

Optimizing Rare Word Accuracy in Direct Speech Translation with a Retrieval-and-Demonstration Approach

Anonymous ACL submission

Abstract

Direct speech translation (ST) models often struggle with rare words. Incorrect translation of these words can have severe consequences, impacting translation quality and user trust. While rare word translation is inherently challenging for neural models due to sparse learning signals, real-world scenarios often allow access to translations of past recordings on similar topics. To leverage these valuable resources, we propose a *retrieval-and-demonstration* approach to enhance rare word translation accuracy in direct ST models. First, we adapt existing ST models to incorporate retrieved examples for rare word translation, which allows the model to benefit from prepended examples, similar to in-context learning. We then develop a cross-modal (speech-to-speech, speech-to-text, text-to-text) retriever to locate suitable examples. We demonstrate that standard ST models can be effectively adapted to leverage examples for rare word translation, improving rare word translation accuracy over the baseline by 17.6% with gold examples and 8.5% with retrieved examples. Moreover, our speech-to-speech retrieval approach outperforms other modalities and exhibits higher robustness to unseen speakers. Our code is in the submission.

1 Introduction

Speech translation (ST) traditionally involves cascading automatic speech recognition (ASR) and machine translation (MT) (Stentiford and Steer, 1988; Waibel et al., 1991) to convert spoken language into text in a different language. However, recent years have witnessed rapid progress in direct ST models (Anastasopoulos et al., 2021, 2022; Agarwal et al., 2023) that bypass intermediate text representations for lower inference latency and reduced error propagation (Sperber and Paulik, 2020). Despite the advancements, accurately translating rare words like person names (Gaido et al., 2021, 2023) remains a significant challenge for ST systems. While infrequent, incorrect translations of

rare words can severely degrade overall translation quality and even users’ trust in the deployed models. Rare word translation is inherently difficult for ST models due to limited or absent learning signals. Practically, however, valuable external resources hold the potential to address this issue. Real-world scenarios often allow access to translations from past recordings on similar topics, sometimes even from the same speaker. Similarly, human translators often leverage existing translations (Bowker, 2005), especially for special terminologies (Brkić et al., 2009). Inspired by these observations, we ask the question: How can we improve the rare word translation performance of direct ST models by leveraging an example pool that contains similar translations?

The envisioned approach faces challenges in both the *retrieval* and *translation* components. First, the retrieval task is complicated by the variability of speech and the locality of rare words. As the speaking condition for the same rare word differs in every utterance, source-side feature matching as often done in text translation (Zhang et al., 2018; Bulte and Tezcan, 2019; Xu et al., 2020; Cai et al., 2021; Hao et al., 2023) is not sufficient to handle the pronunciation variations. Moreover, as rare words only constitute a small portion of the query and candidate utterances, the retriever must be able to locate the relevant information in long speech utterances. For the translation model, integrating retrieved utterance-translation pairs is also non-trivial. Standard models trained on sentence-level data require adaptation to ingest the examples. Besides processing longer inputs, they also need to pinpoint both the acoustic features and corresponding textual translations of rare words.

Addressing the above challenges, we introduce a retrieval-and-demonstration framework (Figure 1) effective for improving rare word translation accuracy of ST models. Specifically, we adapt standard ST models to benefit from prepended examples in

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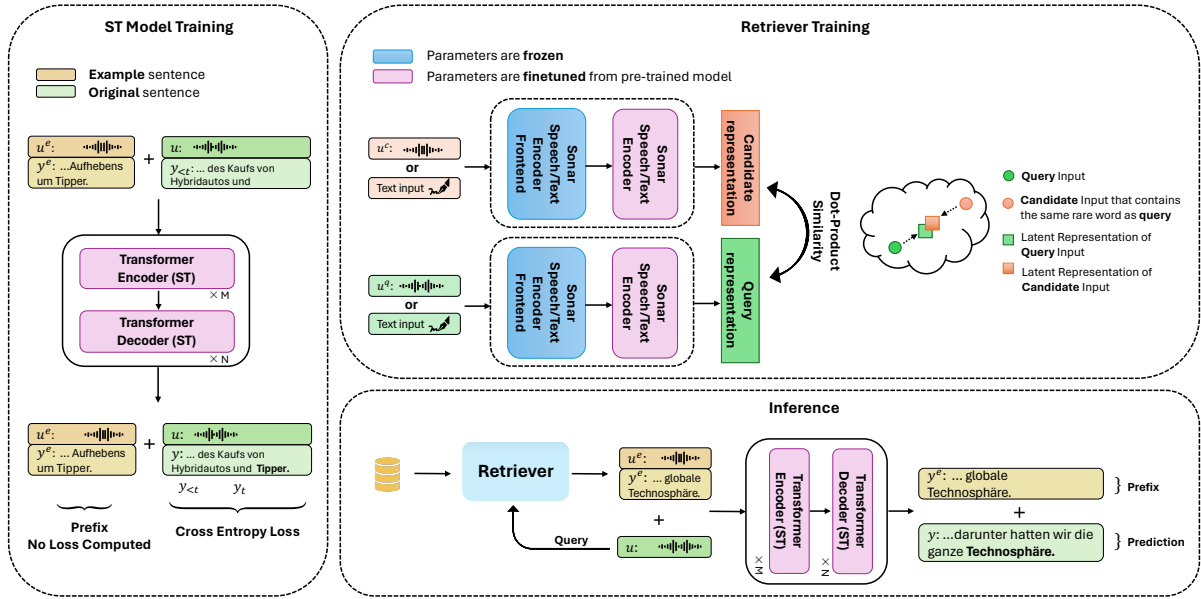


