

# 11492/11692/18495

## Speech Processing

### Lecture 14: Speech-to-Speech Translation

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# TA Introduction

- 3rd Year Ph.D. Student
- Main research focus:
  - speech representation learning and its application
- Broad interests in many downstream tasks:
  - Typical speech tasks: ASR & TTS & ST & SLU
    - architectures
    - decoding
    - aspects in low-resource and multilingual
  - Related music tasks
    - singing voice synthesis
    - singing voice conversion
    - music generation



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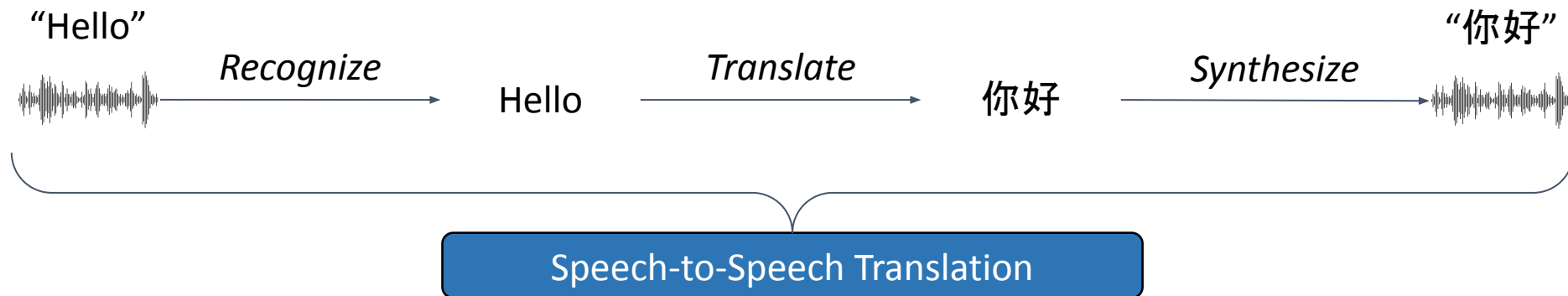
- Speech-to-speech Translation (S2ST)
  - **Introduction**
  - Evaluation metrics
  - Famous datasets and benchmarks
  - Technical overviews
  - References

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- Speaker recognition
  - **Introduction**
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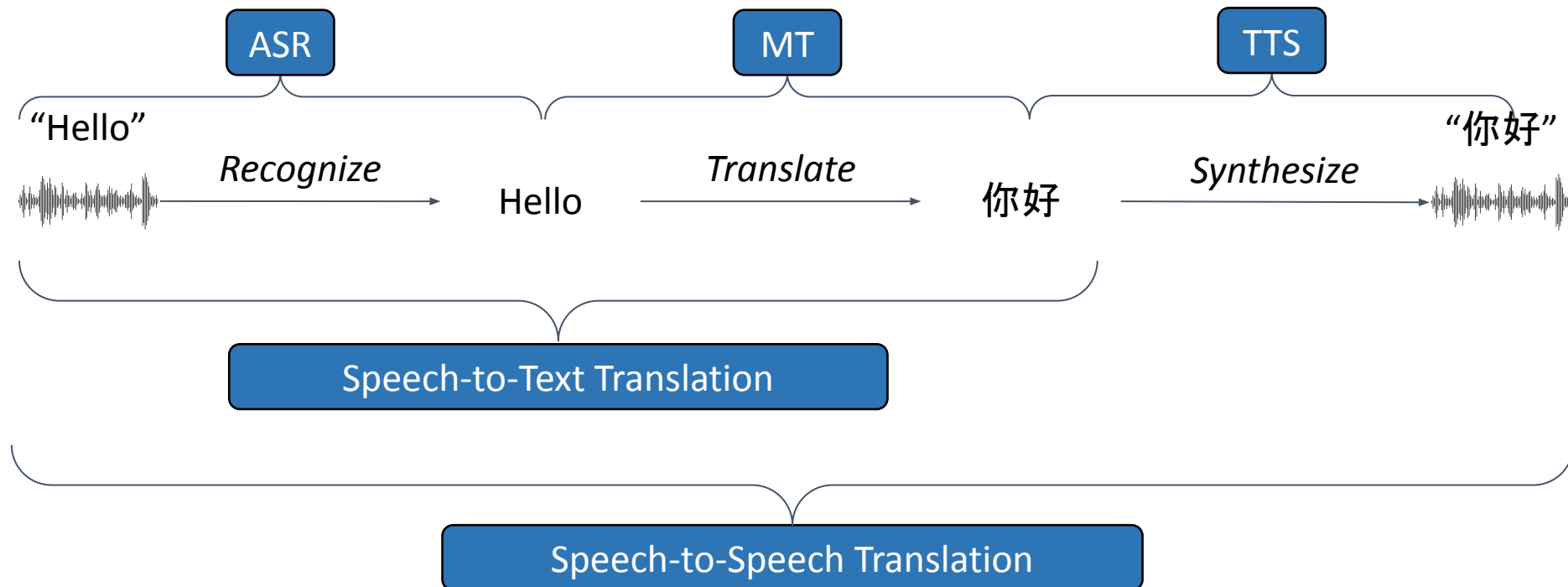
# Speech-to-speech Translation

