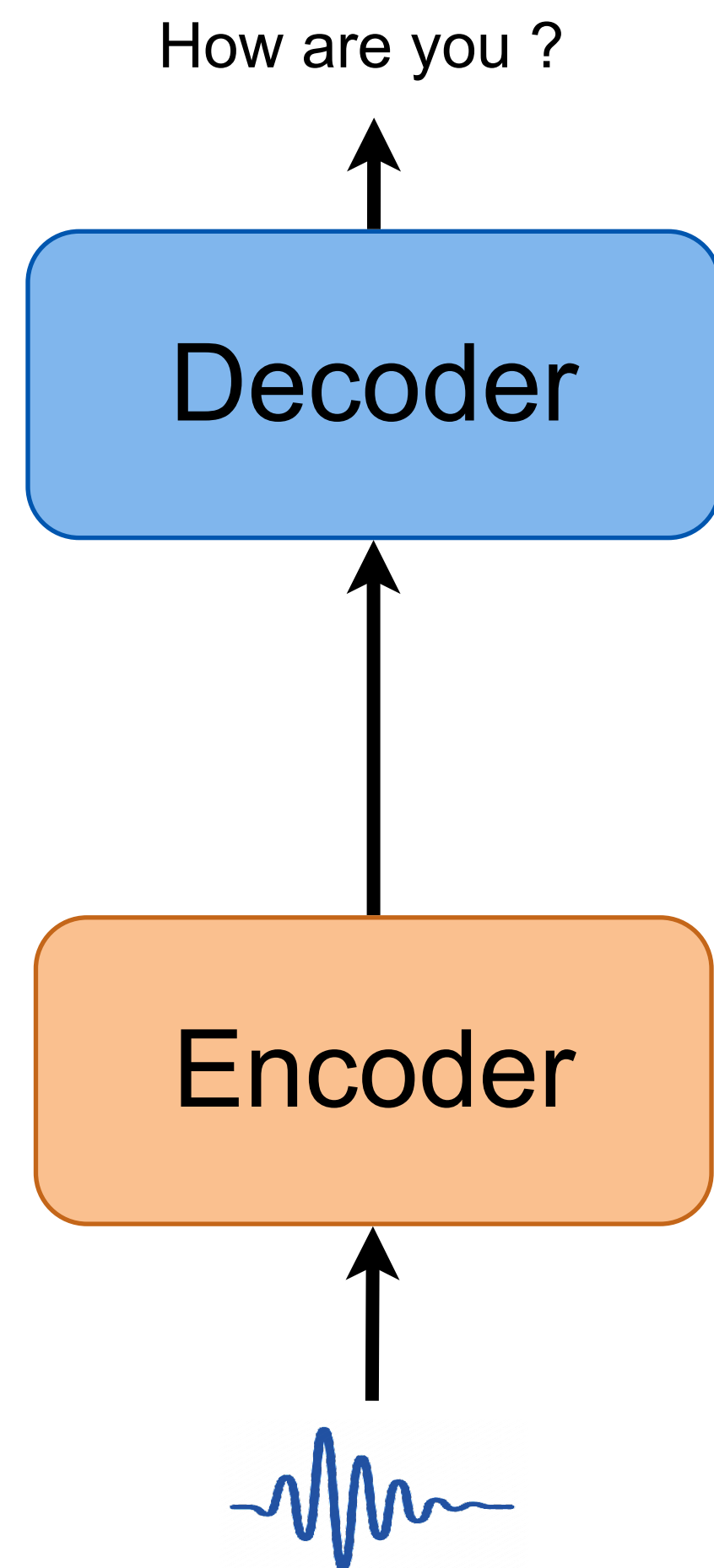
data *with much larger scale* to improve ST performance?

Separate Encoder-Decoder Pre-train

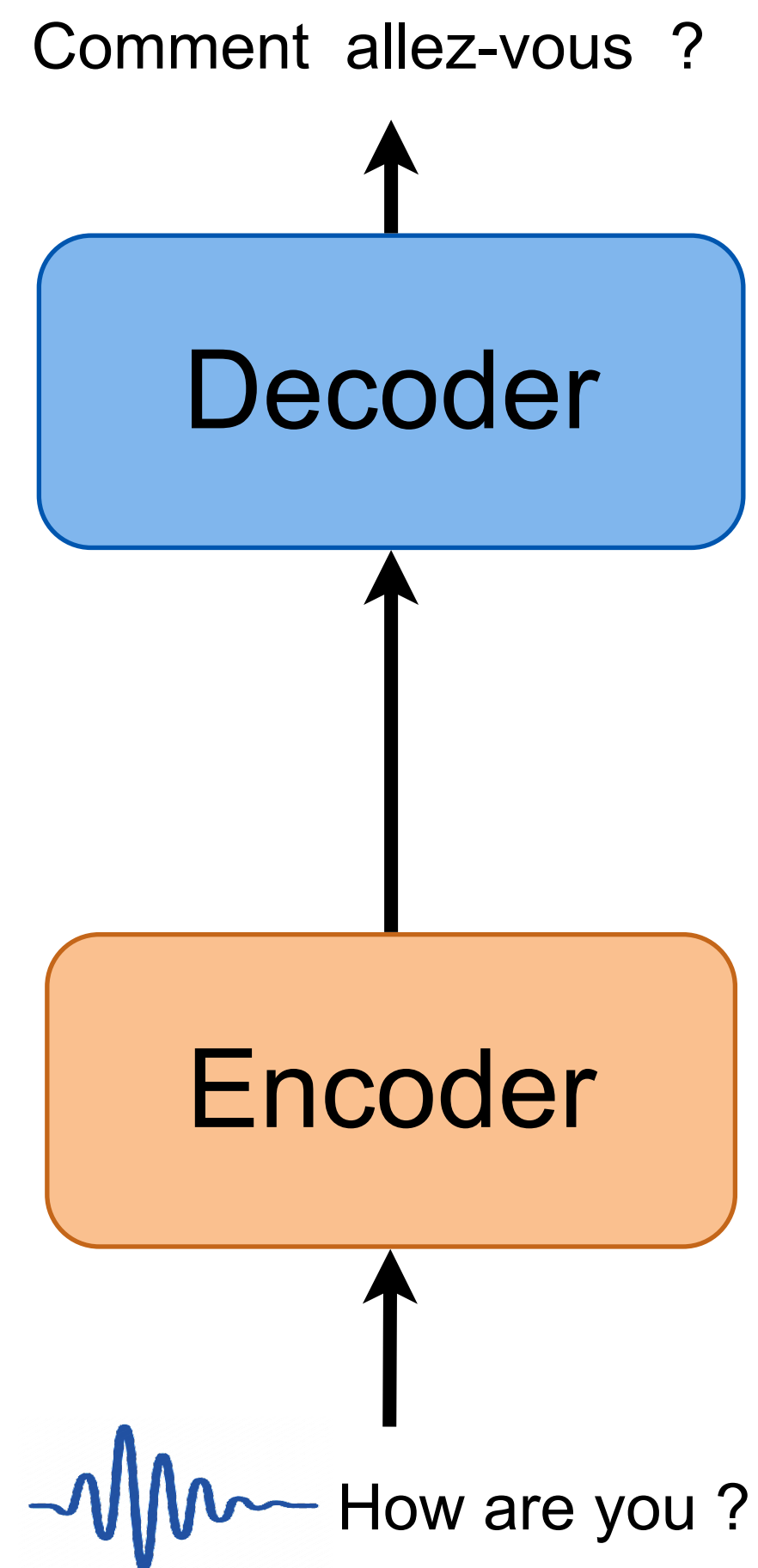
Speech Recognition
LibriSpeech corpus

Speech Translation
fine-tune on ST data

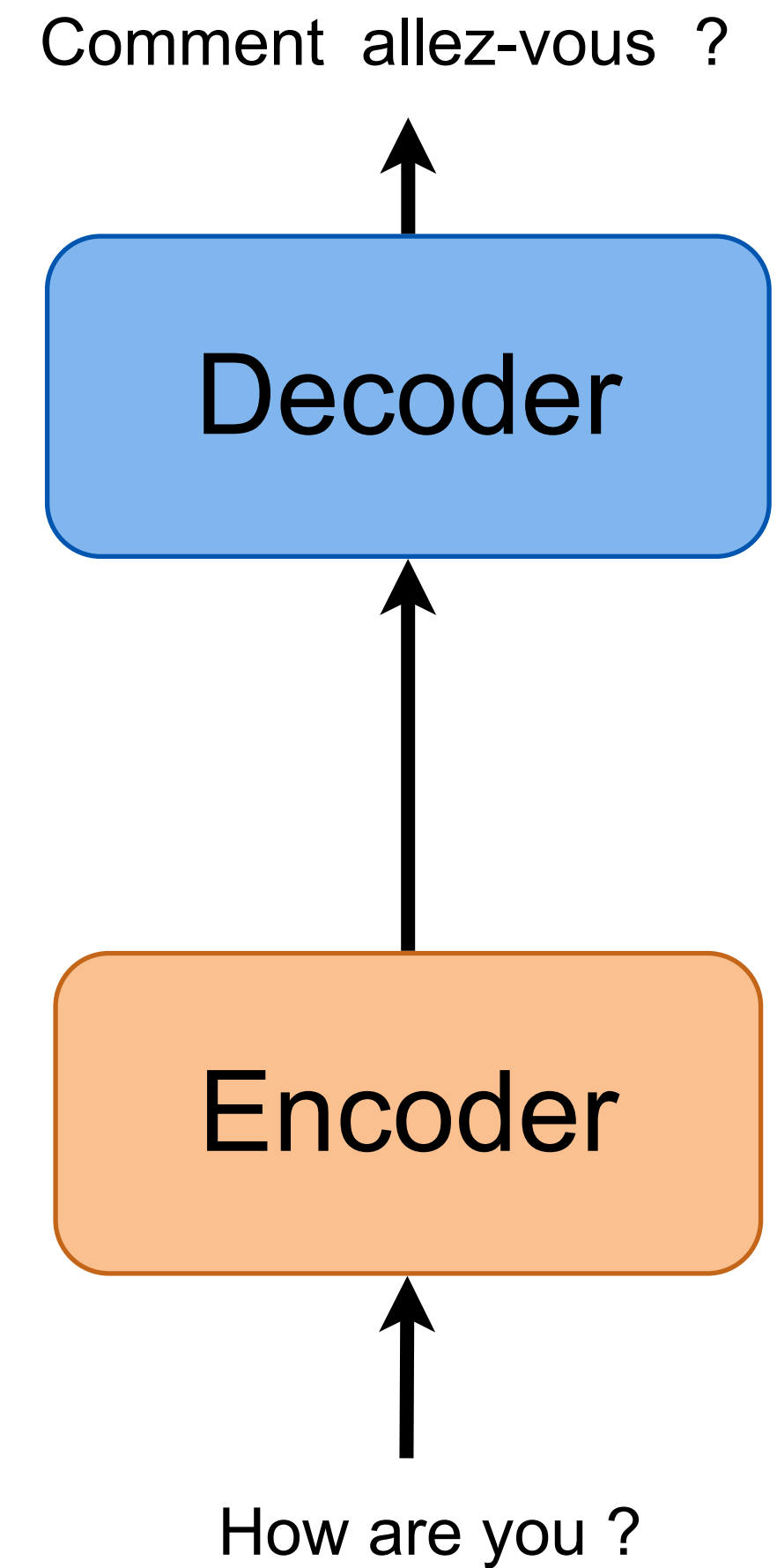
Machine Translation
WMT corpus



init

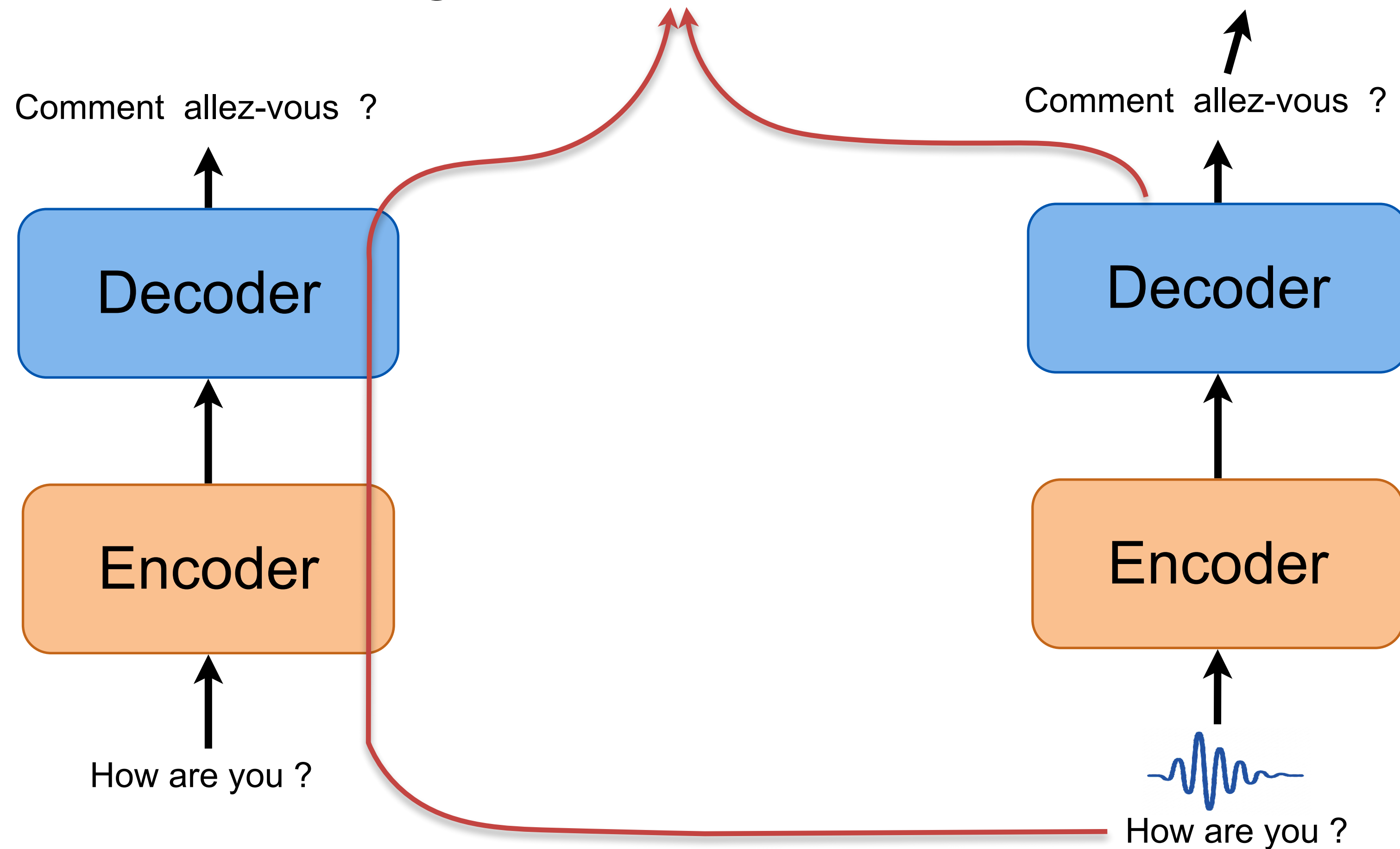


init



Knowledge Distillation from MT model

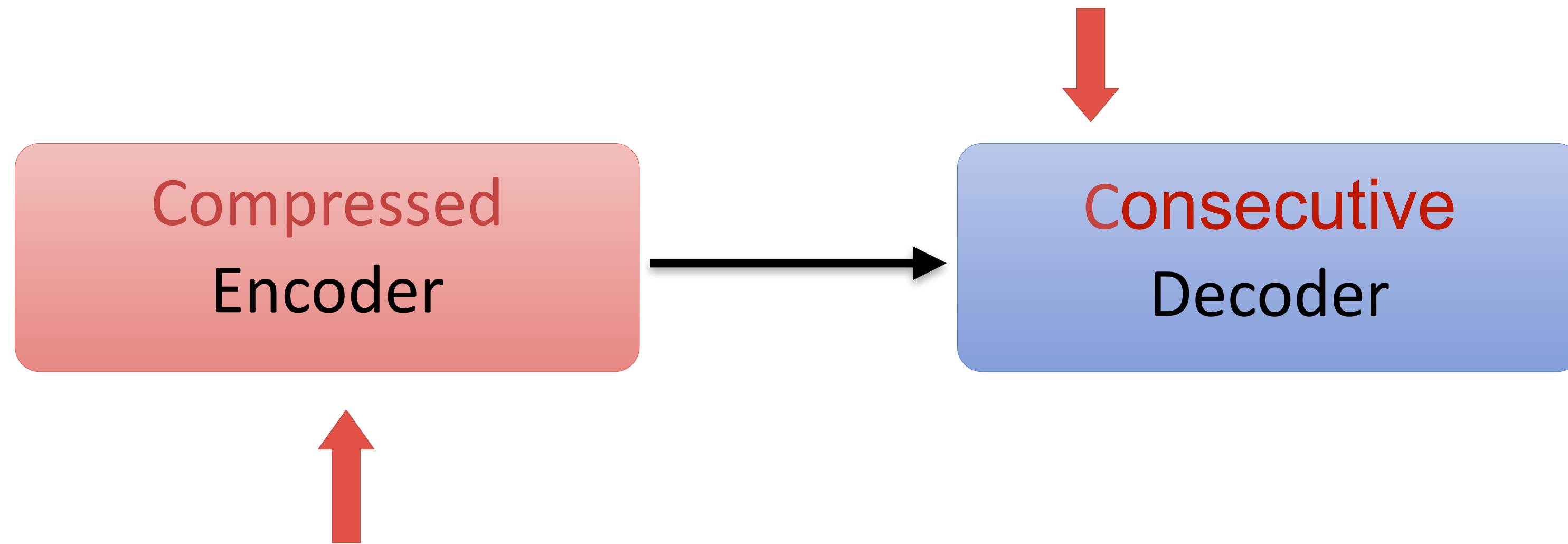
MT pre-training **KL loss + ST Cross-entropy loss**



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!