Figure 1: Proposed retrieval-and-demonstration framework: At the ST model training stage (§2.1), example-prepended training data is used to instill in-context learning abilities in the S2T model. At the retriever training stage (§2.2), SONAR encoders are fine-tuned within the DPR architecture for our rare word task. At the inference stage (§2.3), retrieved examples are used as demonstrations to facilitate the translation of rare words.

a way similar to in-context learning (Brown et al., 2020), and then build a retriever to find suitable examples. Building on recent multi-modal encoders (Duquenne et al., 2023), the retriever supports multiple modalities (speech→speech, speech→text, text→text). Second, we propose an evaluation methodology to adapt standard ST corpora, MuST-C (Di Gangi et al., 2019) in this case, for targeted assessment of rare words translation (§3.1). Our main findings are:

- Standard direct ST models can be easily adapted to benefit from prepended examples for rare word translation, in a way similar to in-context learning (§4.1). This improves rare word translation accuracy over the baseline by 17.6% with gold examples and 8.5% with retrieved examples.
- Text-to-text information retrieval architectures (Karpukhin et al., 2020) can be effectively adapted for speech-based rare word retrieval, yielding 33.3% to 46.6% top-1 retrieval accuracy under different modalities (§4.2).
- Compared to other modalities, speech-to-speech retrieval leads to higher overall translation quality and rare word translation accuracy (§4.3), as well as more robustness to unseen speakers (§5.1).

2 Proposed Framework

Our retrieval-and-demonstration framework is illustrated in Figure 1. First, a trained direct ST model

is finetuned to ingest examples (§2.1), which serve as demonstrations of correctly translating the rare words in question. During inference, given an utterance containing rare words, we retrieve (§2.2) a relevant utterance and its translation as a demonstration to guide the inference (§2.3).

2.1 Adapting ST Models to Ingest Examples

Motivation Human translators often leverage example translations also known as *translation memory* (Bowker, 2005), especially for domain-specific translation with terminologies (Brkić et al., 2009). We aim to apply a similar approach to direct ST models. The underlying idea mirrors that of in-context learning (ICL) (Brown et al., 2020), where providing models with task-specific examples during inference improves the quality of the generated output. While ICL has been primarily observed on text-based LLMs (Brown et al., 2020; Min et al., 2022; Vilar et al., 2023), we explore whether small- or medium-sized encoder-decoder-based speech translation models can also exhibit this capability.

Training To adapt standard ST models to ingest examples, the example utterance and translation must be included as context for training and inference. An intuitive approach is to include the example as prefix in both input and output, as shown in the left side of Figure 1, This allows the output generation to be conditioned on the exam-

ple utterance and translation as context. Formally, given an utterance u , let \hat{y} be the target translation and y the predicted translation. Let (u^e, y^e) be an example utterance-translation pair. We aim to adapt an ST model so that the model maximizes the probability of generating the correct translation \hat{y} , given the input utterance u and example $(u^e, y^e) : y = \arg \max_{\hat{y}} P(\hat{y}|u^e, y^e, u)$. The difference to the standard training is that the example (u^e, y^e) is included as context when generating the target translation. For the training data, for the i -th training utterance u_i , an example utterance u_i^e is prepended to it, forming a concatenated input $u_i^e + u_i$.¹ The targets are also concatenated as $y_i^e + \langle \text{SEP} \rangle + y_i$, where $\langle \text{SEP} \rangle$ is a special token indicating the separator between sentences. During training, the loss is only calculated on y_i to prioritize the translation of the utterance after the example.² In doing so, we encourage the model to predict its outputs based on the context provided by the demonstration example.

2.2 Example Retrieval

Formalization and Challenge Given a query utterance u containing a rare word w , we aim to retrieve a relevant example (u^e, y^e) from an example pool $\mathcal{D} = \{(u^1, y^1), \dots, (u^m, y^m)\}$ with a retrieval model r , such that the rare word w is spoken in utterance u^e . Here u^i indicates the i -th utterance and y^i its translation. As the query u is only in speech, we face additional complexities compared to text-based retrieval. *First*, speech is versatile, unlike text, which often has a standard writing system. The speaking condition for the same word varies in every recording, requiring a robust retriever that accounts for pronunciation variations. *Second*, speech sequences are magnitudes longer than text. The retriever must find fine-grained local features corresponding to the keywords in long sequences. *Third*, transcribing the query utterance first and then using text-based retrieval is suboptimal due to ASR errors, especially on rare words.