- Converts source language speech into target language text / speech
  - **Sequence Transduction Task:** sequence in, sequence out
  - **Compositional Task:** naturally decomposes into subtasks



# Speech Translation

- Converts source language speech into target language text / speech
  - **Sequence Transduction Task:** sequence in, sequence out
  - **Compositional Task:** naturally decomposes into subtasks



# More on system construction

- Shinji has introduced the general concepts of speech translation, including a section for speech-to-speech translation.
- Today, we will focus more on how to build the system of speech-to-speech translation (S2ST)

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- Speech-to-speech translation
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# Speech-to-Speech Translation Metrics

- For speech-to-speech translation, we want to know **the translation quality and the synthesis quality**
- Metrics
  - ASR-BLEU (objective)
  - Naturalness (subjective)
  - Speaker similarity (subjective)
  - EER on speaker (objective)
  - ...

# Speech-to-Speech Translation Metrics

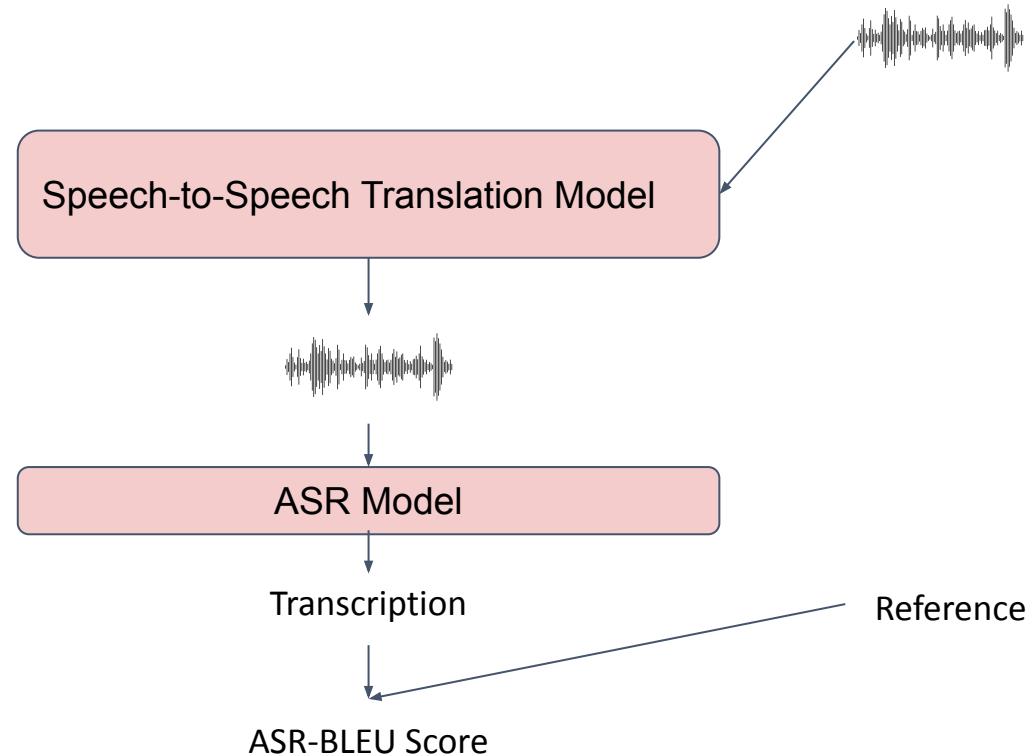
- For speech-to-speech translation, we want to know **the translation quality and the synthesis quality**
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  - ASR-BLEU (objective)
  - Naturalness (subjective)
  - Speaker similarity (subjective)
  - EER on speaker (objective)
  - ...