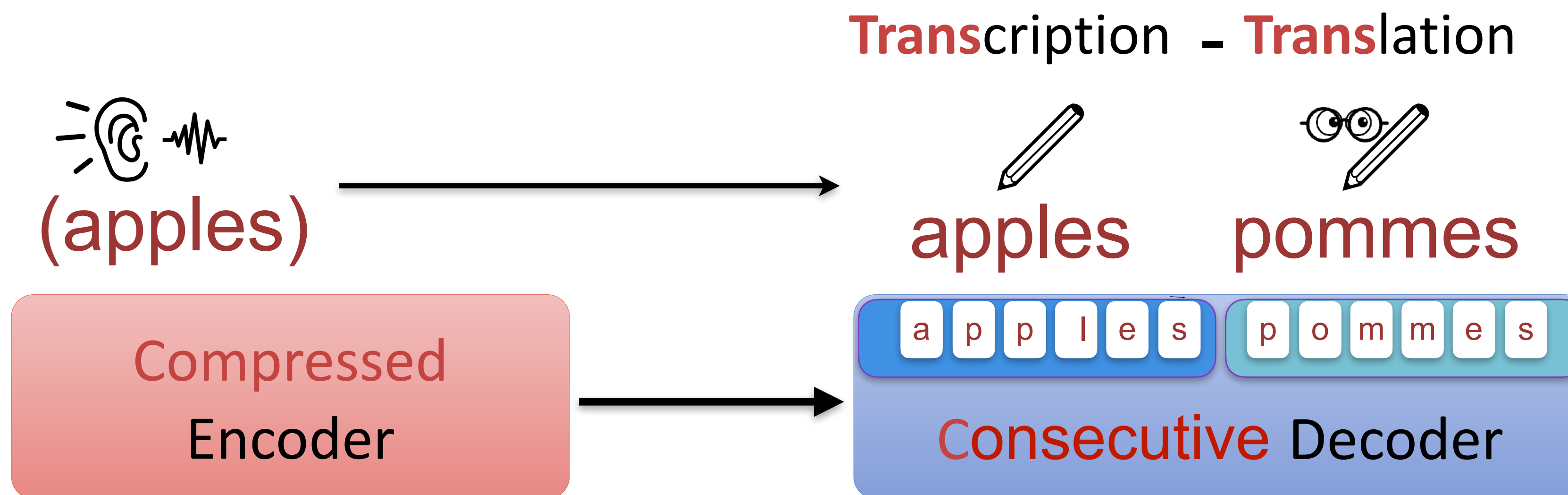
Idea 2: Introduce a **compressed encoder** to relief the model memory.

Pre-train ST's decoder with full MT

How to make a single model's decoder to perform text translation?

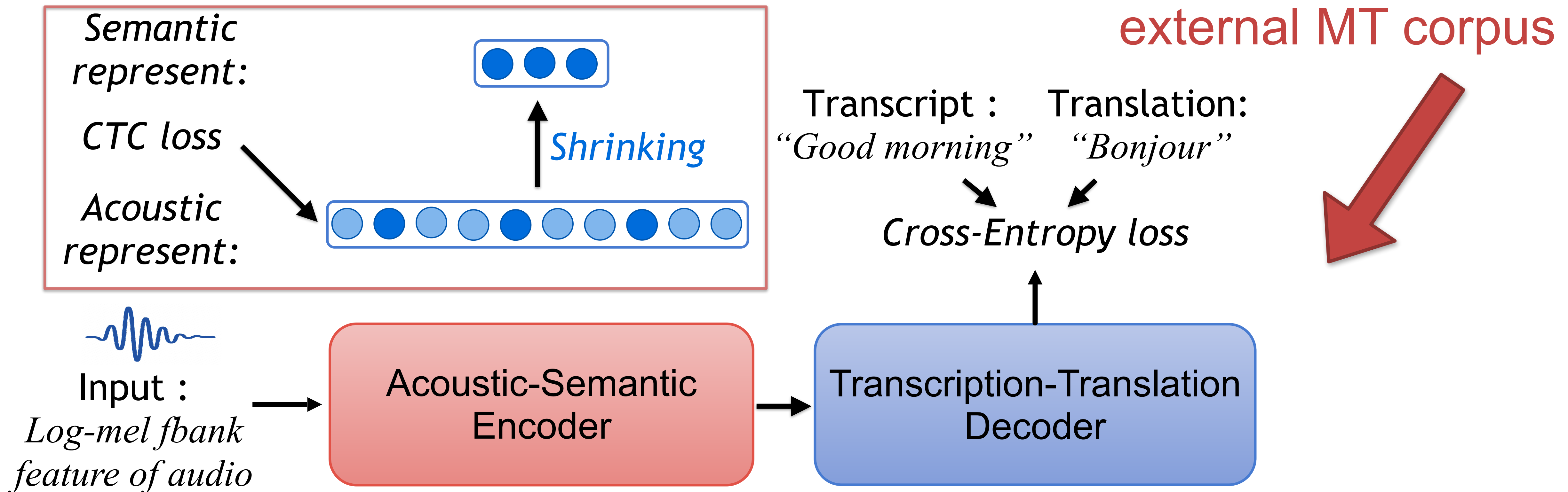
Decoder ==> translation

Encoder -> Decoder ==> transcribe and translation



COSTT for ST

Step 1: Pre-train using external MT corpus

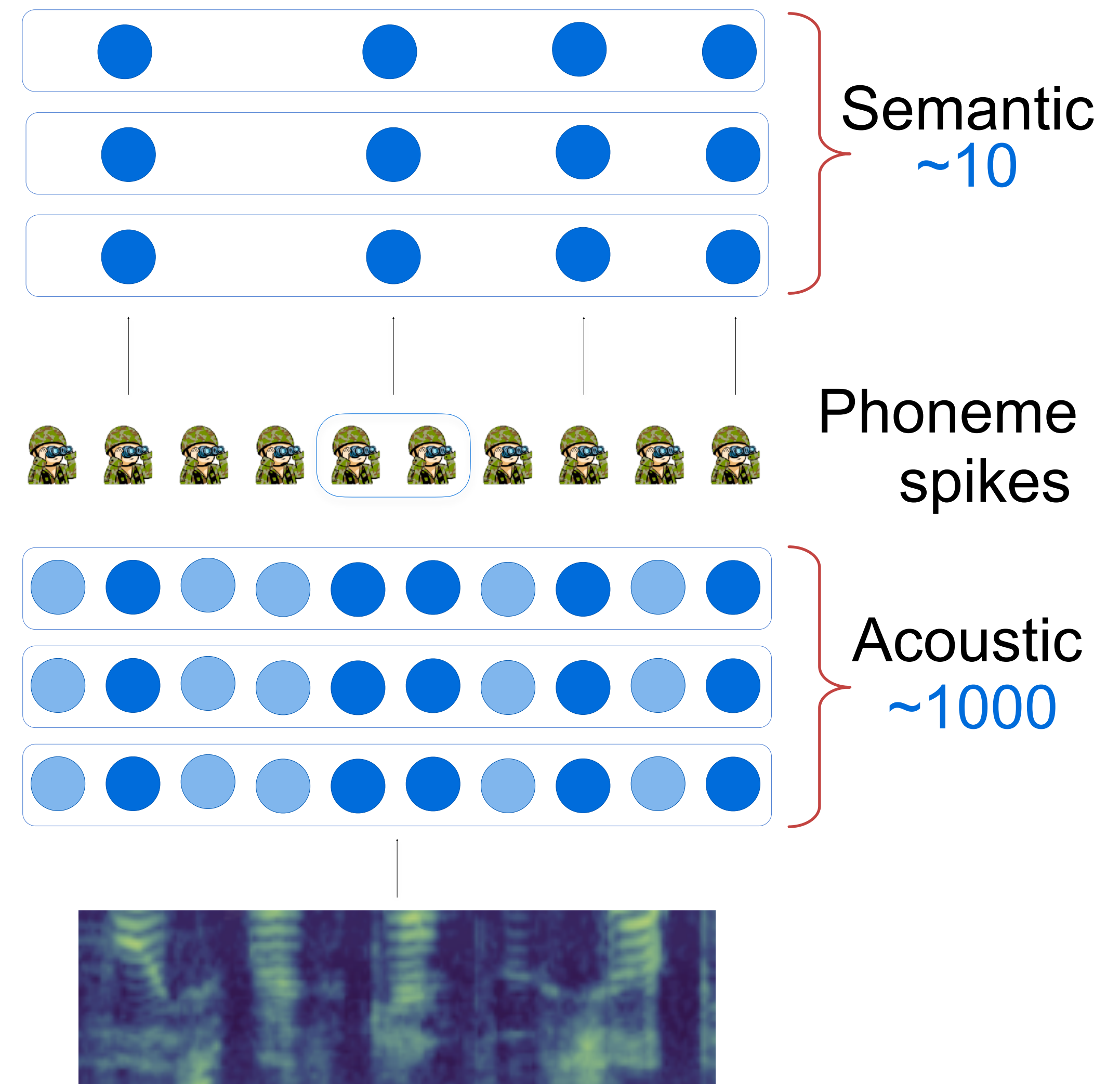


Step 2: Train encoder w/ shrinking module using CTC

Step 3: Train full model on ST data <audio, transcript, translation>

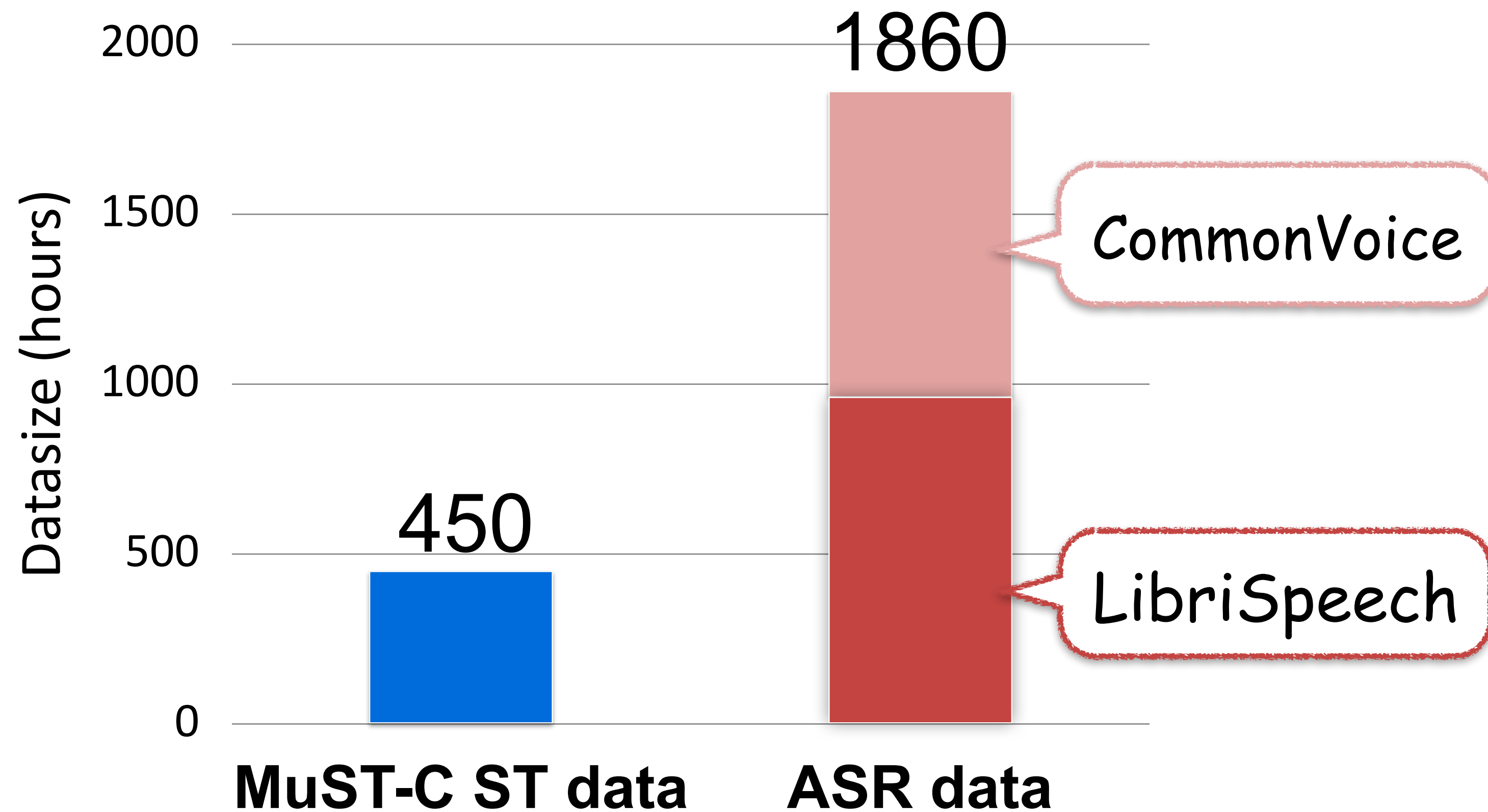
Advantages of COSTT

- 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**



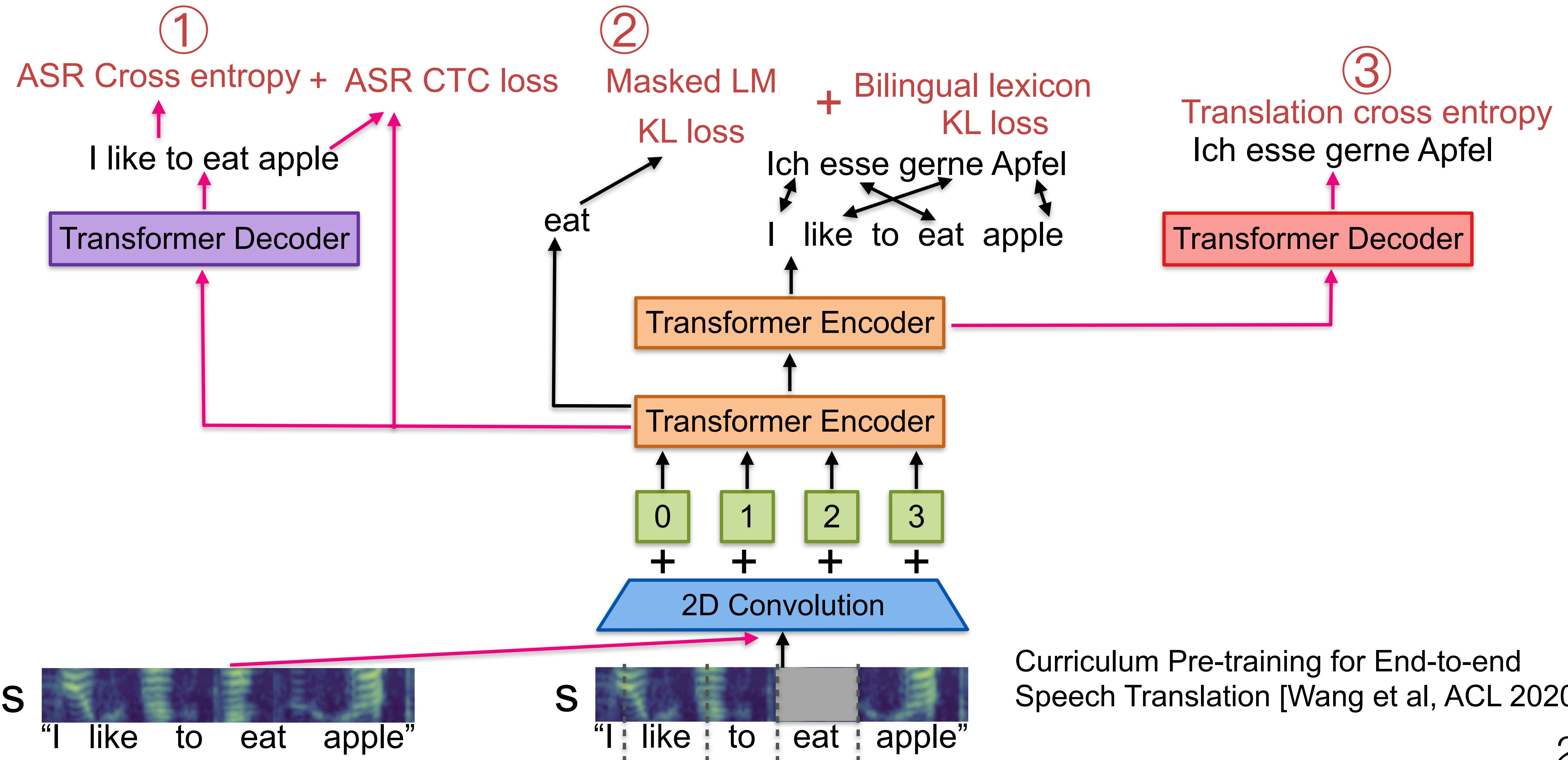
Using external ASR data

Dataset size ST vs ASR

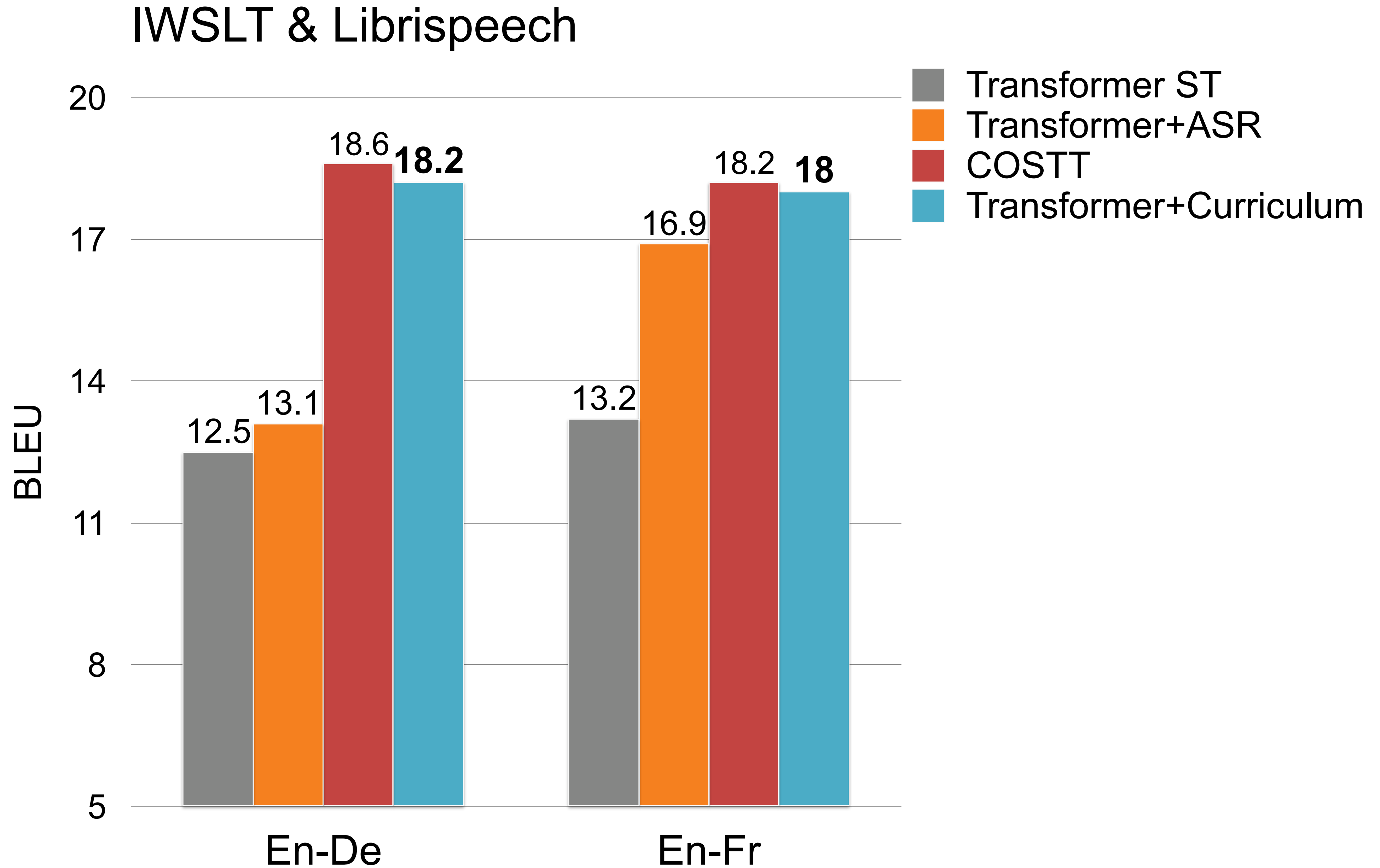


How to use larger external ASR data to improve ST performance?