Architecture As the nature of our example retrieval task resembles information retrieval (IR) where relevant answers are retrieved given a question, we take inspiration from IR approaches for our retriever. In *text-to-text* IR, a prominent architecture is the Dense Passage Retriever (DPR)

(Karpukhin et al., 2020). It has a *dual-encoder* architecture, where one encoder encodes the questions, and the other encodes the passages potentially containing answers to the questions. The retrieval model is trained with a contrastive objective, mapping question-passage (positive) pairs closer to each other in the latent space while pushing irrelevant (negative) pairs further apart. During inference, passages closer to the encoded question by the dot-product similarity are returned as answers. In our case, the utterances containing the same rare words are considered positive pairs, while those not sharing the same rare words are negative pairs. **Speech-to-Speech/Text Retrieval** We propose to extend the DPR model to support querying from speech. As the example utterances to be retrieved often also have text transcripts available, we consider the following retrieval modalities:

- **Speech→speech retrieval:** we retrieve u^e in speech using audio query u .
- **Speech→text retrieval:** we retrieve y^e directly using audio query u . This requires the retriever to support both modalities (text and speech).
- **Naïve text→text retrieval:** first transcribing the query utterance u and then text-to-text retrieval for y^e . As discussed before, the risk of ASR errors especially on rare words renders this approach suboptimal. The additional inference time for running ASR makes it further unpractical.

Given these requirements, instead of initializing the dual encoders with pre-trained BERT (Devlin et al., 2019) as in DPR (Karpukhin et al., 2020), we leverage recent speech-text joint representation models including SONAR (Duquenne et al., 2023) and SpeechT5 (Ao et al., 2022).

2.3 Integrating Examples into ST Model

Inference with Retrieved Examples During inference, the model is provided with a test input u and a retrieved example (u^e, y^e) . The example is prepended to test input in the same way as in training. The example input-output pairs are integrated by forced decoding. After the separator token ($\langle \text{SEP} \rangle$), the model starts to autoregressively generate the output translation, conditioned additionally by the example utterance and translations.

Practical Considerations An advantage of our framework is its modularity. The separation of the ST and retrieval modules enables straightforward upgrades to newer models in either component. Moreover, the retrieval module can be implemented using highly optimized toolkits like FAISS (John-

¹Details on constructing the dataset is in §3.1.

²Including the loss on the prefix leads the finetuning step to end prematurely in preliminary experiments. The loss calculation is formally described in Appendix A.

son et al., 2021), which ensures efficient retrieval without compromising inference speed.

Split	# utt.	Avg. utt. duration (s)	Avg. # tokens	# unique rare words
train (original)	250942	6.5	27.1	9512
tst-COMMON	2580	5.8	25.3	157
rare-word pool	9821	9.7	43.1	8679
dev-rare-word	6932	9.9	42.8	6244
tst-rare-word	2500	9.9	43.1	2358
train-reduced	231689	6.2	25.8	3164

Table 1: Dataset statistics. We split the original training set into the example pool with rare words (rare-word pool), dev/test sets for rare words (dev/tst-rare-word), and a reduced training set (train-reduced). The example pool simulates existing resources for querying.

3 Experimental Setup

3.1 Dataset Construction

For evaluation, we use the English-to-German subset of the MuST-C dataset (Di Gangi et al., 2019), where the task is to translate from English-public speaking audio to German text. To create a targeted test condition for rare words, we extract sentences containing rare words from the original training set to create dedicated sets. The statistics of the original dataset and the newly created splits are in Table 1. The rare-word sets have higher average token counts due to: 1) longer utterance duration and 2) the rare words being segmented into finer-grained subwords. Note that we only re-split the training set, leaving the official validation and test sets (tst-COMMON) unmodified. Below we describe the dataset construction process in detail.

Rare Word Sets Our data partition step is inspired by Niehues (2021), which re-splits parallel data based on word frequencies. Specifically, from the English transcript, we find rare words by their *corpus-level frequency*, choosing those appearing two or three times in the original training set. For rare words occurring twice, we move their corresponding utterances to the rare-word pool and the joint dev/tst set respectively, which creates a *zero-shot* condition where the rare word is never seen in training. For rare words occurring thrice, we follow the same strategy for two occurrences. The remaining third occurrence is retained in the reduced training set to create a *one-shot* learning scenario, where the rare word is seen once in the training set. Finally, the aggregated dev/tst set is split into individual development and test sets for standard

evaluation. We analyze the rare word types in tst-rare-word by a named entity recognition (NER) model³ with results in Table 2. A more detailed categorization of the words is in Appendix B.

tst-rare-word	Person	Location	Tech	Food	Company
2358	130	72	29	27	25

Table 2: NER results on rare words in tst-rare-word with the number of unique words in each category.

Training Data with Prepended Examples To adapt the ST model and to train the retriever, we need training data with prepended examples. As most utterances lack rare words by the previously used corpus-level frequency (3164 rare words in 231k utterances in Table 1), we propose to use *sentence-level* rare words to choose the prepended examples. Specifically, for each piece of the training data (u^i, s^i, y^i) , we identify the word w_s in s^i that has the least corpus-level frequency among all words in its transcript. We then sample another training instance (u^j, s^j, y^j) where s^j contains the same sentence-level rare word w_s as example.

Test Set with Gold Examples We also construct a variant of tst-rare-word set with gold examples, where the rare word in the test utterance is always present in the example. This serves as an oracle condition for evaluating the ST model’s ability to learn from perfect demonstrations. As our data splitting procedure ensures that the rare words also occur in the example pool, we select sentences from the rare-word pool containing the same rare words as those in the tst-rare-word set to serve as example sentences. The example sentences are then prepended to test sentences in a way identical to that in the training set with prepended examples.