Note that we do not have monotonic assumption in speech-to-speech translation, so we can not use:

- **WER**
- **MCD**
- **F0 RMSE**

# ASR-BLEU

- Automatically evaluate speech-to-speech translation models by
  - 1) feeding speech outputs to an **ASR model**,
  - 2) then scoring **BLEU** on the ASR model's transcriptions



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# Frequently Used Corpora

- CVSS (synthesized) - <https://github.com/google-research-datasets/cvss>
  - X-to-En (21 languages); speech from CommonVoice
  - 2 versions: CVSS-C (easier) and CVSS-T (harder); former has a single target speaker voice, latter has multiple and each target matches the source speaker voice
- LibriS2S (synthesized) - <https://github.com/PedroDKE/LibriS2S>
  - En-to-De and De-to-En: speech from librivox audio books
  - Note: no speaker matching
- Voxpopuli (real-world) - <https://github.com/facebookresearch/voxpathuli>
  - 15x15 directions of language pairs: speech from parliament speech
  - Note: no speaker matching
- SpeechMatrix (real-world) - [https://github.com/facebookresearch/fairseq/tree/ust/examples/speech\\_matrix](https://github.com/facebookresearch/fairseq/tree/ust/examples/speech_matrix)
  - 17x17 directions of language pairs: speech from parliament speech
  - Note: no speaker matching
- SeamlessM4T (real-world) - [https://github.com/facebookresearch/seamless\\_communication/blob/main/docs/m4t/seamless\\_align\\_README.md](https://github.com/facebookresearch/seamless_communication/blob/main/docs/m4t/seamless_align_README.md)
  - Note: no speaker matching
  - Initially ~100 directions, now expanding to ~160 directions

# Shared Tasks

- The International Conference on Spoken Language Translation (**IWSLT**) is an annual scientific conference, associated with an **open evaluation campaign on spoken language translation**, where both scientific papers and system descriptions are presented.
- <https://iwslt.org/2024/>
- The speech-to-speech translation is a very hot topic in research community and it is included in **three** tracks:
  - Simultaneous track
  - Speech-to-speech track
  - Dubbing track

## Shared Tasks

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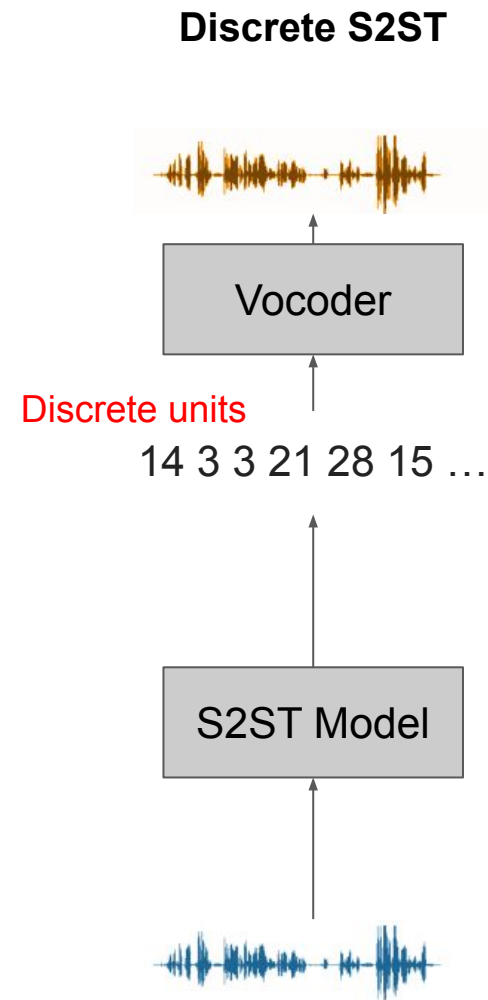
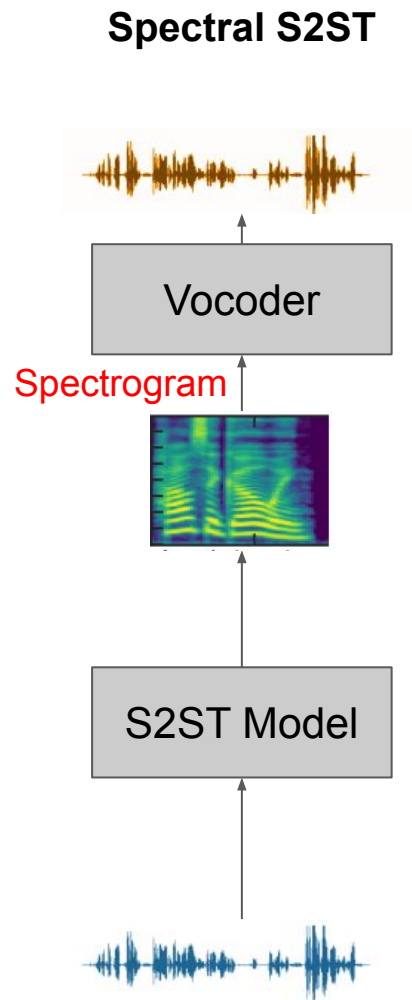
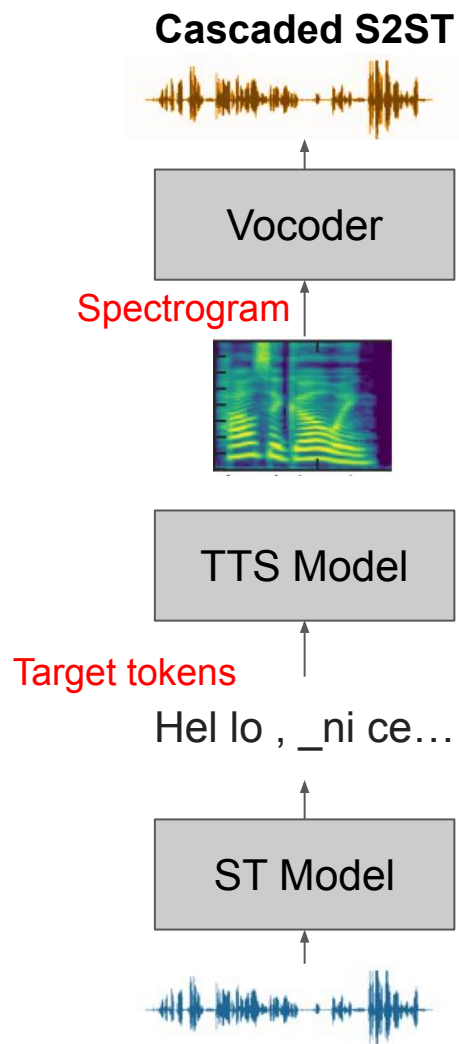
The IWSLT 2024 Evaluation Campaign will host shared tasks featuring the following focus areas:

- **Speech translation campaign tracks:**
  - [Speech-to-speech track](#) (Qianqian Dong, Bytedance, China)
  - [Simultaneous track](#) (Katsuhito Sudoh, NAIST, Japan)
  - [Subtitling track](#) (Mauro Cettolo, FBK, Italy; Evgeny Matusov, AppTek, Germany)
  - [Offline track](#) (Marco Turchi, Zoom, Germany; Matteo Negri, FBK, Italy)
  - [Dubbing track](#) (Brian Thompson, Amazon, USA; Prashant Mathur, AWS AI Labs, USA)
  - [Low-resource track](#) (Antonios Anastasopoulos, George Mason University)
  - [Indic track](#) (Chandresh Kumar Maurya, IIT Indore, India)

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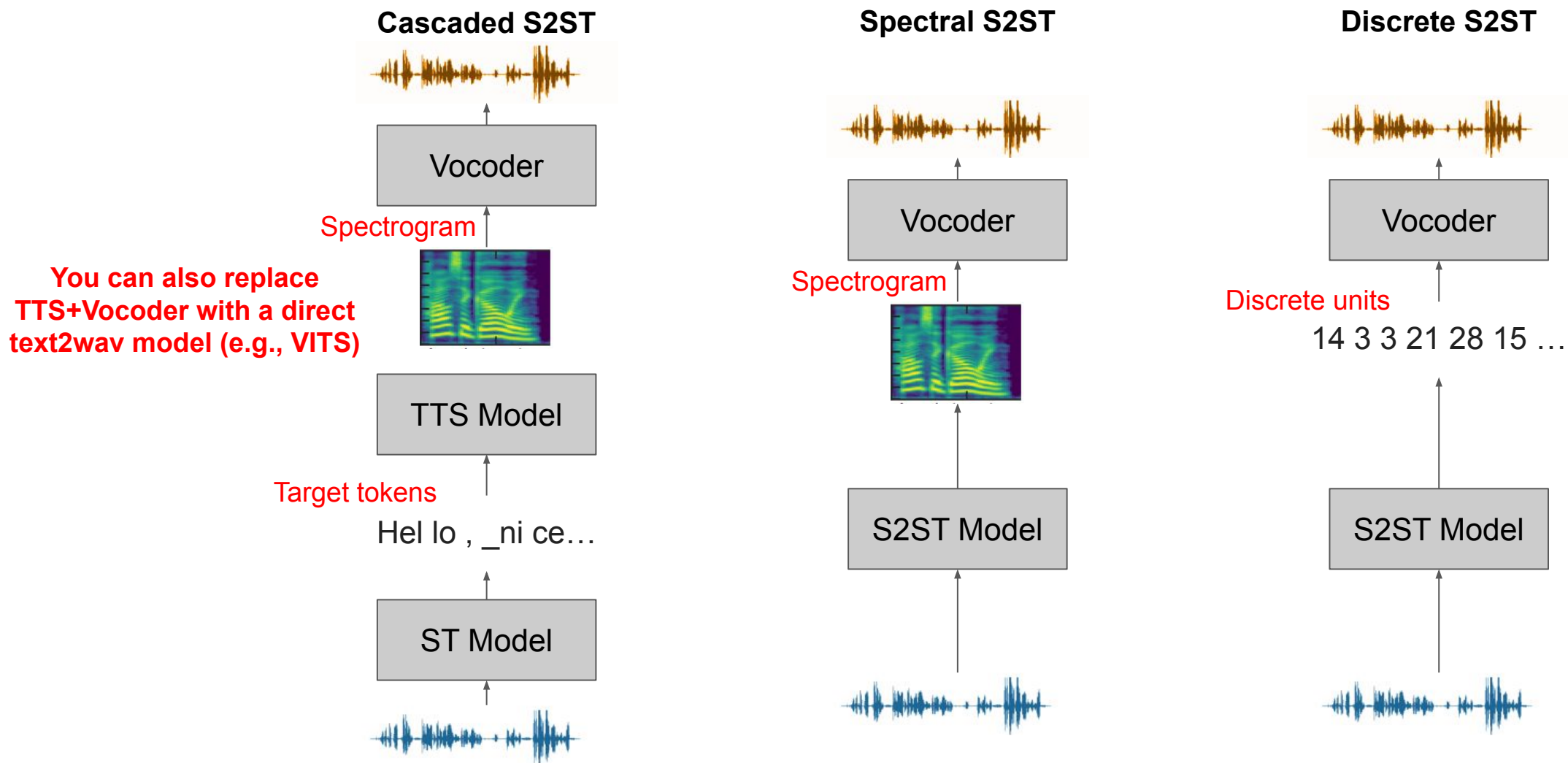
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# Cascaded S2ST vs. Spectral S2ST vs. Discrete S2ST