Curriculum Pre-training with ASR data

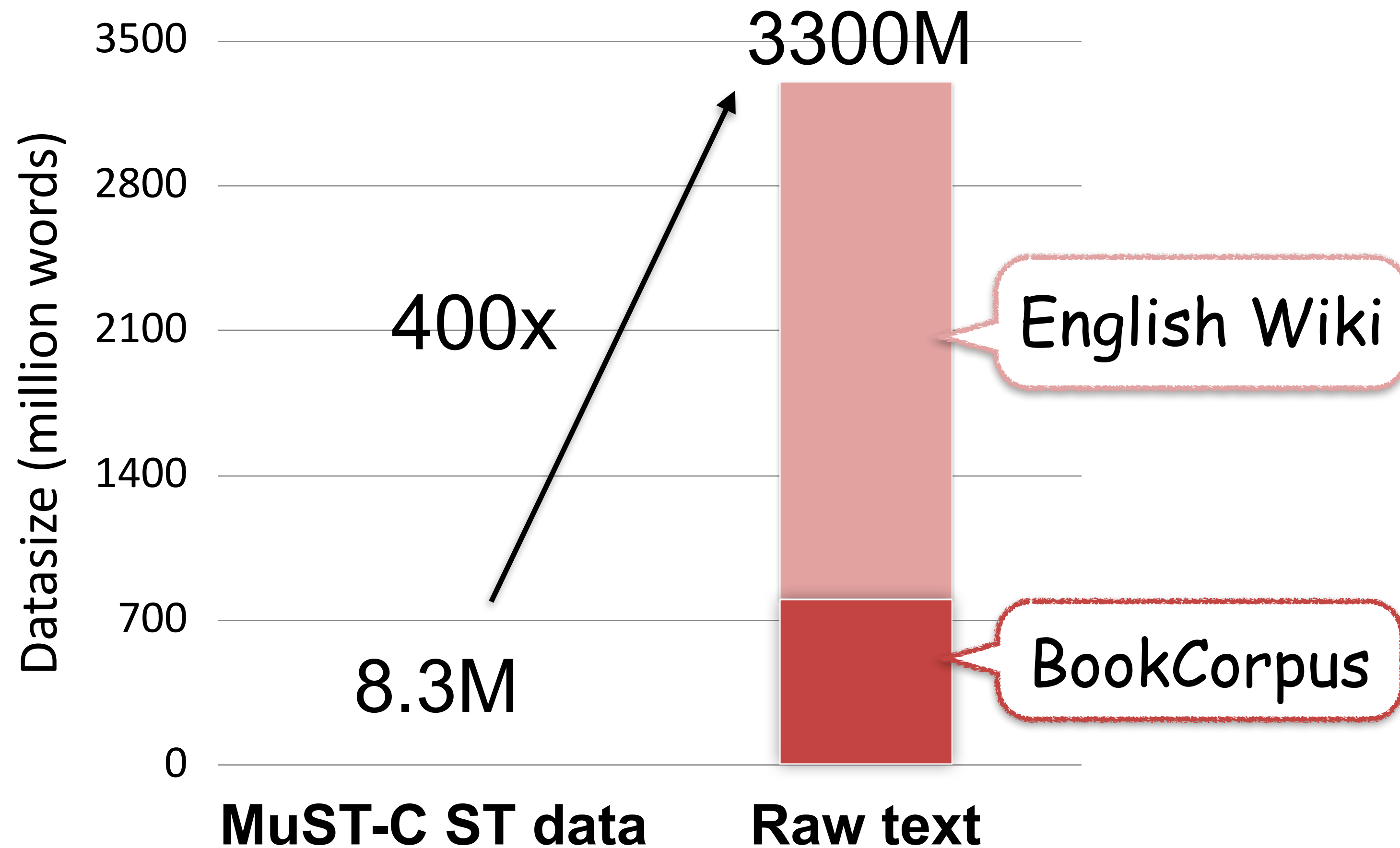


ASR Pre-training helps ST



Raw Text Pre-training

Dataset size ST vs Raw text

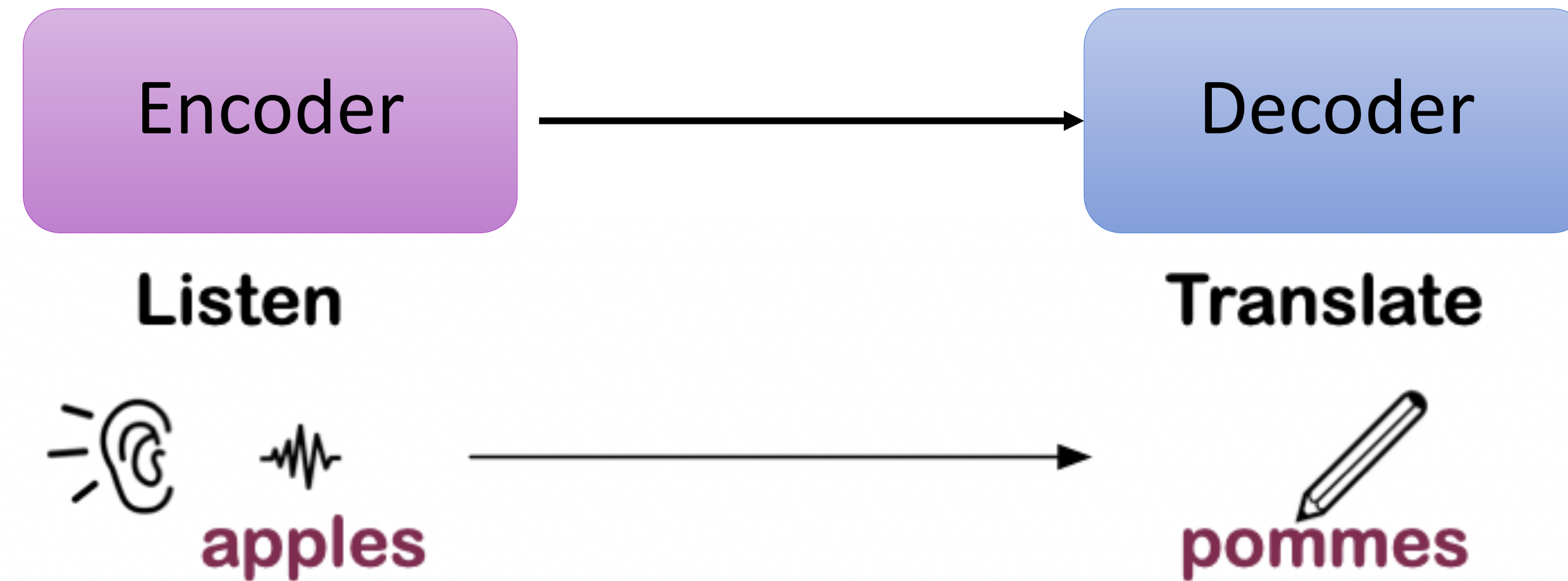


Using pre-trained LM in decoding weighting is easy!

But

🤔 How to use pre-trained **BERT** to improve ST performance?

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?

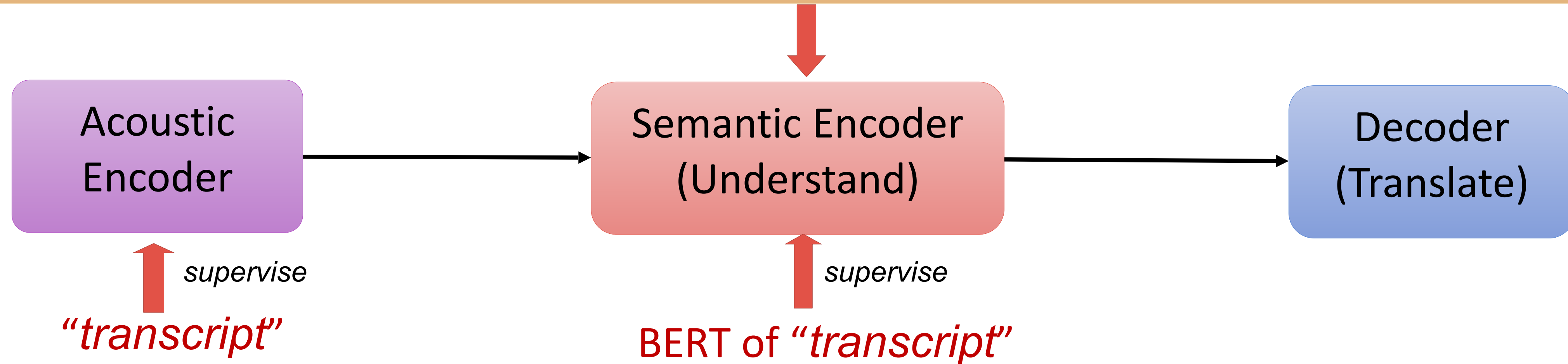


“Listen-Understand-Translate” (LUT) model based motivated by human's behavior

Motivation of Better Encoding

Drawback 1: A single encoder is not enough.

Idea 1: Introduce a **semantic encoder**



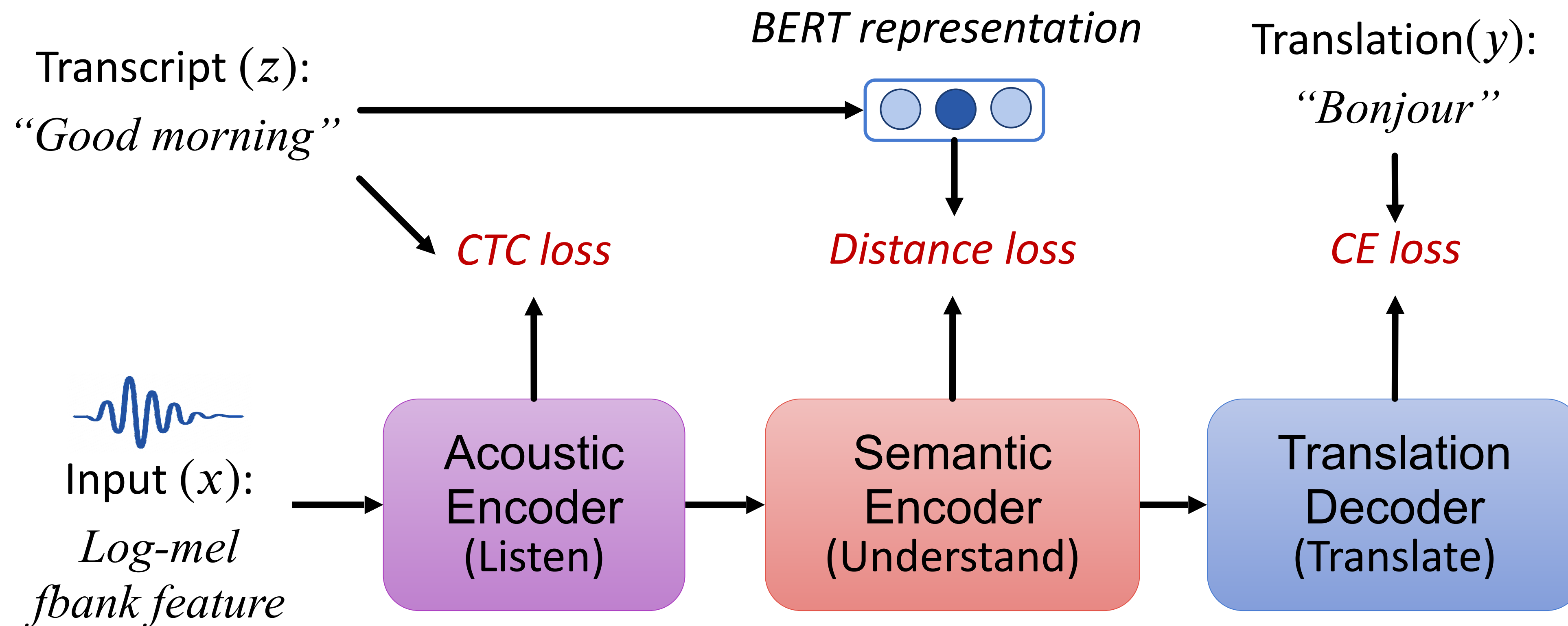
Drawback 2: Limit in using “transcript” info.

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

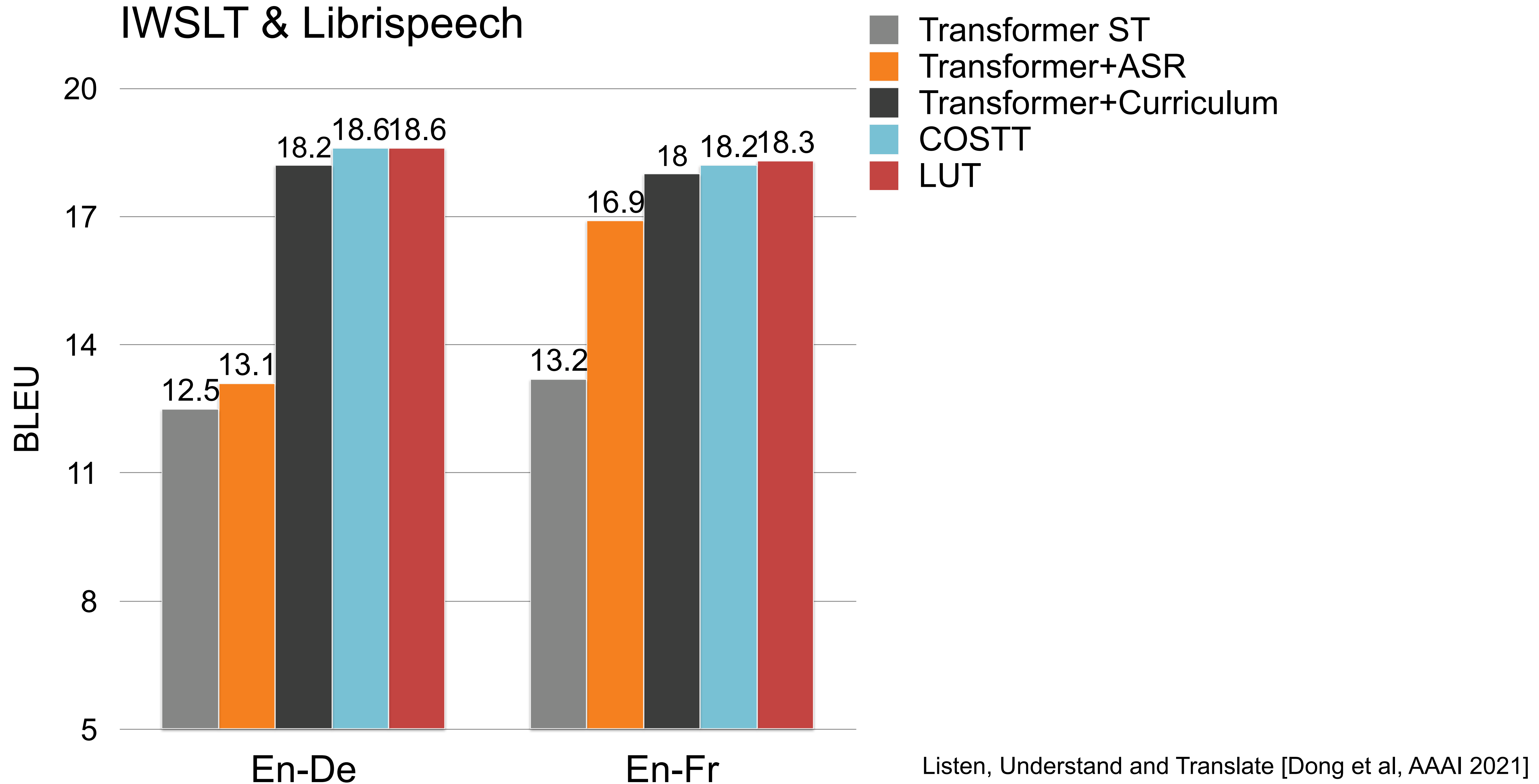
LUT: Utilizing Pre-trained Model on Raw Text