3.2 Model Configuration

ST Model We use the Transformer architecture S2T_TRANSFORMER_S in FAIRSEQ S2T (Wang et al., 2020) for all our ST models. To prevent the tokenizer from seeing the rare words during its training, which will cause an unfair test condition, we train the SentencePiece (Kudo and Richardson, 2018) tokenizer on the reduced train set after the utterances containing rare words are moved to dedicated splits (Table 1). Based on this vocabulary, we train the base model on the train-reduced set, closely following the hyperparameters from Wang et al. (2020). We then adapt the base model to

³Huggingface model by Zaratiana et al. (2023)

ingest examples as described in §2.1 using the reduced training set with prepended examples (§3.1). As the prefix tokens do not contribute to the overall loss (Figure 1), we double the effective batch size to keep the loss scale comparable to before. Further details on training and inference are in Appendix C.

Retriever We use the DPR (Karpukhin et al., 2020) architecture for the retriever. The encoders are initialized with either SONAR (Duquenne et al., 2023) or SpeechT5 (Ao et al., 2022). For both models, we use the encoder only and discard the decoder. DPR requires fixed-size embeddings from its encoders. For SpeechT5, we mean-pool over the sequence length. For SONAR, we use the built-in attention-pooling for the speech encoder and mean-pooling for the text encoder. The dual encoders in DPR are trained on the reduced training set with prepended examples. Each sentence’s example serves as a positive example, while examples from other sentences in the batch are in-batch negatives. Only the top layer of the encoders is trained, as the lower layers of the encoders are likely responsible for extracting low-level acoustic features. These features are considered less relevant for our retrieval task, which focuses on word-level information. Another reason is memory efficiency in training. Further details on training and inference are in Appendix D.

3.3 Evaluation

We evaluate speech translation quality with BLEU (Papineni et al., 2002)⁴ and COMET (Rei et al., 2020)⁵. For the accuracy of rare word translation, we evaluate how many unique lemmatized rare words in the test set are translated. We use the spaCy toolkit (Honnibal et al., 2020) for word lemmatization and used AWESoME Aligner (Dou and Neubig, 2021) for en-de word-level alignment. For rare word accuracy, we further distinguish between rare words appearing once or never appear in the training set (§3.1), which corresponds to the *one-shot* and *zero-shot* accuracy. For the retriever, we use top-1 retrieval accuracy to evaluate the retriever’s performance. Only the top retrieved examples are used as demonstrations in the ST model.

4 Main Results

Before presenting the results of our proposed framework, we confirm that our baseline model performs

⁴sacreBLEU (Post, 2018) signature:
nrefs:1lcase:mixedleff:noltok:13alsmooth:explversion:2.4.2

⁵with Unbabel/wmt22-comet-da; ×100 for readability.
The COMET models take text transcripts as source.

on par with those reported in the literature. The details are in Appendix E.

4.1 Impact of Demonstration

Direct ST models can effectively learn from demonstration at inference time. To independently analyze the ST model’s ability to learn from the prepended examples, we first assume an oracle retrieval model by using gold examples which always contain the rare words in question. The results are in row (2) of Table 3. Compared to the baseline in row (1), this model achieves substantially higher overall rare word translation accuracy (+17.6% abs.), with a larger gain in zero-shot (+18.8%) than one-shot accuracy (+15.3%). Nonetheless, this gain comes at the cost of overall translation quality (−0.2 BLEU, −2.3 COMET). A potential reason is that the prepended example sentences make the input sequences much longer and therefore create more difficulty for learning. Nonetheless, since rare words are often important named entities, capturing them correctly is as crucial if not more than the overall translation quality scores. Overall, the results suggest that task-specific demonstrations provided at inference time can effectively enhance rare word translation accuracy of direct ST models.

Quality of the given demonstration matters.

Next, we study the impact of the demonstration quality. In contrast to the gold examples before, we now use random examples that do not contain rare words relevant to the sentence to be translated. The results are in row (3) of Table 3. This led to a decline in translation quality (−1.3 BLEU, −2.4 COMET) and rare word accuracy. These results indicate that irrelevant demonstrations are harmful.

Seeing rare words only in training does not sufficiently improve their translation accuracy.

Instead of retrieving data from the rare-word pool as demonstration, a simple alternative is to add these data in training. Here, we add the rare-word pool into the training set and train an identical model to the baseline. The results are in row (4) of Table 3. Overall, the rare word accuracy only sees a slight increase compared to row (1), with an absolute accuracy improvement of 3.7%, which is far less than using gold example sentences (+17.6% overall). This indicates that training with rare words alone is insufficient for improving their translation accuracy. This is likely because of the limited training signal for rare words, as each appears only once or twice. Note that the translation quality scores

ST Model	BLEU	COMET	Overall acc (%)	0-shot acc (%)	1-shot acc (%)
(1) baseline model (on train-reduced)	17.2	57.9	11.8	11.0	13.3
(2) adapted + gold example	17.0	55.6	29.4	29.8	28.6
(3) adapted + random example	15.7	53.2	8.8	8.4	9.7
(4) train on {train-reduced + rare-word pool} (more data)	17.9	59.0	15.5	14.7	17.2
Using retrieved examples					
(5) adapted + text (gold transcript)→text	15.2	54.4	20.1	19.6	21.2
(6) adapted + speech→text	15.3	54.0	18.8	18.2	20.2
(7) adapted + speech→speech	16.2	55.3	20.3	20.3	20.2

Table 3: Translation quality (BLEU \uparrow , COMET \uparrow) and rare word accuracy \uparrow (overall, 0- and 1-shot) of different models on the tst-rare-word split. The lower section uses retrieved examples from the retriever (§4.3).