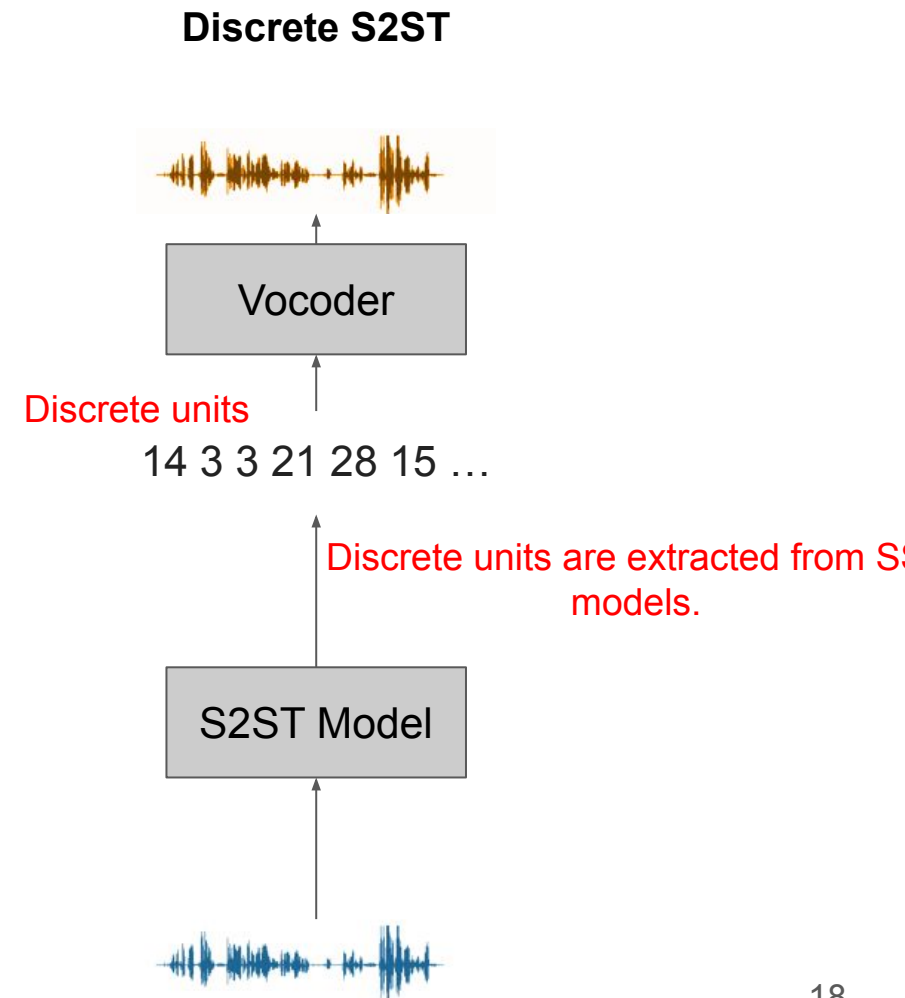
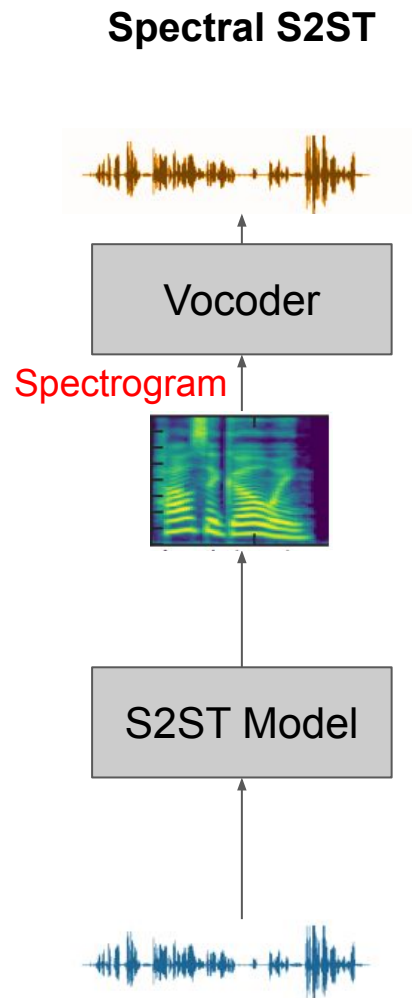
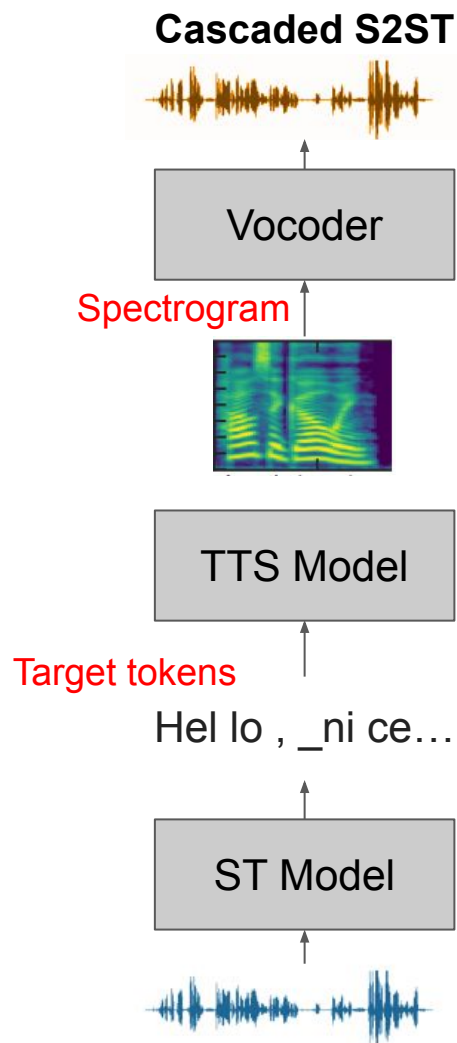