Training data: triples of

$\langle \text{speech}, \text{transcript_text}, \text{translate_text} \rangle$

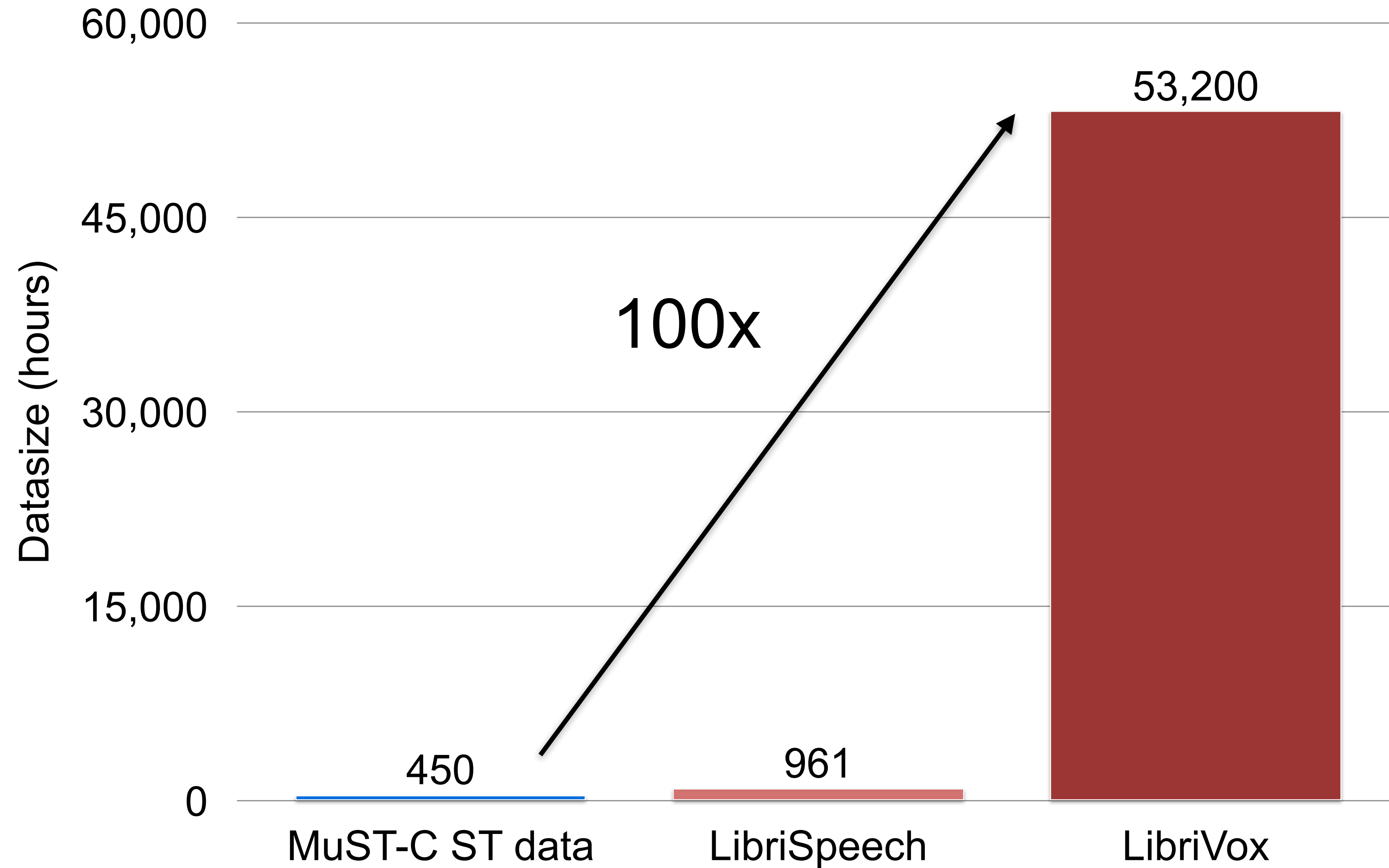


ST Benefits from BERT, with Raw Text Pre-training



Audio Pre-training

Dataset size
ST vs raw Audio

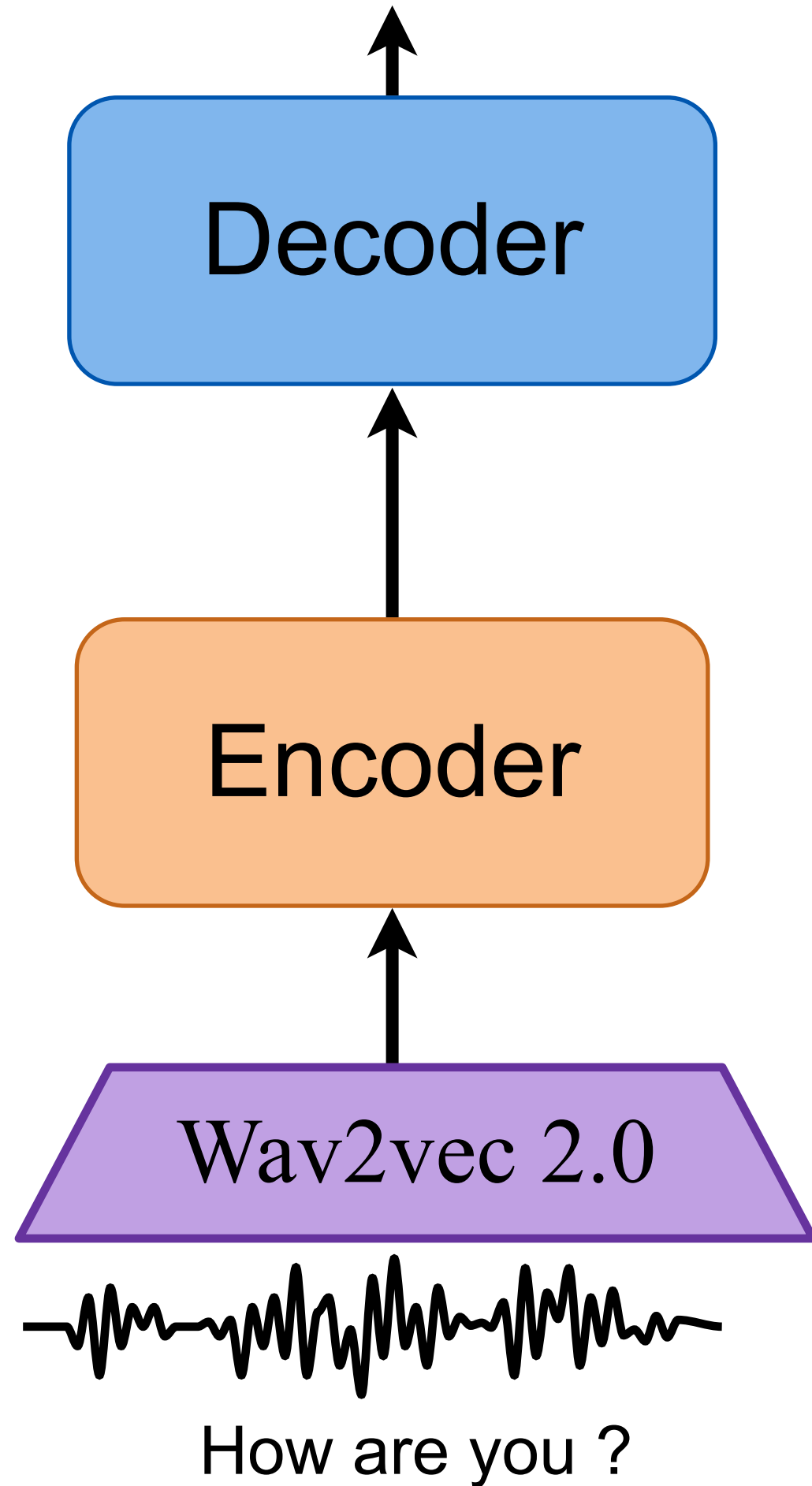


How to use
larger raw audio
data to improve ST
performance?

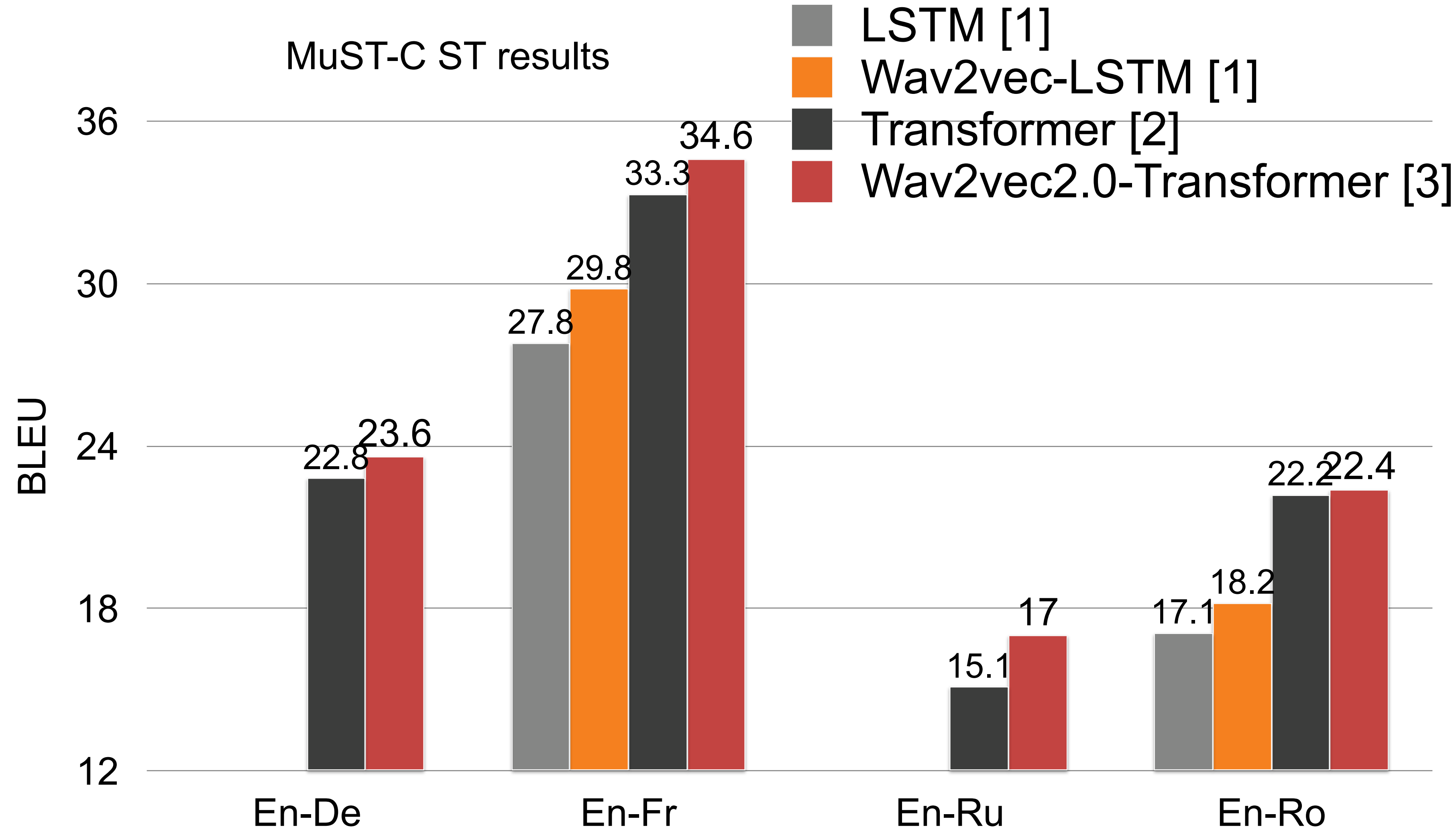
Speech Translation with Audio-Pretrain

Wav2vec Pretrain + Fine-tune on ST

Comment allez-vous ?



MuST-C ST results



[1] Self-supervised Representations improve end-to-end speech translation [Wu et al. InterSpeech 2020]

[2] NeurST toolkit [Zhao et al ACL2021 demo]

[3] End-to-end Speech Translation [Ye et al. InterSpeech 2021]

Self-training with Audio data

Step 0. Audio-only pre-training for Wav2vec2.0

Step 1. Freeze Wav2vec2.0, train on ST

Step 2. Self-train on generated pseudo-translation with LibriVox audio

Comment allez-vous ?

Transformer Decoder

Wav2vec 2.0

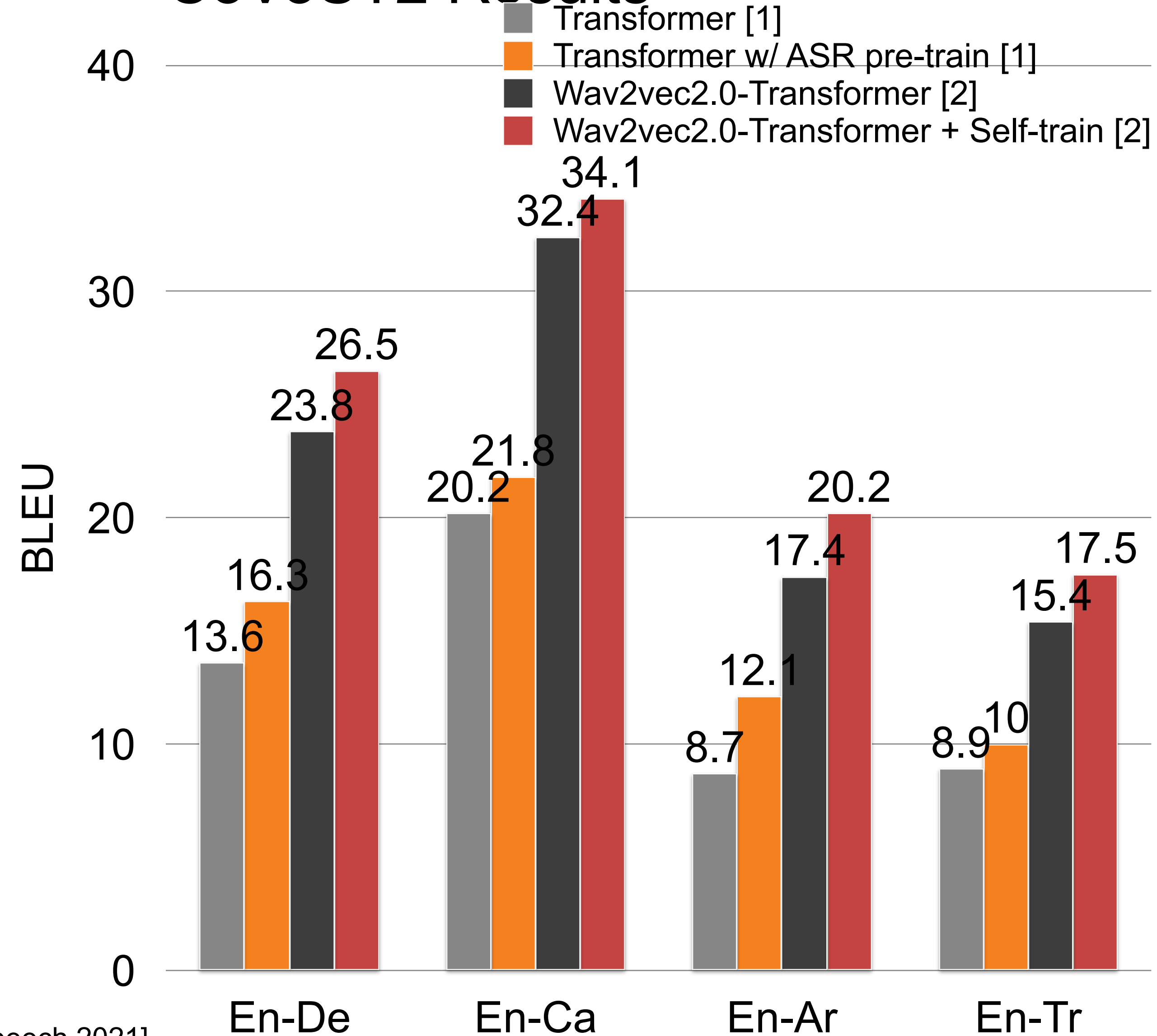
Transformer

CNN



How are you ?

CoVoST2 Results

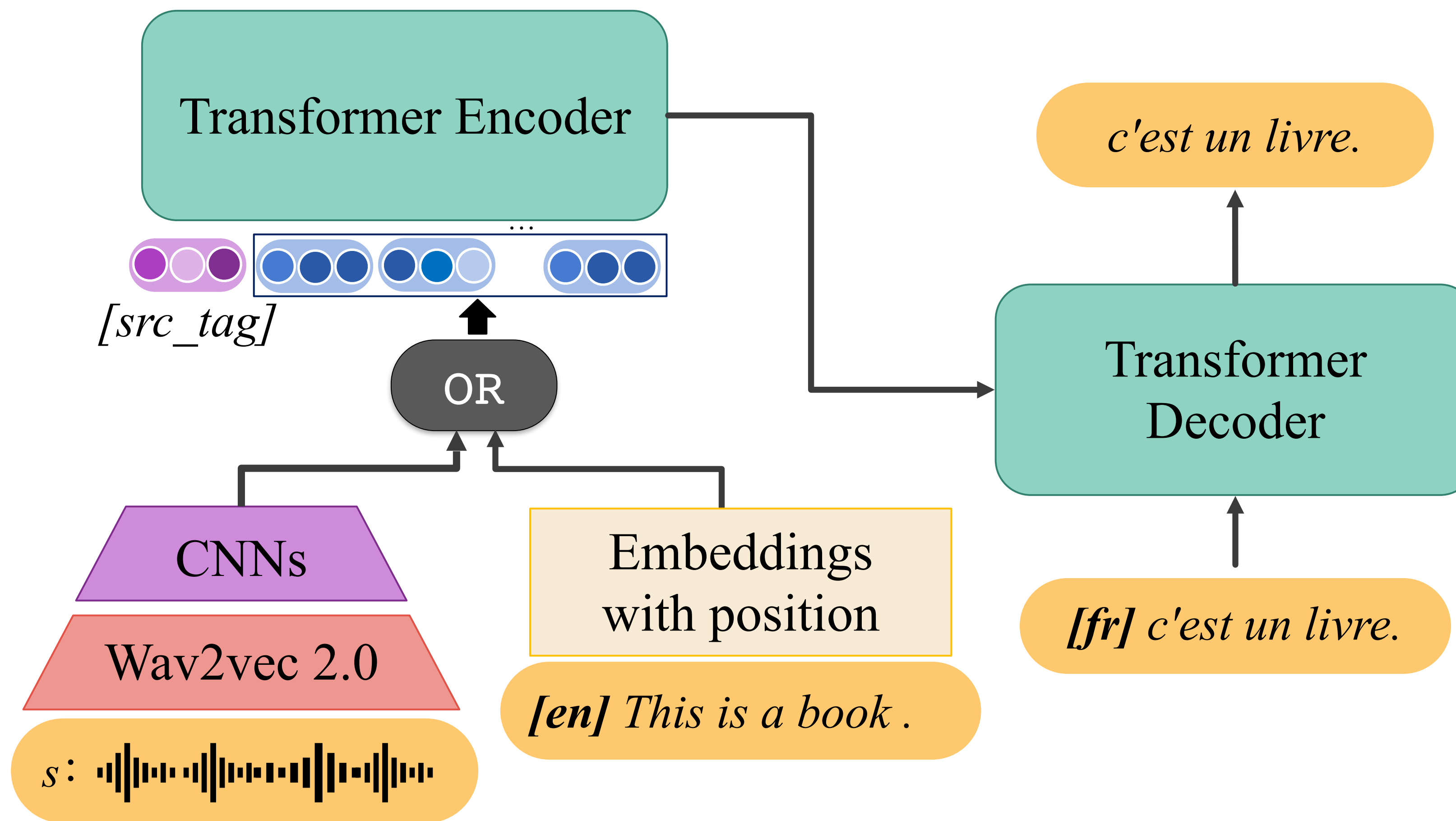


[1] CoVoST 2 and Massively Multilingual Speech-to-Text Translation, [Wang et al InterSpeech 2021]

[2] Large-Scale Self- and Semi-Supervised Learning for Speech Translation [Wang et al. 2021]

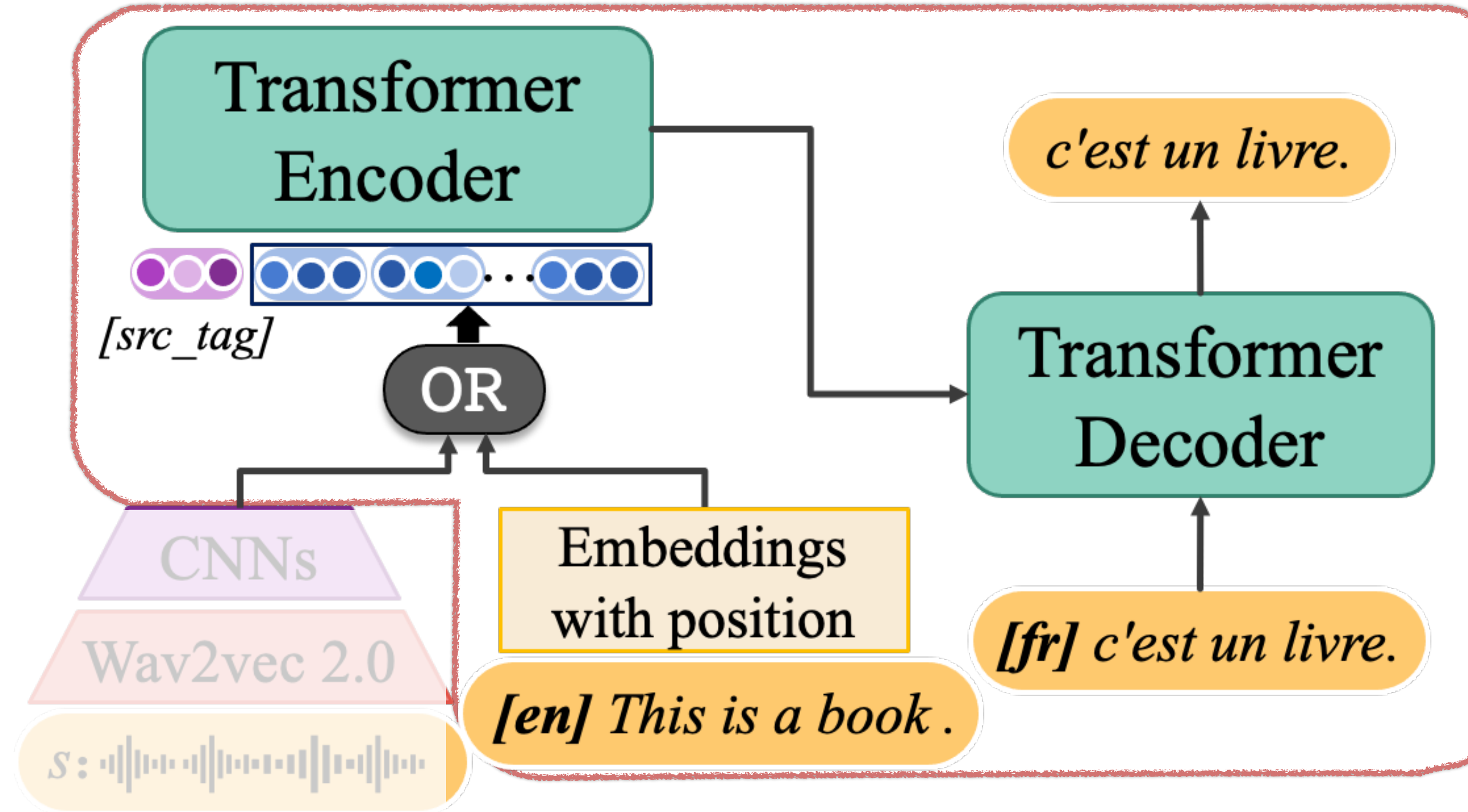
Fine-tuning Strategy for ST

Cross Speech-Text Network (XSTNet)