Retrieval Model	T→T	S→T	S→S
(1) Orig. DPR w/ BERT (pretrained)	2.0	–	–
(2) Orig. DPR w/ BERT (finetuned)	55.8	–	–
(3) DPR w/ SpeechT5 (finetuned)	0.1	0.0	0.0
(4) DPR w/ SONAR (pretrained)	28.7	22.3	20.6
(5) DPR w/ SONAR (finetuned)	46.6	33.3	41.3

Table 4: Top-1 retrieval accuracy (%) of different retrievers on 3 modalities of text-to-text (T→T), speech-to-text (S→T), and speech-to-speech (S→S) on the tst-rare-word split. T→T retrieval uses gold transcripts as query.

under this data condition also improved, which is likely a result of the additional training data.

4.2 Retrieval Performance

Before integrating retrieved examples into the ST model, we analyze the retrieval performance alone with results in Table 4. To establish the upper bounds of retrieval performance, we first use the original DPR model for text-to-text retrieval with gold transcripts of the query utterances and examples. As shown in row (1) of Table 4, directly using the pretrained DPR for QA is not sufficient for our task of rare word retrieval. Fine-tuning DPR’s encoders (row (2)) on our task enables effective rare word retrieval in a text-to-text setting (55.8%).

Encoder choice is crucial for successful retrieval.

We proceed by adapting the original DPR to retrieval from speech. Overall, we notice that the choice of the encoder heavily impacts the retrieval performance. With SONAR, using the pretrained encoders already achieves partial success in fulfilling the task (row (4) in Table 4), with finetuning further improving the results (row (5)). However, finetuning SpeechT5 proves insufficient for learning the task (row (3)). We believe that the discrepancy primarily arises from the models’ ability to aggregate information over the sentence length:

SONAR is explicitly trained to aggregate it into fixed-size embeddings while SpeechT5 lacks such a mechanism. Naïve mean-pooling over sequence length fails to create meaningful embeddings over long sequences like speech, as well as character-level text representations used in SpeechT5.

Speech→speech outperforms speech→text retrieval.

While we initially expected speech-to-speech retrieval to be more challenging than speech-to-text retrieval due to the high variability of speech, the finetuned retriever in (5) of Table 4 shows stronger performance on speech→speech retrieval than speech→text (41.3% vs. 33.3%). We suppose that the reason is the modality gap between text and speech, which makes it more challenging to bridge the two different types of data.

4.3 ST Performance with Retrieved Examples

Correlation between retrieval accuracy and translation quality:

As the retriever based on finetuned SONAR showed the most promising retrieval results (Table 4), we use the retrieved examples from this model to guide the ST. The results are in rows (5), (6), and (7) of Table 3. When comparing the performance of the three retrieval modalities, retrieval accuracy does not always translate to improved overall translation quality or rare word accuracy. Although text-to-text retrieval using gold transcripts had the highest retrieval accuracy (Table 4), its integration into the ST model resulted in lower translation quality compared to speech-to-speech retrieval. Moreover, in practice, we still need an ASR model to derive the transcripts that likely contain errors, especially on rare words. This introduces additional limitations to the text-to-text retrieval approach. Overall, these results show that speech-speech retrieval is more effective than the other modalities in improving rare word translation

accuracy. Despite the improvement in rare word translation accuracy, we also note the drop in translation quality compared to the baseline (row (7) vs. (1); -1.0 BLEU and -2.6 COMET). We expect that increasing the robustness of the ST model to examples containing incorrect rare words, for instance by including such examples in training, could mitigate this negative impact.

Does speech→speech retrieval help by implicit speaker adaptation? Speech-to-speech retrieval could be particularly effective in finding same-speaker utterances due to the access to acoustic information. This raises the hypothesis that if the prepended example originates from the same speaker as the utterance to be translated, translation quality could be improved by implicit speaker adaptation (Saon et al., 2013), where the model benefits from adapting to the specific speaker’s voice characteristics. To test this, we analyze the proportion of retrieved sentences from the same speaker across different retrieval modalities. The results in Table 5 show similar percentages for all three scenarios, indicating that the gains by speech-to-speech retrieval do not stem from speaker adaptation.

DRP + SONAR finetuned	T→T	S→T	S→S
Examples from same speaker (%)	50.3	53.4	50.2

Table 5: Proportion of retrieved examples from the same speaker as the utterance to be translated for the three retrieval modalities on tst-rare-word.

5 Further Analyses and Discussions

5.1 Effects on Unseen Speakers

Now we push the approach further under the challenging scenario of unseen speakers, i.e., the example pool does not contain any utterance from the speaker of the test utterance. Specifically, during retrieval, we ignore utterances from the same speaker as the query utterance. As shown in Table 6, this harms retrieval accuracy substantially, losing 14.9% to 23.4% compared to Table 4 for the three modalities. This is mainly due to the limited coverage of the rare-word pool, which contains only one sentence for most rare words. Excluding the speaker also excludes the rare word. However, the BLEU scores and overall rare word translation accuracy change only slightly compared to Table 3: T→T (-0.6 BLEU, -1.5%), S→T (-0.3 BLEU, -3.2%), S→S ($+0.2$ BLEU, -1.0%). This demon-

strates that our approach, especially when using speech→speech retrieval, is relatively robust to unseen speakers.

