uDistil-Whisper: Label-Free Data Filtering for Knowledge Distillation in Low-Data Regimes

Anonymous ACL submission

Abstract

Recent work on distilling Whisper's knowledge into small models using pseudo-labels shows promising performance while reducing the size by up to 50%. This results in small, efficient, and dedicated models. However, a critical step of distillation from pseudo-labels involves filtering high-quality predictions and using only those during training. This step requires ground truth labels to compare and filter low-quality examples making the whole process supervised. In addition to that, the distillation process requires a large amount of data thereby limiting the ability to distill models in low-resource settings. To address this challenge, we propose a distillation framework that does not require any labeled data. Through experimentation, we show that our best distilled models outperform the teacher model by 5-7 points in terms of WER compared to those without filtering and are on par with or perform better than similar supervised data filtering setups. When we scale the data, our models significantly outperform all zero-shot and supervised models. We demonstrate that it is possible to distill large Whisper models into relatively small ones without using any labeled data. Our distilled models are also 25-50% more compute- and memory-efficient while maintaining performance equal to or better than that of the teacher model.

1 Introduction

In an effort to democratize automatic speech recognition (ASR), significant attention has been given to multilingual models (Pratap et al., 2023; Radford et al., 2023; Communication et al., 2023; Zhang et al., 2023). These powerful systems, thanks to their massive number of parameters and training data, can simultaneously transcribe in hundreds of languages. That being said, low-resource languages tend to trail behind in performance compared to high-resource counterparts such as English (Radford et al., 2023). For instance, while OpenAI's latest ASR model Whisper-large-v3 shows the word error rate (WER) in the single digits on English and Spanish, a much lower performance is shown for East Asian and African languages[1] (Talafha et al., 2023). While the performance reported on these multilingual models is not as low for Arabic, it is important to highlight that it does not fully and accurately represent the diversity of the different varieties that characterize the Arabic language family (Abdul-Mageed et al., 2020; Habash, 2022). In addition to the disparity in performance, multilingual models are extremely compute-intensive and are thus not equally accessible to everyone. In order to alleviate this issue, knowledge distillation has been the method of choice for multiple works (Hinton et al., 2015; Yang et al., 2023; Sanh et al., 2020; Frantar et al., 2022).

Knowledge distillation has proven to be highly effective in reducing model size, thereby lowering compute and memory requirements while maintaining performance comparable to larger teacher models. For example, Gandhi et al. (2023) and Waheed et al. (2024) demonstrate its effectiveness for English and Arabic, respectively. However, a significant limitation in these works is their reliance on ground-truth labels to filter out low-quality pseudo-labels generated by the teacher model—a resource that is often scarce, particularly for low-resource languages. Additionally, Waheed et al. (2024) shows that training on unfiltered data leads to sub-optimal performance. This reliance on labeled data underscores the need for an unsupervised approach to data filtering in knowledge distillation, enabling distilled models to perform better without depending on any ground-truth labels.

In this work, we propose frameworks to distill

[1] https://github.com/openai/whisper/discussions/1762

knowledge from teacher models that do not rely on filtered pseudo-labels. Our method achieves a similar performance to supervised distillation and outperforms pseudo-label distillation without filtering. Our contributions are the following:

- We explore various unsupervised methods to filter out low-quality pseudo-labels, eliminating the need of labeled data for distillation.
- We assess the performance of our distilled models across diverse settings and extend the existing evaluation to include new datasets.
- We analyze the effectiveness of our best metrics in detecting low-quality pseudo-labeled examples.

We organize the paper as follows: we present a review of the existing literature in Section 2 before introducing knowledge distillation along with our methodology in Section 3. Section 4 contains particulars of our experiments and Section 5 delves into the results we obtain. We discuss these results in Section 5, conclude in Section 6, and outline our work's limitations in Section 7.