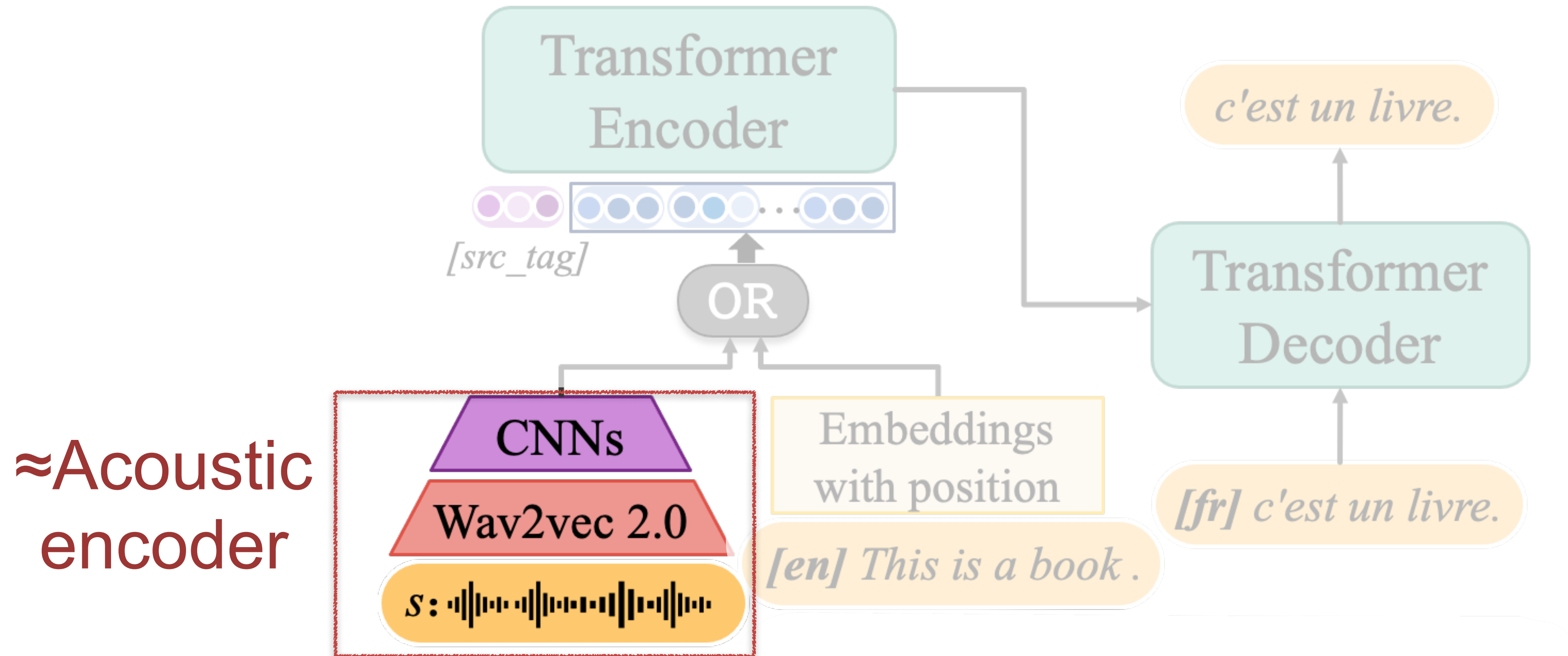
Supports to train MT data

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




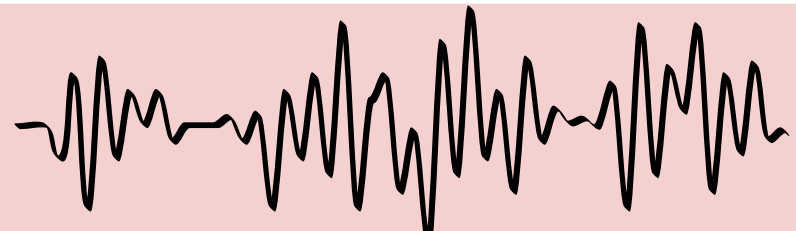
Supports inputs of two modalities

- ☑ Wav2vec2.0^[1] as the acoustic encoder
- ☑ We add two convolution layers with 2-stride to shrink the length.



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 $\langle \text{speech}, \text{translation} \rangle$

(3) D_{ASR} with $\langle \text{speech}, \text{transcript} \rangle$

Progressive:

Don't stop

training D_{MT-ext}

XSTNet achieves State-of-the-art Performance

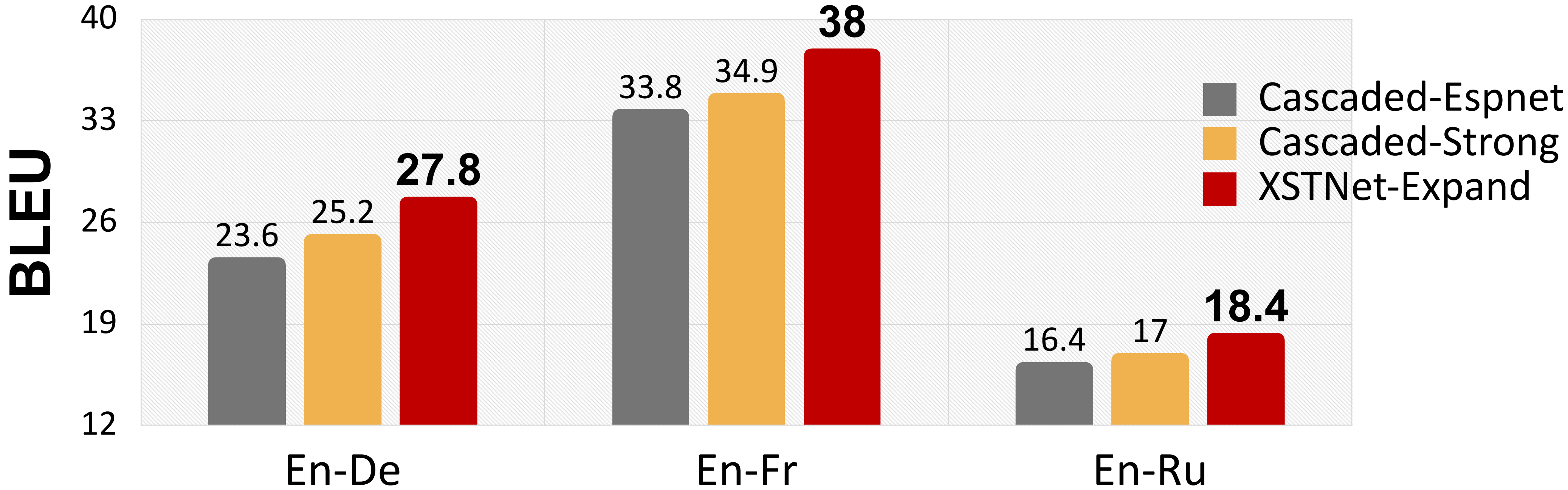
Models	External Data	Pre-train Tasks	De	Es	Fr	It	Nl	Pt	Ro	Ru	Avg.
Transformer ST [13]	×	ASR	22.8	27.4	33.3	22.9	27.2	28.7	22.2	15.1	24.9
AFS [31]	×	×	22.4	26.9	31.6	23.0	24.9	26.3	21.0	14.7	23.9
Dual-Decoder Transf. [15]	×	×	23.6	28.1	33.5	24.2	27.6	30.0	22.9	15.2	25.6
Tang et al. [2]	MT	ASR, MT	23.9	28.6	33.1	-	-	-	-	-	-
FAT-ST (Big) [6]	ASR, MT, mono-data [†]	FAT-MLM	25.5	30.8	-	-	30.1	-	-	-	-
W-Transf.	audio-only*	SSL*	23.6	28.4	34.6	24.0	29.0	29.6	22.4	14.4	25.7
XSTNet (Base)	audio-only*	SSL*	25.5	29.6	36.0	25.5	30.0	31.3	25.1	16.9	27.5
XSTNet (Expand)	MT, audio-only*	SSL*, MT	27.8[§]	30.8	38.0	26.4	31.2	32.4	25.7	18.5	28.8

Table 1: Performance (case-sensitive detokenized BLEU) on MuST-C test sets. [†]: “Mono-data” means audio-only data from Librispeech, Libri-Light, and text-only data from Europarl/Wiki Text; *: “Audio-only” data from LibriSpeech is used in the pre-training of wav2vec2.0-base module, and “SSL” means the self-supervised learning from unlabeled audio data. [§] uses OpenSubtitles as external MT 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?

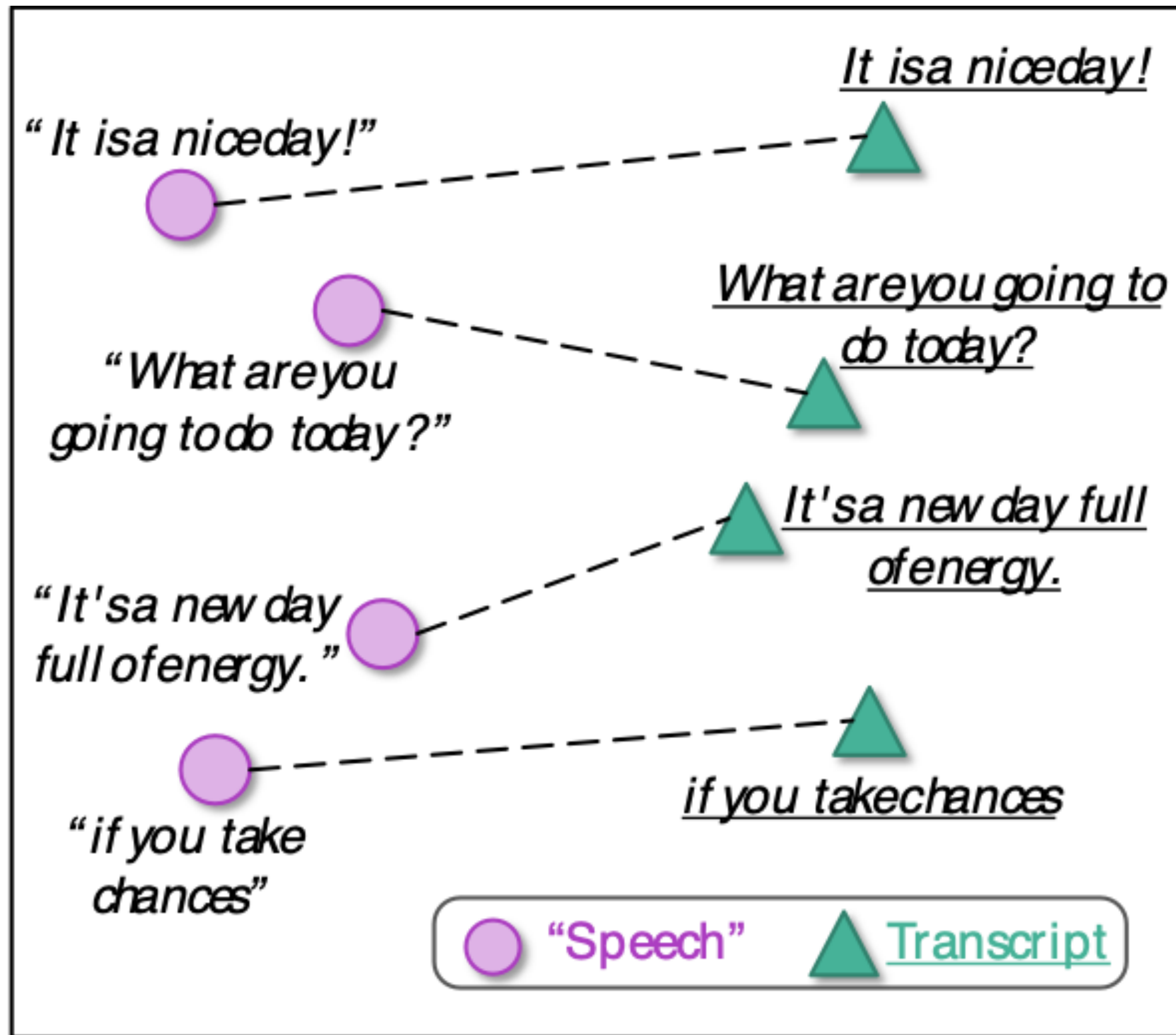
Strong ASR model + Large-scale MT data

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

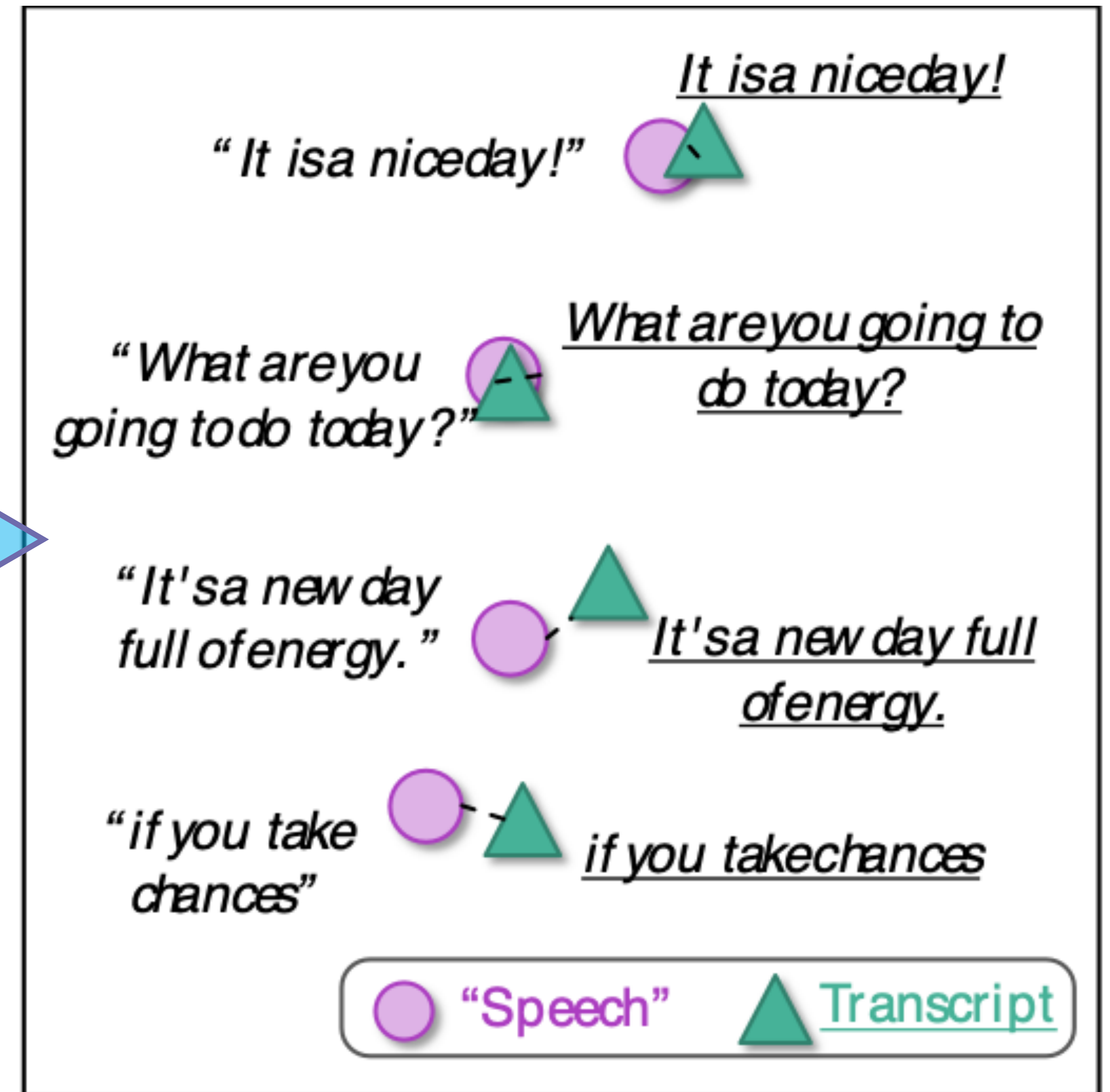
Learning Better Speech-Text Bimodal Representation

- ConST: Contrastive Learning to bridge the gap between text and speech [Ye et al 2022]
 - WACO: Contrastive learning at word-level with better aligned representation [Ouyang 2023]
- Chimera: Learning Fixed-size Shared Space for both audio and text, audio+MT pretraining [Han et al. 2021]
- Wav2vec2.0-mTransformer LNA: Use both audio pertaining + multilingual pertained language model, and selective efficient fine-tuning [Li et al. ACL 2021]
- FAT-ST: Masked pre-training for fused audio and text [Zheng et al. ICML 2021]

Text and speech with same meaning should be **similar** in representation!