Retrieval modality	Retrieval acc (%)	BLEU	Overall acc (%)	0-shot acc (%)	1-shot acc (%)
(5) T→T	23.2	14.6	18.6	18.5	18.7
(6) S→T	18.4	15.0	15.6	15.6	15.7
(7) S→S	23.5	16.4	19.3	18.8	20.2

Table 6: Retrieval and ST performance on unseen speakers. Compared to Table 3, S→S retrieval has the least decrease in translation quality and rare word accuracy.

5.2 Qualitative Example

Table 7 shows an example where our approach creates partially correct translation for the named entities “Patrice and Patee”. To avoid cherry-picked results, we include more examples where our approach succeeds and fails in Appendix F.

Source (transcript): Patrice and Patee set out most days to go out hunting in the forest around their homes.
Baseline (Table 3 row (1)): Die Bäume und Petes (Trees and Petes) setzten die meisten Tage hinaus, um in den Wäldern um ihre Häuser zu pumpen.
Adding rare-word pool to training (Table 3 row (4)): Patrizinpathie (Patrizinpathie) setzte sich in den meisten Tagen um die Jagd in den Wäldern um ihre Häuser.
Speech→speech example (Table 4 row (5)): Sie heißen Patrice und Patee (Their names are Patrice and Patee.).
Adapted ST + speech→speech (Table 3 row (7)): Patrice und Pateetee setzten die meisten Tage, um in den Wäldern um ihre Häuser herum jagen zu können.
Target : Patrice und Patee (Patrice and Patee) gehen fast jeden Tag jagen in dem Wald rundum ihr Heim.

Table 7: An example of our retrieval-and-demonstration approach improving the translation of rare words.

5.3 Analyses of Retrieval Performance

In our main experiments, we partially finetuned the DPR encoders. We now investigate the impact of different numbers of trainable parameters in the retriever. As shown in Figure 2, the retrieval performance of the SONAR-based retriever is stable across 100 to 500M trainable parameters out of a total of over 1.3B parameters. This indicates that the retriever can maintain nearly consistent performance despite changes in model capacity.

5.4 Potential of Using More Examples

Few-shot learning is more often performant than one-shot learning because it provides the model with a broader context and more varied examples.

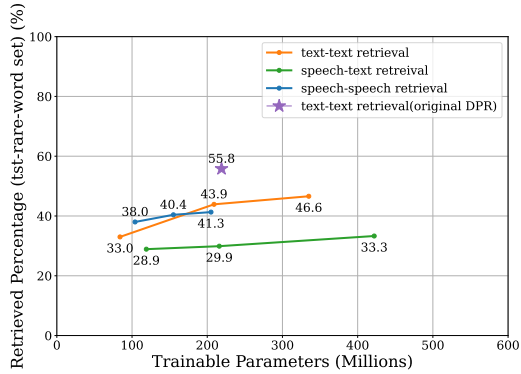


Figure 2: Retrieval performance of the SONAR-based retriever for different numbers of trainable parameters.

However, as shown in Table 8, the increase in retrieval accuracy with additional top-10 examples is still not substantial compared to the top-1 result. Including multiple examples also makes input sequences significantly longer, especially as audio inputs are factors longer than text. This not only poses a challenge for the model but would also significantly slow down the inference speed, which we aim to avoid. For these reasons, we do not further explore the potential of using more examples.

DPR + SONAR ft.	T→T	S→T	S→S
Top 1	46.6	33.3	41.3
Top 5	60.4	48.0	56.2
Top 10	64.6	53.1	61.1

Table 8: Top-10 retrieval performance (%) of the SONAR-based retriever on the tst-rare-word set.

6 Related Work

Retrieval-Augmented Translation Our work falls within the paradigm of retrieval-augmented translation (RAT) (Simard and Langlais, 2001; Koehn and Senellart, 2010; Tu et al., 2018; Khandelwal et al., 2021), which augments a translation model with results retrieved from a translation memory. Prior works on RAT primarily focus on text-to-text translation (Zhang et al., 2018; Gu et al., 2018; Bulte and Tezcan, 2019; Xu et al., 2020; Cai et al., 2021; Hoang et al., 2023; Hao et al., 2023), where retrieval relies on textual feature matching such as n -gram overlap. These methods are therefore not readily applicable to direct ST due to the continuous nature of speech and much longer input lengths. In ST, Du et al. (2022) use k NN-MT (Khandelwal et al., 2021) for domain adaption. This approach requires a joint model for speech and text input, with

a fully text-based datastore. Our work does not require modifying the ST model to support speech and text inputs, and enables the retriever to query from speech to speech or text. Our retrieval module is related to the recent work by Lin et al. (2024) as both are based on DPR. The main difference is that their model is for informational retrieval and does not support cross-modal retrieval.