2 Related Work

Scaling speech foundation models in both data and model size has led to systems capable of handling a wide range of speech tasks (Pratap et al., 2023; Zhang et al., 2023; Radford et al., 2023; Communication et al., 2023). These models effectively transcend language barriers, demonstrating robust performance across numerous languages, including many low-resource ones (Hsu et al., 2024; Yeo et al., 2024). However, these models face two key limitations: (1) their large size (Radford et al., 2023; Communication et al., 2023), which results in high computational costs, and (2) the under-representation of low-resource languages in training, leading to suboptimal performance on these languages (Radford et al., 2023; Talafha et al., 2023). To address the first issue various efficient decoding methods (SYSTRAN; Segal-Feldman et al., 2024; Malard et al., 2023; Leviathan et al., 2023) were proposed. However, resulting models do not necessarily enjoy a reduced memory footprint.

To address the efficiency issue, knowledge distillation (Hinton et al., 2015; Gou et al., 2021) is proven to be very effective. Different approaches for knowledge distillation have been explored such as Patient Knowledge Distillation (PKD) (Sun et al., 2019), T-S learning (Manohar et al., 2018), Cross-Modal Hashing (CMH) (Hu et al., 2020), Joint Unsupervised Domain Adaptation with KD (Zhang et al., 2021) and Data-Free KD (Lopes et al., 2017) among others in the NLP and CV fields alike. Knowledge distillation has also been applied to speech recognition: Shao et al. (2023) shrink a Whisper (Radford et al., 2023) model tangibly ($\sim 80\%$ of its original size), all while improving its performance; Chang et al. (2021) reduce a Hubert model by to 75% of its initial size through layer-wise distillation without a significant drop in performance.

Knowledge distillation has also been applied to speech-to-text tasks (Nayem et al., 2023; Hentschel et al., 2024; Tian et al., 2022) to reduce the memory and compute requirements. More recently, it has been used to distill multilingual models from Whisper (Ferraz et al., 2024) and strong monolingual models (Gandhi et al., 2023), as well as in low-resource settings. An example of the latter is Waheed et al. (2024) who investigate pseudo-label distillation methods (Gandhi et al., 2023) across various Modern Standard Arabic (MSA) and dialectal Arabic datasets. Their exploration demonstrate the efficacy of distillation in enhancing both efficiency and performance, showcasing that smaller dedicated models can outperform larger multilingual ones. While the pseudo-labeling approach has proven effective across diverse languages (Gandhi et al., 2023; Waheed et al., 2024), it traditionally requires labeled data for filtering low-quality pseudo-labeled examples. In this work we address this limitation by introducing an *unsupervised* framework based on data filtering methods, thereby eliminating the dependency on labeled data altogether.

3 Methodology

3.1 Knowledge Distillation

Knowledge distillation is a framework through which a small student model learns the behavior of a bigger teacher model (Hinton et al., 2015; Sanh et al., 2020; Kim and Rush, 2016). Gandhi et al. (2023) introduce knowledge distillation via pseudo-labeling, which generates (English) predictions from a teacher Whisper model and filters them through a WER threshold to only keep the most accurate ones. These pseudo-labels are then used to train a smaller student model. Waheed et al. (2024) build on this approach and show that a Whisper student model beats its teacher model in its average

performance in an Arabic multi-dialectal setting. We follow this standard student-teacher framework to distill Whisper into small yet powerful models. The objective of the distillation process can be stated as:

$$\mathcal{L}_{KD} = \alpha_{KL}\mathcal{L}_{KL} + \alpha_{PL}\mathcal{L}_{PL} \quad (1)$$

where:

- $\mathcal{L}_{KL}$ is the Kullback-Leibler (KL) divergence loss, encouraging the student model to match the teacher's probability distribution.
- $\mathcal{L}_{PL}$ trains the student using pseudo labels as ground truth.

The coefficients α_{KL} (0.8) and α_{PL} (1.0) balance the contributions of each loss in the overall distillation loss $\mathcal{L}_{KD}$.

3.2 Label-Free Data Filtering

Previous data filtering methods, such as those used by Gandhi et al. (2023) and Waheed et al. (2024), rely on ground-truth labels by calculating WER between teacher-generated pseudo-labels and reference transcripts, discarding examples with high WER. However, this dependence on labeled data limits their use in low-resource languages where such labels are scarce.

Our approach introduces label-free filtering methods that assess pseudo-label quality using the teacher model's logits, synthetic speech, proxy models, and multimodal embeddings. Each method is detailed below.