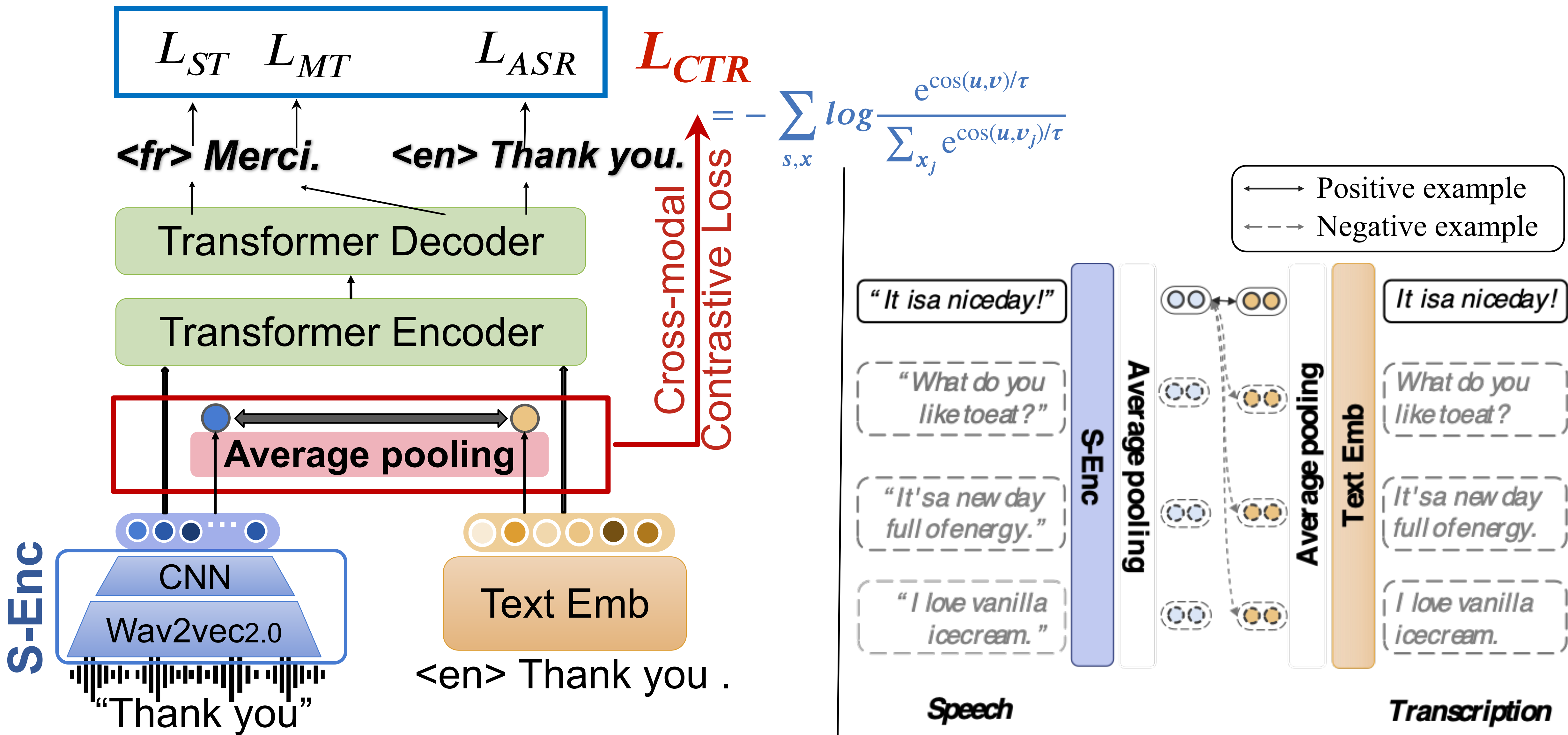


(a) Current models

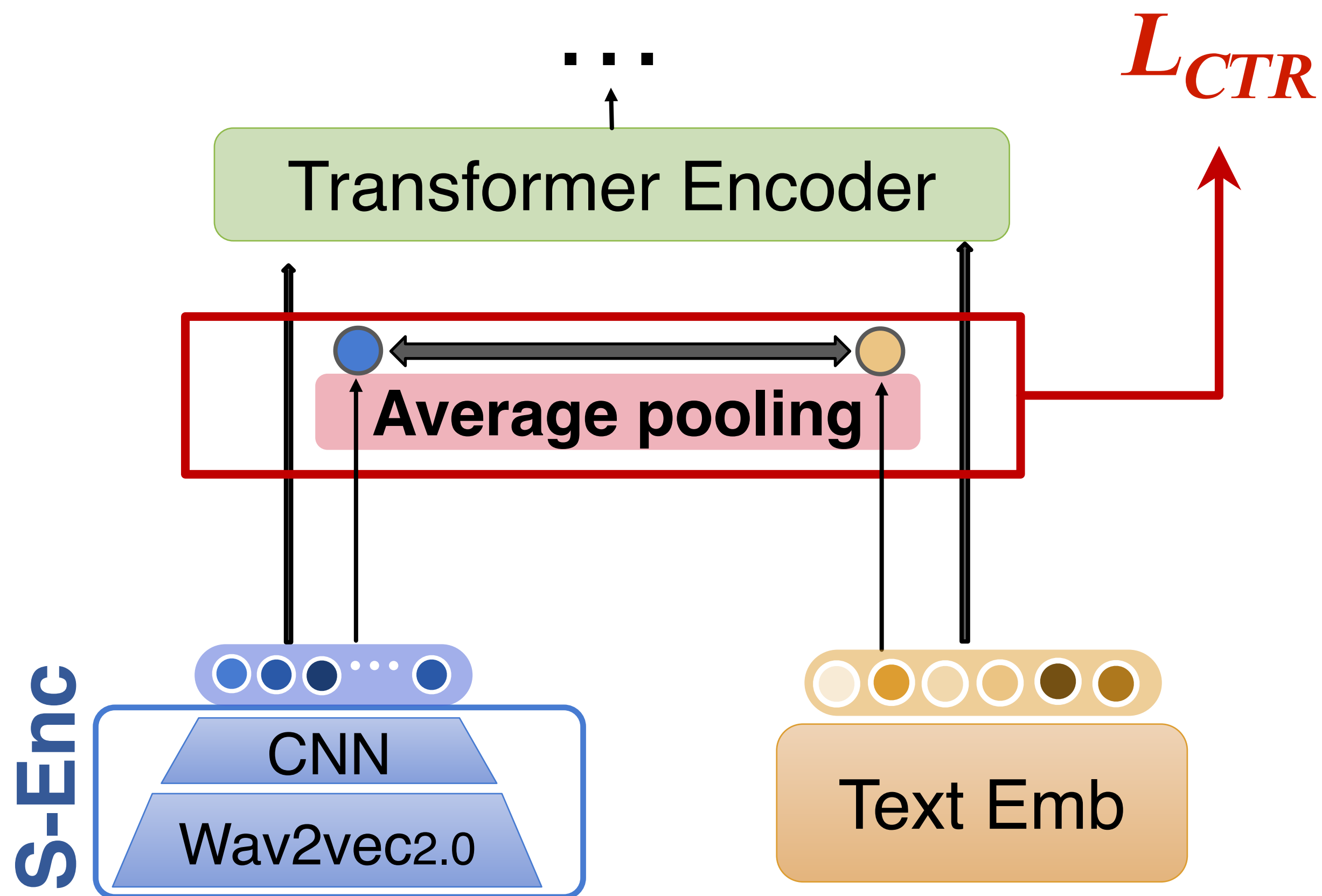


(b) Expected

onST Contrastive Learning for Speech Translation



Mining more hard examples



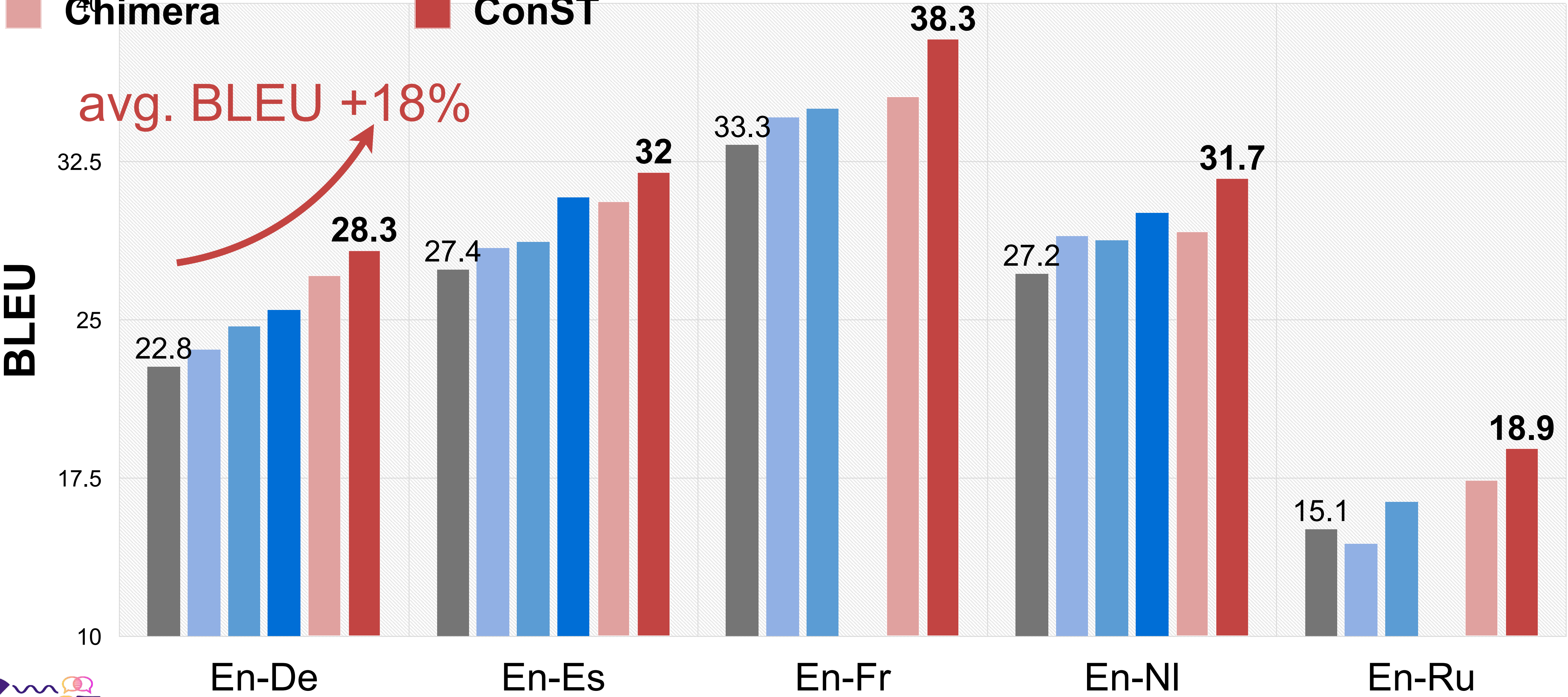
We introduce three hard example mining operations.

- ① Span-Masked Aug. (SMA)
- ② Word Repetition (Rep)
- ③ Cut-off

“Thank you”

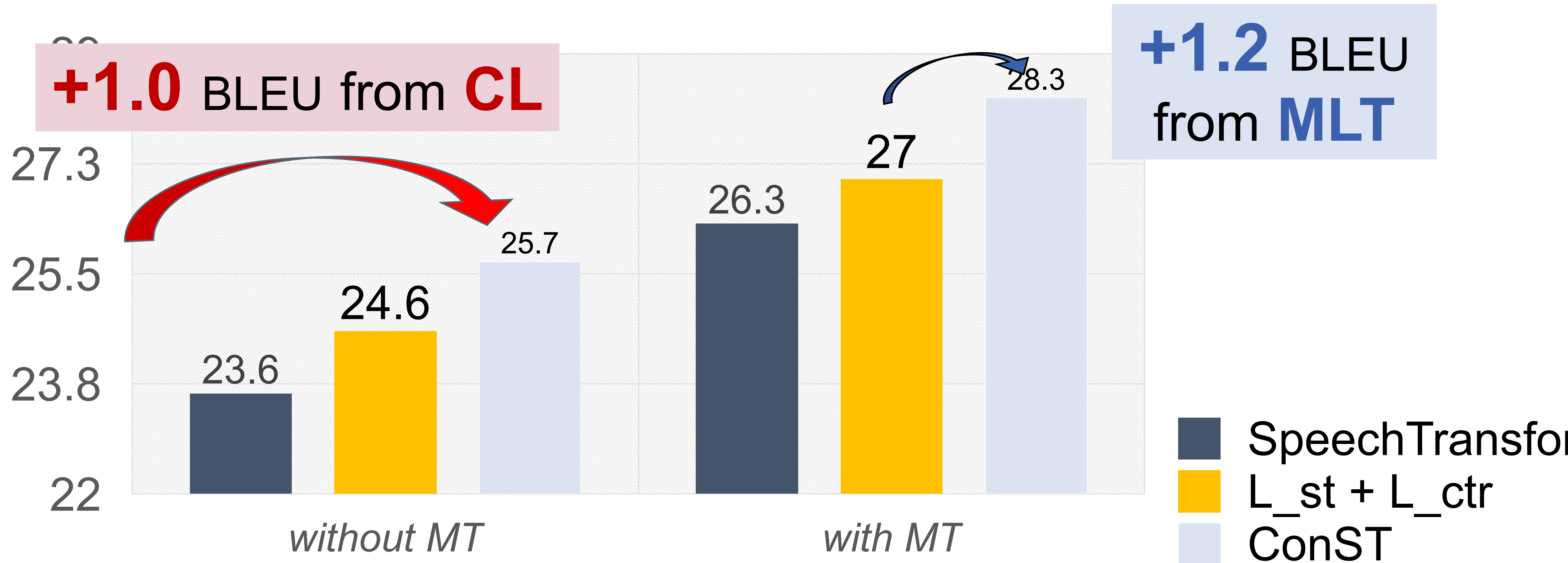
<en> Thank you .

Proposed ConST Significantly Improves Translation Performance



Both **Multi-task** and **Contrastive** Learning are important!

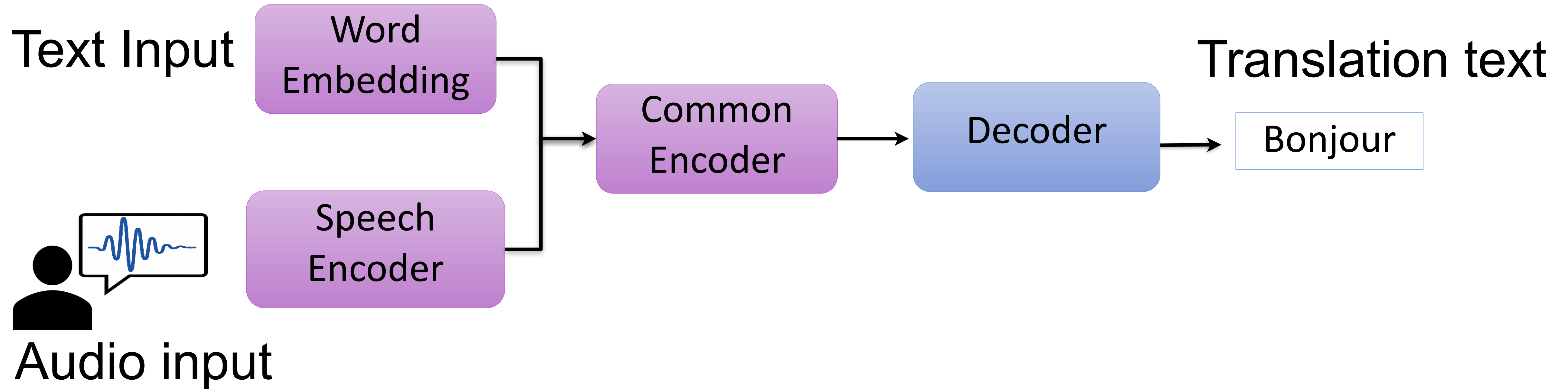
$$\mathcal{L} = \mathcal{L}_{ST} + \mathcal{L}_{ASR} + \mathcal{L}_{MT} + \lambda \mathcal{L}_{CTR}$$