Rare Words in ASR, MT, and ST In ASR, some representative approaches to handle rare words include language model rescoring or fusion (Raju et al., 2019; Yang et al., 2021; Huang et al., 2022; Weiran et al., 2022; Mathur et al., 2023), data augmentation by text-to-speech (TTS) (Guo et al., 2019; Zheng et al., 2021; Qu et al., 2023), and context enhancement by an additional memory module (Bruguier et al., 2019; Jain et al., 2020; Chang et al., 2021; Huber et al., 2021; Qiu et al., 2022; Huber and Waibel, 2024). In MT, rare word translation has been tackled by, among other techniques, constrained decoding (Chatterjee et al., 2017; Hasler et al., 2018; Ailem et al., 2021; Zhang et al., 2023), copying by source annotations (Dinu et al., 2019; Song et al., 2019; Bergmanis and Pinnis, 2021) or pointing mechanisms (Gulcehre et al., 2016; Pham et al., 2018; Gu et al., 2019; Zhang et al., 2021), and retrieval-augmented translation (Martins et al., 2023; Liu et al., 2023). In **direct ST**, translating rare words is a significant challenge due to the combined complexities of ASR and MT. The amount of prior work is also relatively sparse. Gaido et al. (2022) use multilingual models to improve the accuracy of non-English names. Gaido et al. (2023) propose to first detect named entities (NEs) in the source audio that are present in a given contextual dictionary and then inject these NEs in text form into the decoder. Our approach does not assume a readily available contextual dictionary, but can instead leverage unprocessed parallel data.

7 Conclusion

We introduced a retrieval-and-demonstration approach to improve rare word translation accuracy in direct ST. For real-world applications, e.g., translating scientific talks, we recommend adding utterances from the same speaker to the example pool and using speech-to-speech retrieval to identify examples. When feasible, one should consider incorporating an additional verification step to ensure the relevance of the retrieved sentences, by human-in-the-loop or automated techniques.

619 Limitations

620 **Language Coverage in Experiments** Our ex-
621 periments were limited to the English-to-German
622 language pair due to resource constraints. Experi-
623 ments on additional language pairs, especially dis-
624 tant ones, would further substantiate the findings.

625 **Robustness to Irrelevant Examples** Our ap-
626 proach effectively improves the accuracy of rare
627 word translation. However, as elaborated in the re-
628 sult discussions, we also observed that incorrectly
629 retrieved examples tend to harm translation quality.
630 As a next step, we hope to increase the robustness
631 of the ST models to irrelevant examples. This could
632 for instance be achieved by incorporating incorrect
633 rare words during training to enhance the model’s
634 resilience to such errors.

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1121 A Details on Masked Loss

1122 During the training of our adapted ST model, ex-
1123 ample sentences are prepended to sentences in the
1124 reduced training set. The translation of the exam-
1125 ple sentence is used as a prefix and masked during

1126 loss calculation. The cross-entropy loss function
1127 we use for training can be expressed as Equation 1:

$$\mathcal{L} = - \sum_{t=1}^T M_t \log P(y_t | y_{<t}, u^e, y^e, u) \quad (1) \quad 1128$$

1129 With M_t as a mask function Equation 2:

$$M_t = \begin{cases} 0 & \text{if position } t \text{ is part of } y^e \\ 1 & \text{if position } t \text{ is part of } y \end{cases} \quad (2) \quad 1130$$

1131 B Details of Rare Word Types

1132 The detailed rare word analysis results for Table 2
1133 are in Table 9.

Rare Word Type	Frequency
Person	130
Location	72
Technology	29
Food	27
Company	25
Biology	23
Organization	18
Health	18
Culture	14
Transport	14
Religion	14
Fashion	13
Medicine	12
Science	12
Geography	11
Chemics	11
Language	11
History	10
Politics	9
Architecture	9
Military	9
Environment	8
Education	7
Sport	7
Law	6
Society	4
Data	4
Book	4
Physics	4
Game	3
Economy	3
Literature	2
Art	2
Music	1
Entertainment	1
Award	1

Table 9: Detailed NER results on rare words in tst-rare-word with the number of unique words in each category.

1134 C ST Training and Inference Details

1135 C.1 Training Details

1136 We use the Transformer architecture
1137 S2T_TRANSFORMER_S in FAIRSEQ S2T

(Wang et al., 2020) For all our ST models, the encoder-decoder architecture consists of 12 transformer encoder blocks and 6 transformer decoder blocks, with a model dimension of 256 and an inner dimension (FFN) of 2,048.

We initialized the ST model from a pre-trained ASR model⁶. Subsequently, we fine-tuned the pre-trained model for the ST task with hyperparameters following (Wang et al., 2020), specifically, we set dropout rate 0.1 and label smoothing 0.1. The ST training used a tokenizer with a vocabulary size of 8,000. To prevent the tokenizer from seeing the rare words during its training, which will cause an unfair test condition, we train the SentencePiece (Kudo and Richardson, 2018) tokenizer on the reduced train set after the utterances containing rare words are moved to other splits as discussed in §3.1.

During the training of the adapted ST model with examples, we doubled the effective batch size to maintain a comparable loss scale since the prefix tokens do not contribute to the overall loss. Additionally, we set dropout rate to 0.2 after doing a search in {0.1, 0.2, 0.3} based on the dev loss during the training of the adapted ST model. The training was stopped after the validation performance did not improve for 30 consecutive epochs (patience 30). For evaluation, we averaged the last 10 checkpoints.