# Cascaded S2ST vs. Spectral S2ST vs. Discrete S2ST

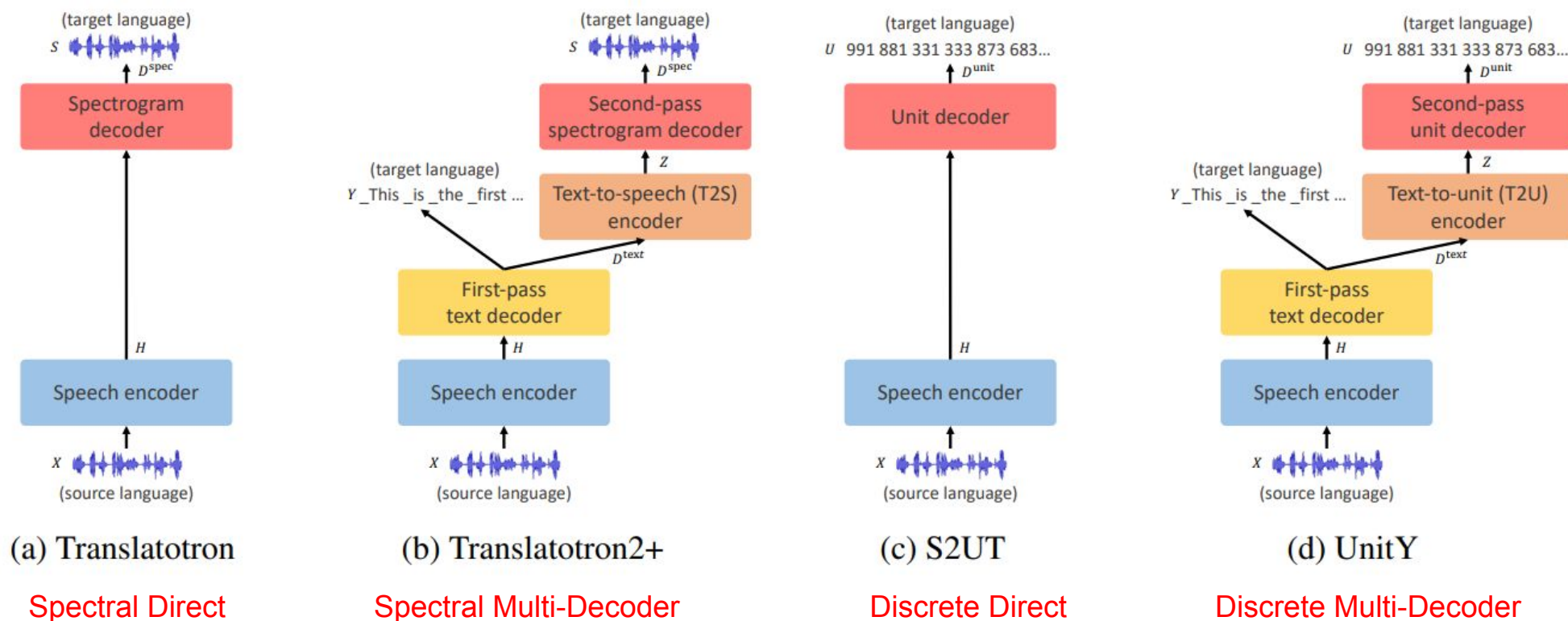


# Cascaded S2ST vs. Spectral S2ST vs. Discrete S2ST



# Multi-Decoder Speech-to-Speech Translation

- Use Multi-Decoders to build S2ST models with **speech-to-text intermediates**



# Spectral vs. Discrete unit

- Spectral

- Pros

- Standardized feature (easy to build common interface)
    - Expressiveness
    - Flexibility
    - Techniques in TTS pre-training

- Cons

- Training takes longer time
    - Difficult to converge
    - No good training monitor



- Discrete unit

- Pros

- Training is faster
    - Convergence is fast and stable
    - Accuracy is a good indicator of performance
    - Techniques in text pretraining

- Cons

- Dependency in SSL
    - Bad generalization to language, data sources, speakers, prosody
    - Feature is not standardized (difficult to share/reuse)

# Typical steps of building cascaded S2ST

- Step1: Data Preparation
  - Data
    - Source language speech
    - Source language text
    - Target language text (not necessary but strongly recommend)
    - Target language speech



# Typical steps of building cascaded S2ST

- Step2: Model training
  - End-to-end ST or ASR+MT
    - Input: Source language speech
    - Intermediate output: source language text
    - Output: target language text
  - TTS model
    - Input: target language text
    - Output: target language speech (spectral features)
  - Vocoder
    - Input: target language speech (spectral features)
    - Output: target language speech (waveform)



# Typical steps of building cascaded S2ST

- Step3: Inference
  - First use the ST model to translate source language speech
  - Then use TTS model to convert source language speech into spectral features
  - Then use vocoder to convert spectral features into waveform.



# Typical steps of building discrete direct S2ST

- Step1: Data Preparation

- Data

- Source language speech
    - Source language text (not necessary but strongly recommend)
    - Target language text (not necessary but strongly recommend)
    - Target language speech

- Pre-trained model

- Speech SSL model pre-trained on target language speech
      - (can be obtained by either pre-trained from scratch or fine-tune from existing SSLs in other languages)





# Typical steps of building discrete direct S2ST

- Step 2: Model Preparation
  - Unit extraction model
    - Extract SSL feature (usually just a certain layer) from target speech
    - Conduct Kmeans clustering over the extracted features (unit size can be 100 -> 1000)
    - The Kmeans model is the unit extract model



# Typical steps of building discrete direct S2ST

- Step 2: Model Preparation

- Unit extraction model

- Extract SSL feature (usually just a certain layer) from target speech
    - Conduct Kmeans clustering over the extracted features (unit size can be 100 -> 1000)
    - The Kmeans model is the unit extract model

- Unit vocoder

- Using extracted unit sequences to train a unit-based vocoder
    - Responsible for converting the unit sequence to final waveform



# Typical steps of building discrete direct S2ST

- Step 2: Model Preparation

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- Extract SSL feature (usually just a certain layer) from target speech
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- Using extracted unit sequences to train a unit-based vocoder
    - Responsible for converting the unit sequence to final waveform



What is tradeoff of different unit size?