+1.0 BLEU from **CL**

+1.2 BLEU from **MLT**

Bi-modal Encoding Architecture for ST

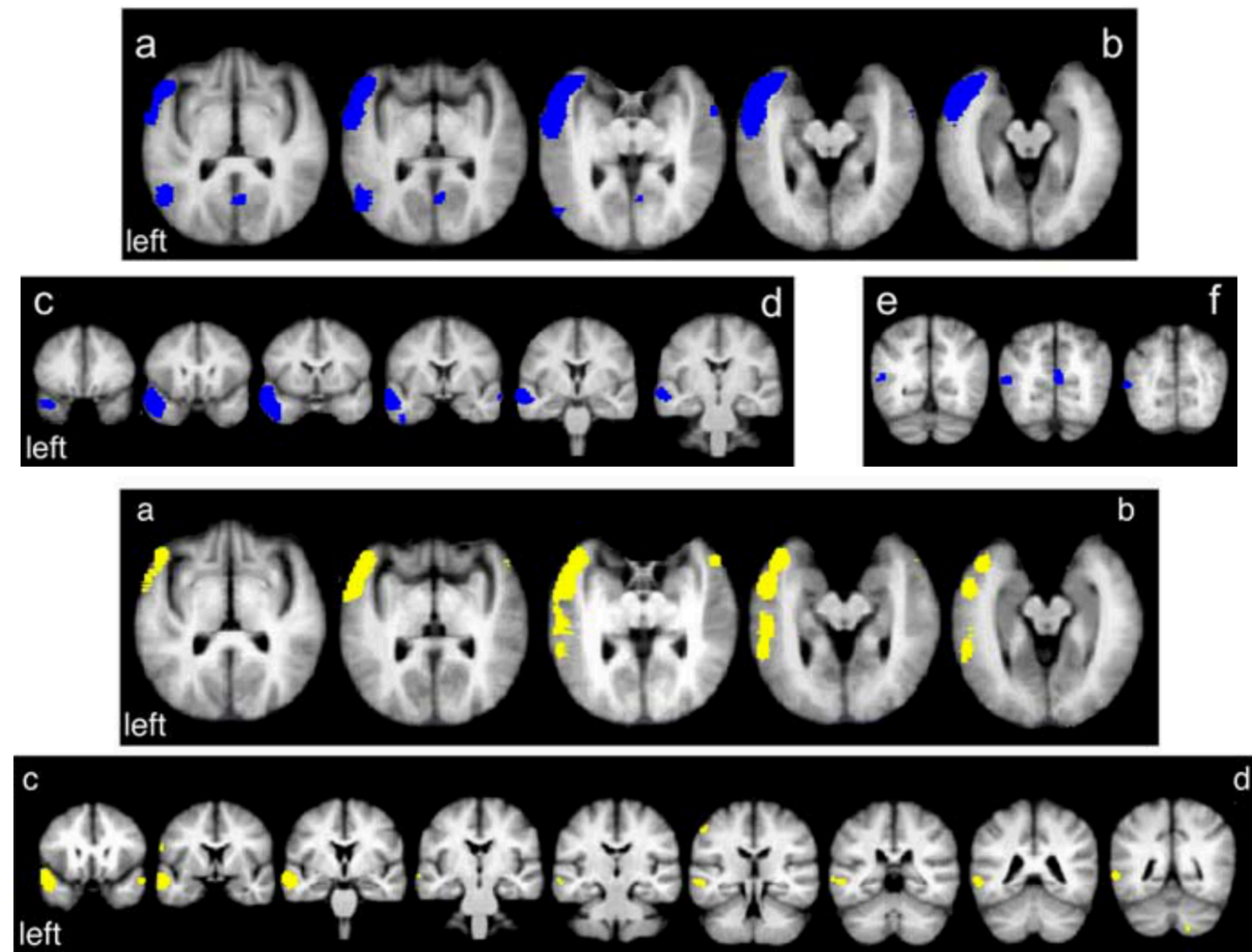


Challenges: gap between text and audio

1. Length: ~20 (text) vs. ~ 1k-10k (audio)
2. Embedding space disparity

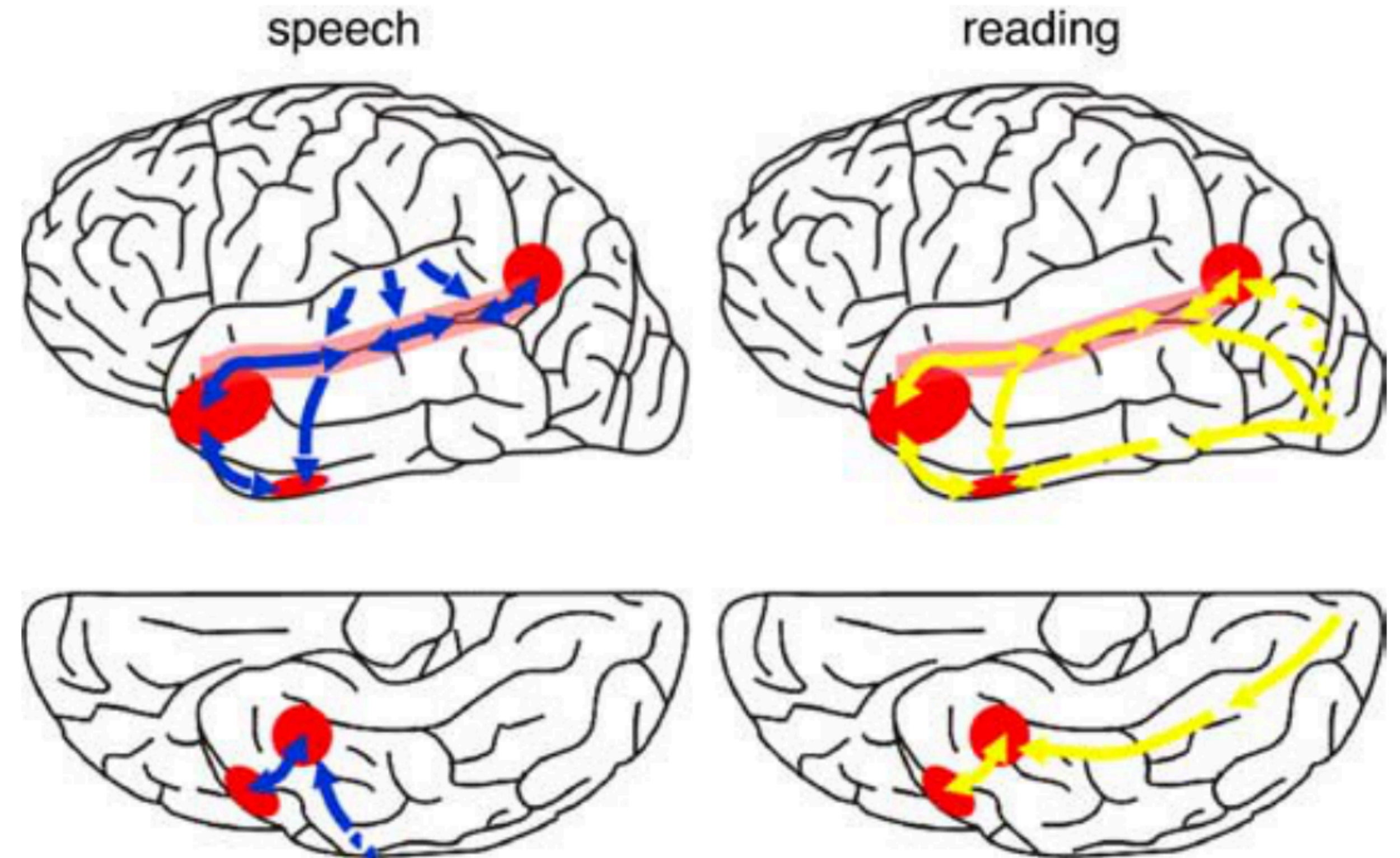
Insights from Cognitive Neuroscience

Speech and text interfere with each other in brain^[1]



activation map

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



processing paths

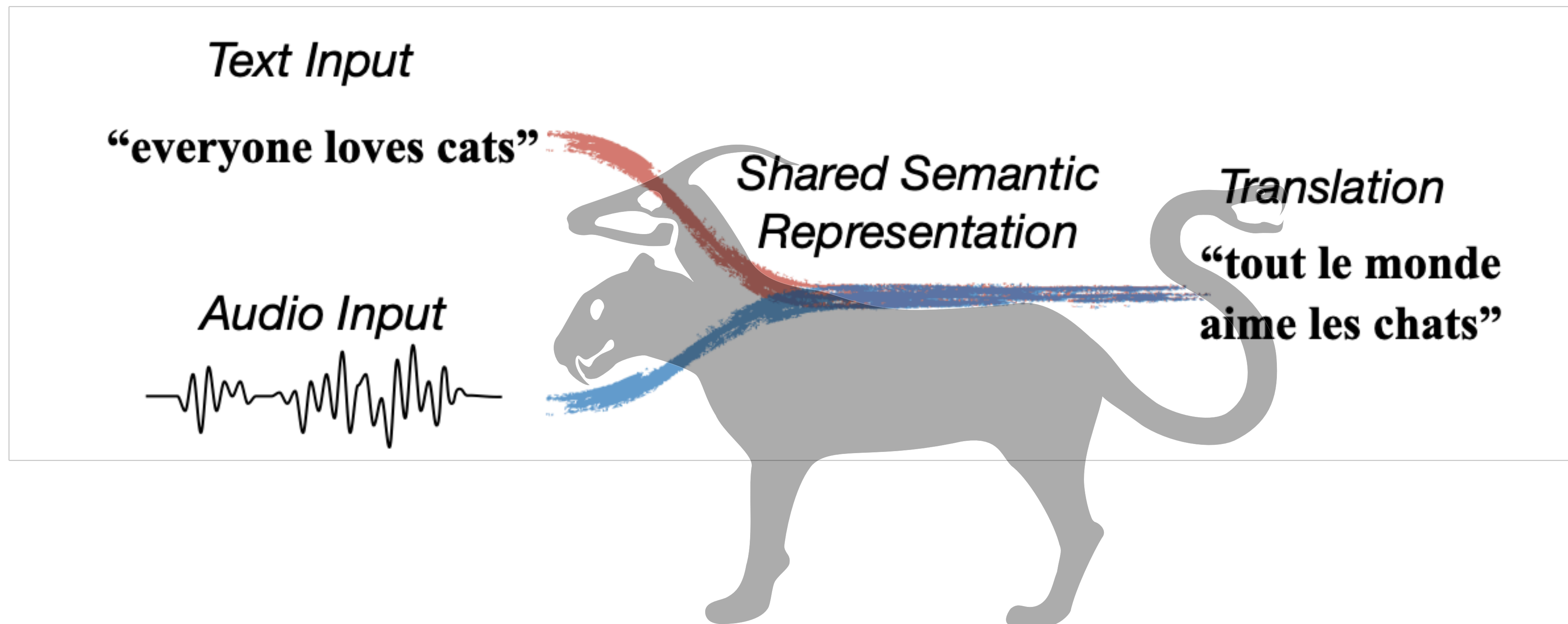
[1] Van Atteveldt, Nienke, et al. "Integration of letters and speech sounds in the human brain." *Neuron* 43.2 (2004): 271-282.

[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:

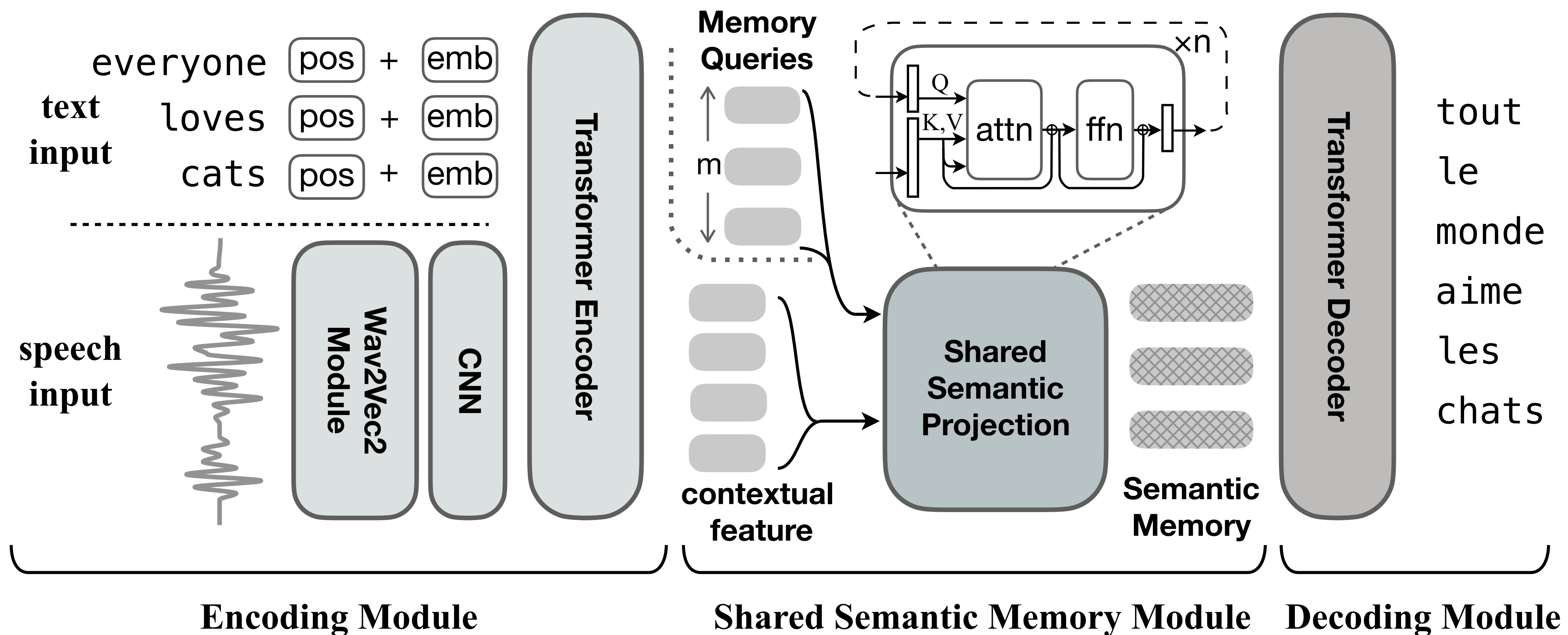
<speech, transcript_text, translate_text>



Chimera Model for ST

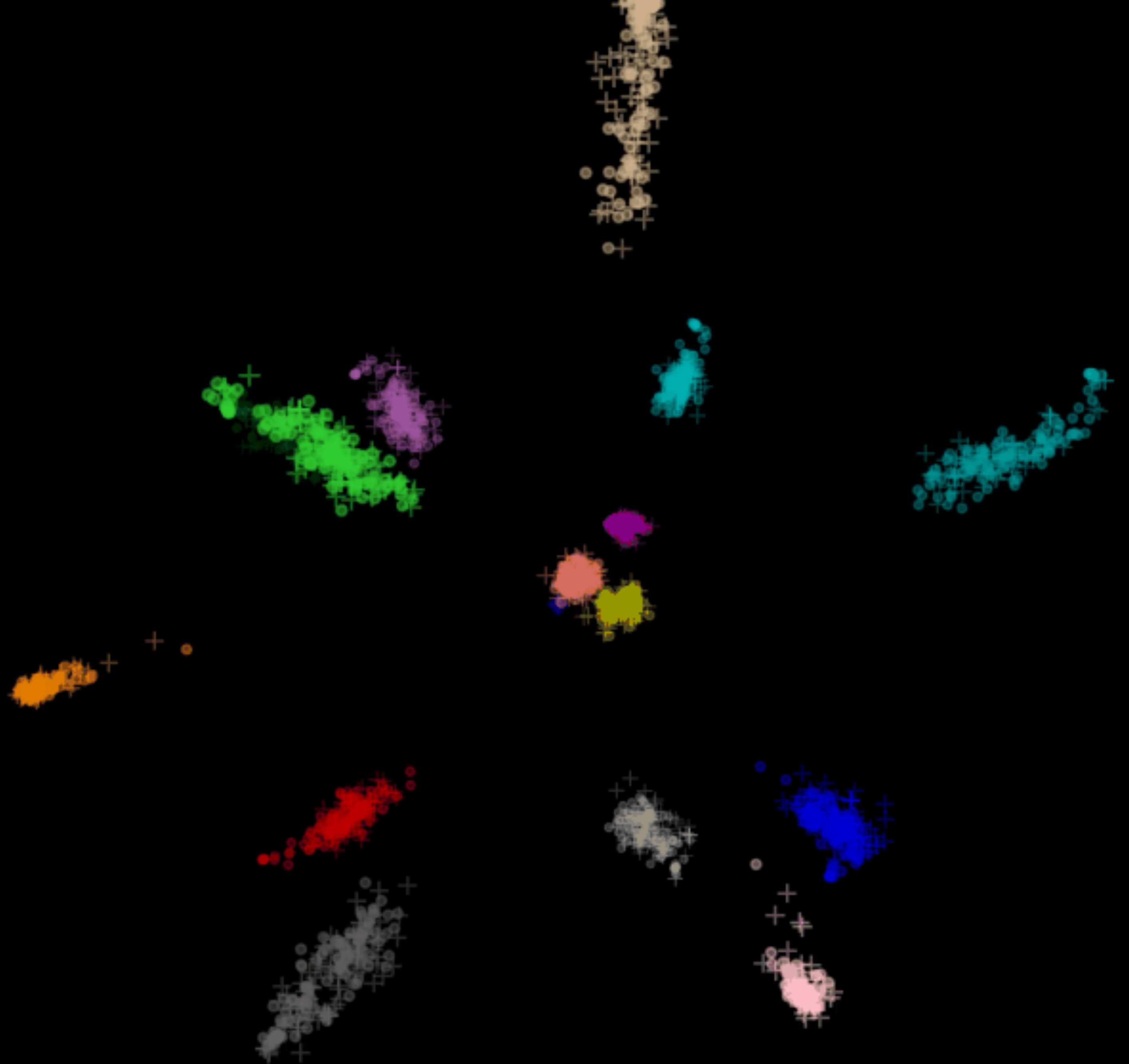
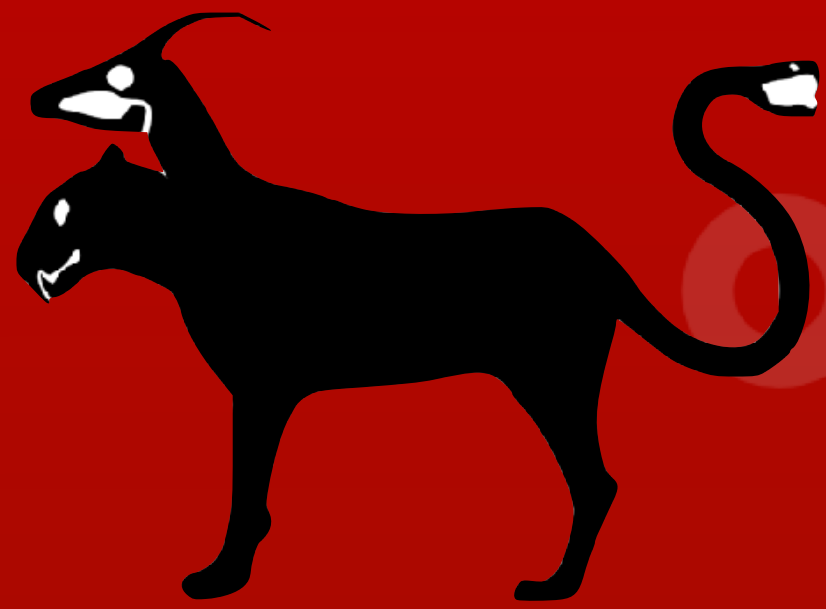
Training with auxiliary objectives: ST + MT + Contrastive loss

Benefit: able to **exploit large external MT data**





Shared Semantic Space Learned by Proposed Chimera



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 [†]	×	×	×	22.7	32.9	15.3	27.2	22.7	21.9	28.1	27.3
Espnet ST [‡]	×	×	×	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 [◇]	×	×	×	23.6	33.5	15.2	28.1	24.2	22.9	30.0	27.6
STATST [#]	×	×	×	23.1	-	-	-	-	-	-	-
MAML ^b	×	×	✓	22.1	34.1	-	-	-	-	-	-
Self-Training [◦]	✓	✓	×	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	✓	×	✓	27.1 [•]	35.6	17.4	30.6	25.0	24.0	30.2	29.2

Audio and Multilingual Text Pretrain for Multilingual ST

Comment allez-vous ?