C.2 Inference Details

The inference uses a beam size of 5. Since the rare-word-tst dataset includes example-prepended sentences, the sentences are longer than typical translation sentences. To keep all utterances in the rare-word-tst set, we set a large allowed source size with `-max-source-positions 30000`. This ensures that even the longest utterances are not excluded from the rare-word-tst set.

D Retriever Training and Inference Details

D.1 Training Details

Our retriever is based on the DPR (Karpukhin et al., 2020) architecture, where a dense passage encoder E_P and a question encoder E_Q is constructed to map candidate input c and query input q to latent representation vectors respectively. The similarity between the candidate representation and the query representation is defined as the dot-product of their

vectors as shown in Equation 3:

$$\text{sim}(q, c) = E_Q(q)^T E_P(c) \quad (3)$$

The encoders E_P and E_Q of DPR are initialized with SpeechT5 encoder (Ao et al., 2022) or SONAR encoder (Duquenne et al., 2023).

Speech T5 The SpeechT5 speech/text encoder transforms speech or text input into a 768-dimensional embedding vector. It comprises 12 Transformer encoder blocks, each with a model dimension of 768 and an inner feed-forward network (FFN) dimension of 3,072. Before the encoder, a speech/text-encoder pre-net preprocesses the input. The speech-encoder pre-net includes the convolutional feature extractor of wav2vec (Baevski et al., 2020) for waveform downsampling. The text-encoder pre-net applies positional encoding to convert character-level tokenized indices into embedding vectors.

SONAR The SONAR speech/text encoder encodes speech/text input to an embedding vector of 1,024. The encoder consists of 24 transformer encoder blocks with a model dimension of 1,024 and an inner dimension (FFN) of 8,192. The speech encoder-frontend applies the wav2vec feature extractor (Baevski et al., 2020), while the text encoder-frontend uses a position encoder.

Training The dual encoders in DPR are trained on a reduced training set with prepended examples. Each sentence’s example works as a positive example, while examples from other sentences in the batch serve as in-batch negatives. We set a batch size of 4 and a learning rate of $2e-5$ for training.

Given the large size of the SONAR encoder, for memory efficiency, only the top layer of the encoder is trained. This approach is not only for memory efficiency but also because the lower layers likely extract low-level acoustic features, which are less relevant for our retrieval task focused on word-level information. We further investigate the retrieval accuracy under different numbers of trainable parameters. As shown in Figure 2. We use the settings with the best retrieval accuracy for our ST task, which are:

- For the speech-to-speech retriever, the top 2 layers of both speech encoders are trained, resulting in 205 million trainable parameters.
- For the speech-to-text retriever, the top 8 layers of both the text and speech encoders are trained, with 422 million trainable parameters.

⁶https://dl.fbaipublicfiles.com/fairseq/s2t/mustc_de_asr_transformer_s.pt

- For the text-to-text retriever, the top 8 layers of both text encoders are trainable, totaling 335 million trainable parameters.

D.2 Inference Details

During inference time, we apply the passage encoder E_P to all the candidates in the rare-word pool. Given a question q , we can derive its embedding $v_q = E_Q(q)$ and then retrieve the top-1 candidate whose embedding is the closest to v_q from the rare-word pool.

E Comparison to Existing Results

We confirm that our baseline model performs on par with those reported in the literature with the results in Table 10.

	BLEU
FAIRSEQ S2T (Wang et al., 2020)	22.7
Our baseline model	23.6

Table 10: The performance of our baseline model on the tst-COMMON split of MuST-C is comparable to existing baselines. Both models have the identical architecture using S2T_TRANSFORMER_S.

F Additional Examples

Here we present two additional translation examples for comparison among the baseline model, the model trained with an additional rare-word pool, and our approach. In the first example, our approach successfully translates a zero-shot word perfectly. In the second example, we demonstrate a case where our approach does not perform well.

source (transcript): **Murali Krishna** (Murali Krishna) comes from one of those villages.

baseline model (on train-reduced) (Table 3 row (1)): **Moralische Christen** (Moral Christians) sind aus einem dieser Dörfer.

train on {train-reduced + rare-word pool} (Table 3 row (4)): Das **Marate Krishna** (Marate Krishna) kommt aus einem dieser Dörfer.

speech→speech example (Table 4 row (5)): Sie arbeitet mit Leuten wie **Murali Krishna**. (She works with people like Murali Krishna.)

adapted + speech→speech (Table 3 row (7)): **Murali Krishna** (Murali Krishna) kommt aus einem dieser Dörfer.

target: **Murali Krishna** (Murali Krishna) kommt aus einer dieser Dörfer.

source (transcript): The **McLaren** (McLaren) just popped off and scratched the side panel.

baseline model (on train-reduced) (Table 3 row (1)): Und der **Klient** (client) stoppte ab und kratzte die Seite des Paddels.

train on {train-reduced + rare-word pool} (Table 3 row (4)): Und der **Spieler** (player) stürzte einfach ab und kratzte auf den Bürgersteig.

speech→speech example (Table 4 row (5)): Aber als Nebeneffekt sammelt er Kornette. (But as a sideline, he happens to collect cornets.)

adapted + speech→speech (Table 3 row (7)): Als der **Klainer** (Klainer) gerade ankam, stopfte er ein Nebenpannel.

target: Der **McLaren** (McLaren) bekam eine Beule und einen Kratzer an der Seitenkarosserie.

Table 11: Additional examples of our retrieval-and-demonstration approach.

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