# Typical steps of building discrete direct S2ST

- Step 3: Model Training
  - Train an end-to-end model
    - Input: source speech
    - Output: target speech in units
- Step 4: Inference
  - First use end-to-end model to predict unit sequence
  - Then use vocoder to convert it into the final waveform
- Step 5: Evaluation
  - Use an existing ASR model to generate ASR transcripts
  - Use ASR transcripts and reference target text to compute ASR-BLEU



# More extensions

- Speech-to-speech translation has become one of the major tasks in recent speech foundation models, due to **its compositional feature**

- AudioPalm

<https://arxiv.org/pdf/2306.12925.pdf>

**Types of tasks** We apply our method to the problems of speech recognition, speech synthesis and speech-to-speech translation. All datasets used in this report are speech-text datasets which contain a subset of the following fields.

- Audio: speech in the source language.
- Transcript: a transcript of the speech in Audio.
- Translated audio: the spoken translation of the speech in Audio.
- Translated transcript: the written translation of the speech in Audio.

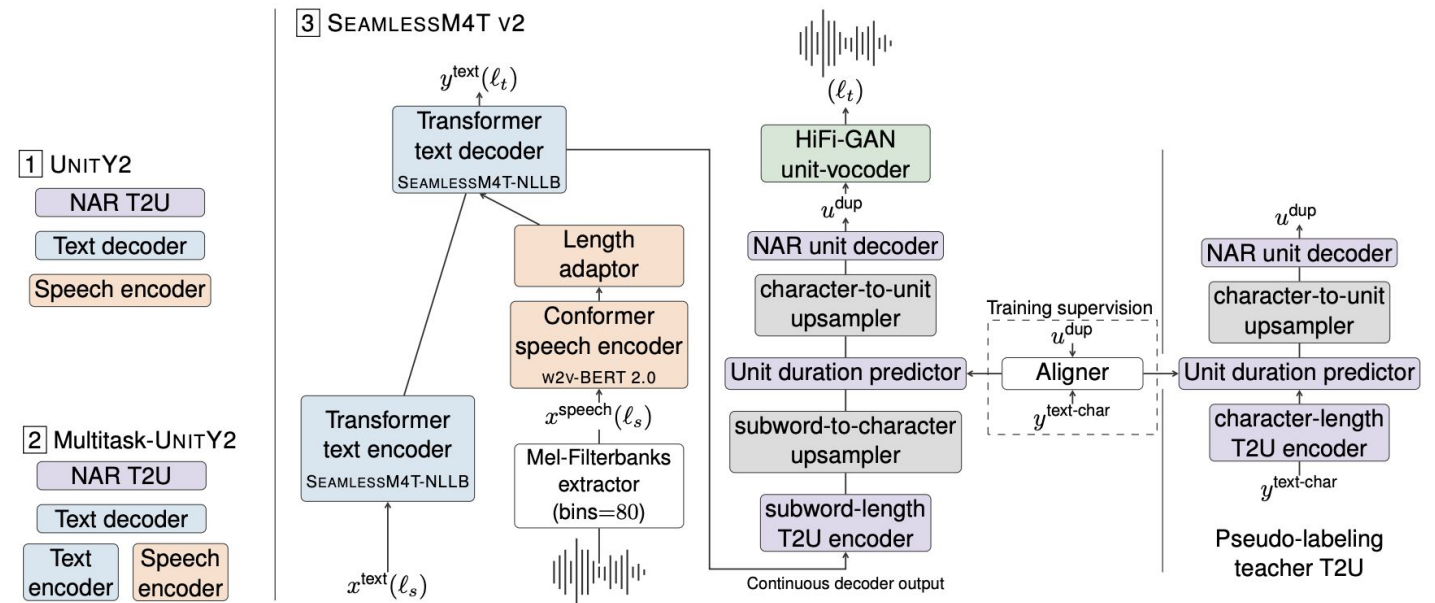
The component tasks that we consider in this report are:

- ASR (automatic speech recognition): transcribing the audio to obtain the transcript.
- AST (automatic speech translation): translating the audio to obtain the translated transcript.
- S2ST (speech-to-speech translation): translating the audio to obtain the translated audio.
- TTS (text-to-speech): reading out the transcription to obtain the audio.
- MT (text-to-text machine translation): translating the transcript to obtain the translated transcript.

# More extensions

- Speech-to-speech translation has become one of the major tasks in recent speech foundation models, due to **its compositional feature**
- Seamless

<https://ai.facebook.com/research/publications/seamless-multilingual-expressive-and-streaming-speech-translation/>



# Summary

- Speech-to-speech translation
  - Recap of basic speech-to-speech translation information
  - Technicals
    - Discrete and spectral approaches
    - Detailed steps of building a speech-to-speech translation system

# References

- Papineni, Kishore, et al. "Bleu: a method for automatic evaluation of machine translation." Proceedings of the 40th annual meeting of the Association for Computational Linguistics. 2002.
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