Transformer
Decoder

CNN

Wav2vec 2.0
Transformer

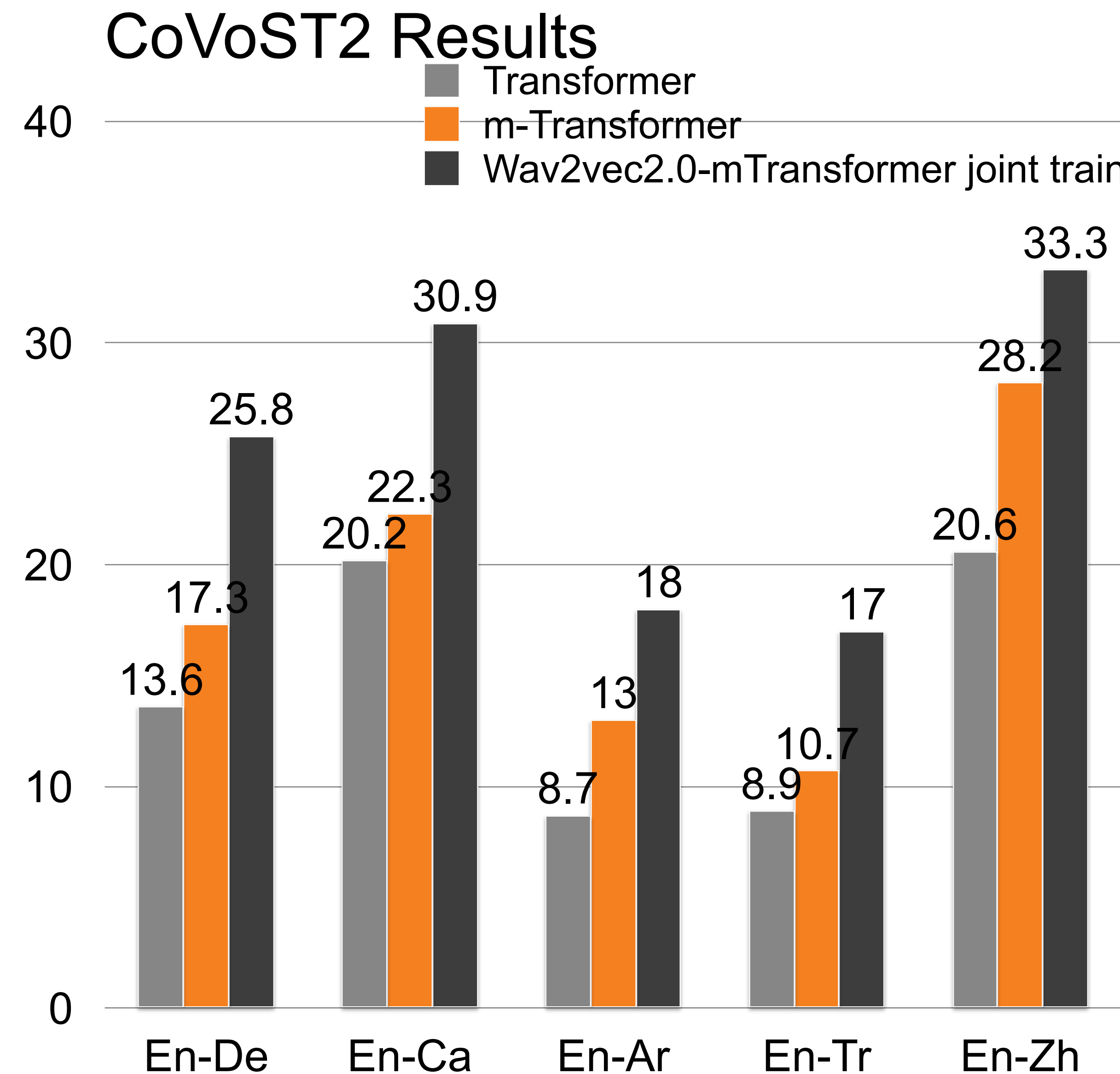
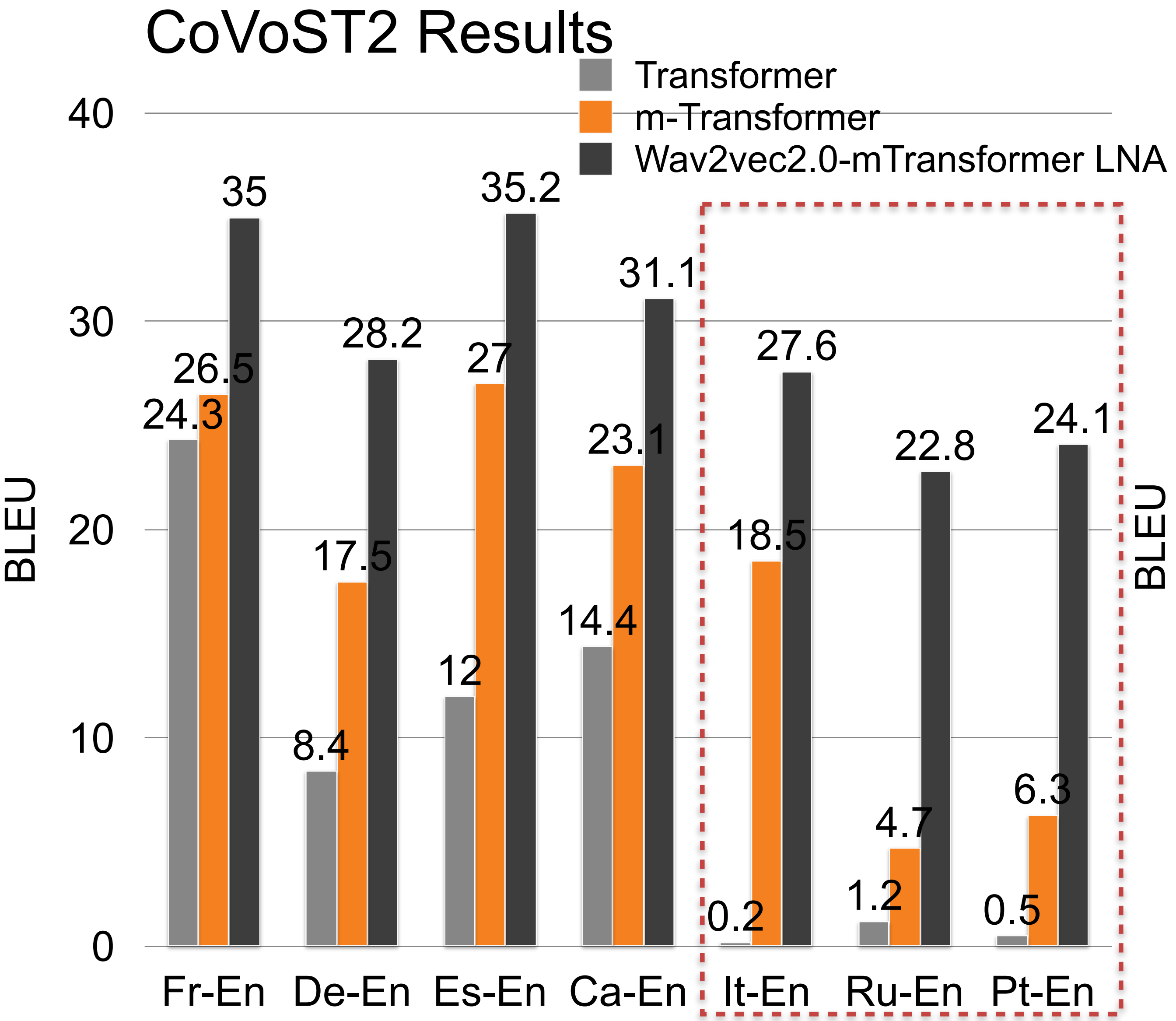
CNN



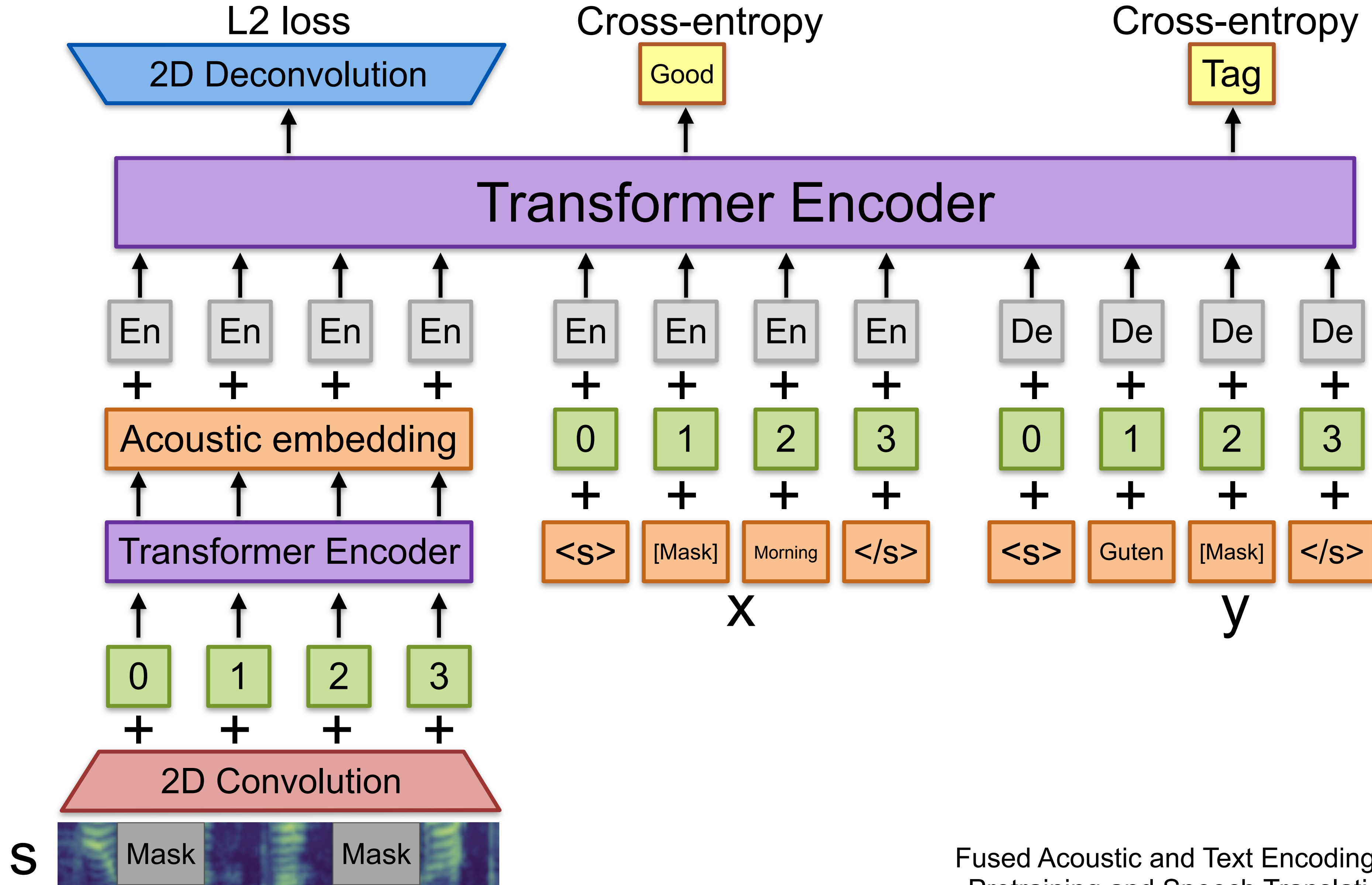
How are you ?

- Encoder uses Wav2vec2.0 pre-trained on LibriVox-60k audio
- Decoder: mBart pre-trained on 50 monolingual text and 49 bitext
- ST finetune strategy (LNA):
 - Only fine-tune layer-norm and attention layers
- MT+ST multitask joint train with further parallel bitext data

Wav2vec2.0 retraining + Multilingual training effectively transfers to low resource source language



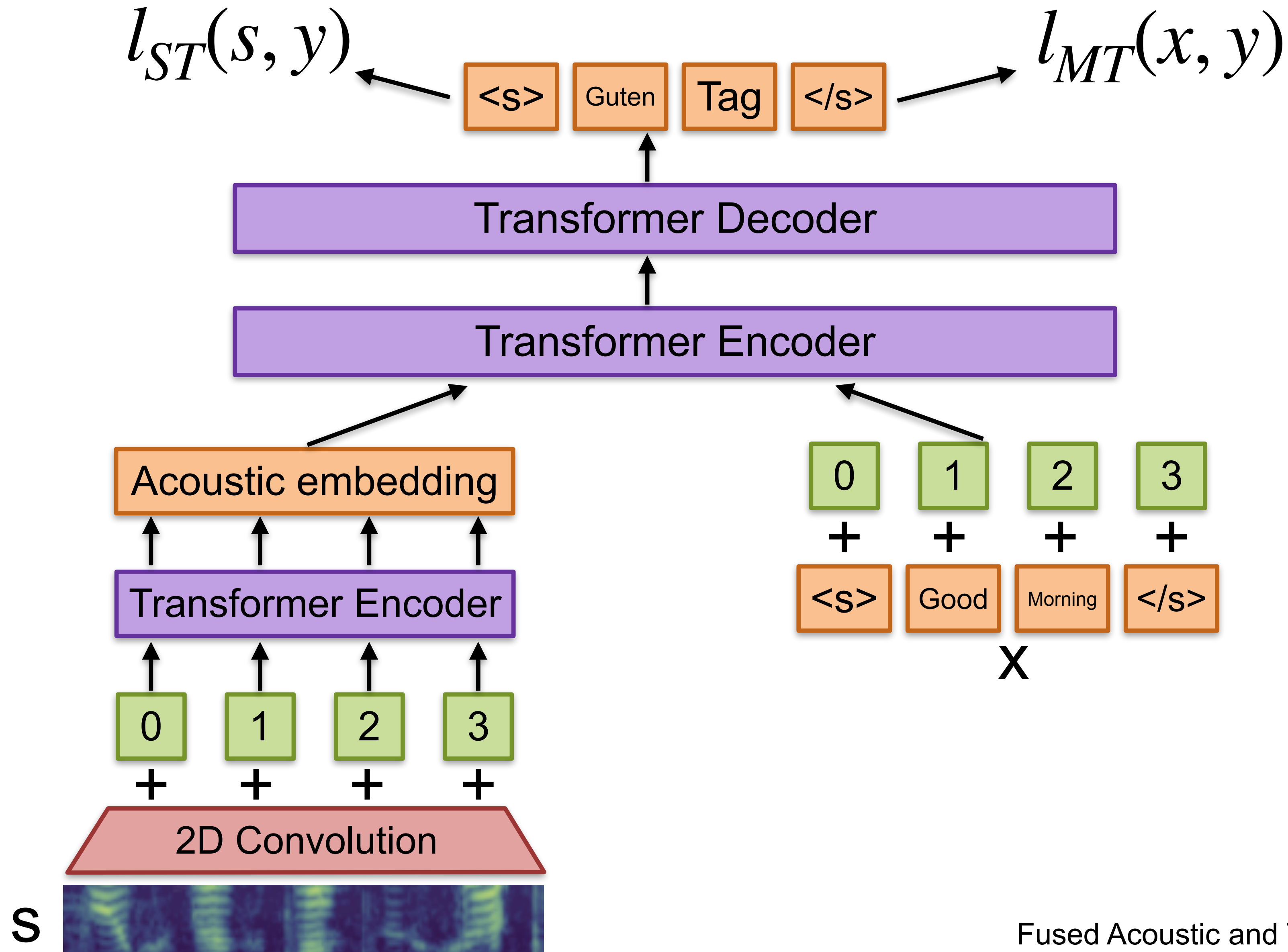
Fused Acoustic and Text Masked Language Model (FAT-MLM)



Pre-training data

1. Librispeech ASR 960h
2. Libri-light audio 3,748h
3. Europarl/wiki text 2.3M
4. MuST-C 408h
5. Europarl MT 1.9M

FAT-ST



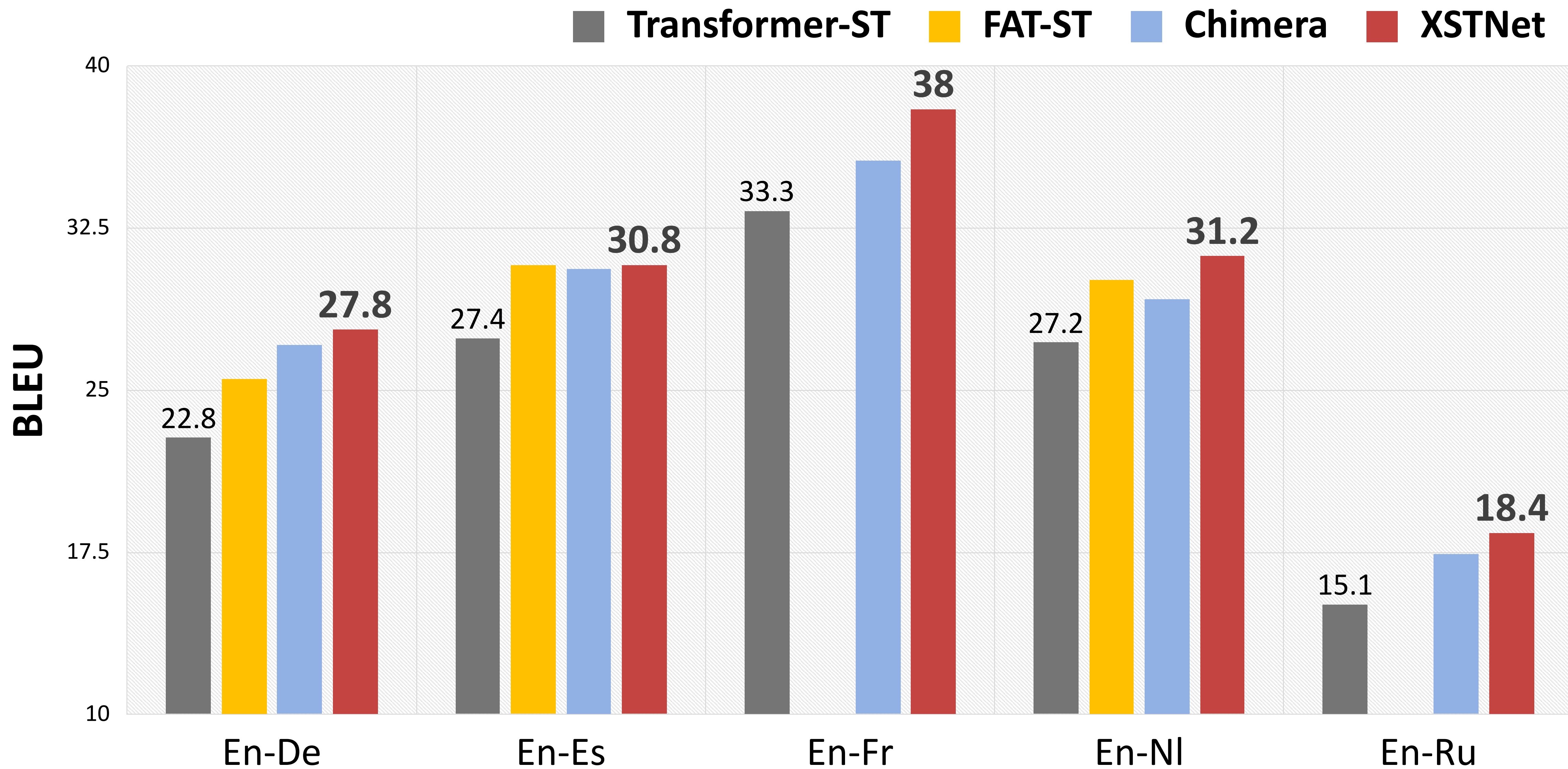
Training:

- Pre-train FAT-MLM with all data
- Init FAT-ST with FAT-MLM, decoder copy encoder
- Further fine-tune on MuST-C ST data.

Joint audio&text Pre-training task helps ST

Pretrain Method	Models	En→De	En→Es	En→Nl	Avg.	Model Size
No Pretraining	ST	19.64	23.68	23.01	22.11	31.25M
	ST + ASR	21.70	26.83	25.44	24.66 (+2.55)	44.82M
	ST + ASR & MT	21.58	26.37	26.17	24.71 (+2.60)	56.81M
	ST + MAM	20.78	25.34	24.46	23.53 (+1.42)	33.15M
	ST + MAM + ASR	22.41	26.89	26.49	25.26 (+3.15)	46.72M
	Liu et al. (2020b)	22.55	-	-	-	-
	Le et al. (2020)	23.63	28.12	27.55	26.43 (+4.32)	51.20M
	Cascade [§]	23.65	28.68	27.91	26.75 (+4.64)	83.79M
<hr/>						
ASR & MT	FAT-ST (base).	22.70	27.86	27.03	25.86 (+3.75)	39.34M
	<hr/>					
ASR & MT	ST	21.95	26.83	26.03	24.94 (+2.83)	31.25M
	ST + ASR & MT	22.05	26.95	26.15	25.05 (+2.94)	56.81M
<hr/>						
MAM	FAT-ST (base)	22.29	27.21	26.26	25.25 (+3.14)	39.34M
<hr/>						
FAT-MLM	FAT-ST (base)	23.68	28.61	27.84	26.71 (+4.60)	39.34M
	FAT-ST (big)	23.64	29.00	27.64	26.76 (+4.65)	58.25M

Pre-training Improves ST Performance



Summary

	Direct Supervision	Contrastive	Masked LM	Knowledge distillation	Progressive train	Selective Fine-tune	Self-training
MT Parallel Text	COSTT			[Liu et al. 2019]	XSTNet		
ASR Speech-Transcript	LUT	ConST, WACO					
Audio-only		Wav2vec Wav2vec 2.0					[Wang et al. 2021]
Raw text				LUT			
Speech+Text		Chimera	FAT-ST		XSTNet	LNA	

Language in 10