3.2.1 Proxy Models

We use a trained ASR model, SeamlessM4T-large-v2 (Communication et al., 2023), as a proxy to generate reference transcripts for the input speech. Then, the quality of pseudo-labels generated by the teacher model is evaluated by calculating the WER between the proxy's and the teacher's outputs, referred to as pWER. Lower pWER values indicate higher agreement and examples exceeding a defined pWER threshold are discarded.

3.3 Uncertainty Quantification

We leverage uncertainty in the teacher's output to filter low-quality examples, using two common metrics:

Entropy. Entropy measures uncertainty in the teacher's predicted probability distribution. High entropy suggests low confidence. It is calculated as:

$$H = -\sum_{i=1}^{N} p_i \log_2(p_i) \quad (2)$$

where p_i is the predicted probability for the ith word.

Geometric Mean of Confidence Scores. The geometric mean of the confidence scores, representing the probability of each decoded token in the teacher-generated pseudo-label, is used to assess the overall confidence in the pseudo-label. It is calculated as:

$$G = \sqrt[N]{\prod_{i=1}^{N} c_i} \quad (3)$$

where c_i is the confidence score for each word. Low entropy and high geometric mean indicate better pseudo-labels.

3.3.1 Negative Log-Likelihood

We compute the negative log-likelihood (NLL) using AceGPT-7B (Huang et al., 2024) to assess the quality of pseudo-labels. Lower NLL suggests higher pseudo-label accuracy. It is calculated as:

$$NLL = -\sum_{t=1}^{T} \log(p(y_t|y_{1\ldots t-1})) \quad (4)$$

where $p(y_t|y_{1\ldots t-1})$ is the probability assigned to the tth word.

3.3.2 Multimodal Embeddings

We use SONAR (Duquenne et al., 2023) to generate embeddings for input speech and pseudo-labels. The dot product is used to compute similarity, with higher scores indicating better alignment and pseudo-label quality.

3.3.3 Similarity with Synthetic Speech

Synthetic speech is generated from pseudo-labels using XTTS-v2[2] and compared with input speech. Similarity is assessed via PESQ and MCD, where high PESQ and low MCD scores indicate higher label accuracy.

3.4 Training Data

We follow Waheed et al. (2024) for the data mixture to train our models. As a result, we randomly select segments from a diverse set of datasets, including MGB2 (Ali et al., 2019a), MGB3 (Ali et al.,

[2] https://huggingface.co/coqui/XTTS-v2

2017), FLEURS (Conneau et al., 2022), Common-Voice 15.0 (Ardila et al., 2020), QASR (Mubarak et al., 2021), the Arabic Speech Corpus (Halabi, 2016), and the Massive Arabic Speech Corpus (MASC) (Al-Fetyani et al., 2021). Specifically, we sample 100K and 500K segments, equivalent to approximately 100 and 500 hours of pseudo-labeled speech data, respectively. Our dataset compilation strictly includes the train splits only from each source. We filter out roughly 27% bad quality examples from our training data using the metrics described in Section 3.

3.5 Teacher and Student Models

In line with previous work (Gandhi et al., 2023; Waheed et al., 2024), we utilize the *whisper-large-v2* checkpoint[3] for both pseudo-labeling and as the teacher model during training. We train two variants of students, differing in the number of layers removed from the teacher model. Following Gandhi et al. (2023) and Waheed et al. (2024), we initialize the student models with maximally spaced layers in the encoder and decoder block of the teacher model. We refer to the models with 16-16 and 32-16 encoder-decoder blocks as *UDW-16-16* and *UDW-32-16*, respectively, when we train with 100K segments and as *UDW-16-16++* and *UDW-32-16++* when we train with 500K segments.

4 Experiments

For a fair comparison and analysis, we keep our experimental setup quite identical to Waheed et al. (2024). Apart from evaluating our models on five standard benchmark datasets, and novel dialectal data introduced in Waheed et al. (2024), we extend the evaluation to include the SADA (Alharbi et al., 2024) dataset. We describe our evaluation datasets below.

4.1 Evaluation Dataset

FLEURS. The Few-shot Learning Evaluation of Universal Representations of Speech (FLEURS) (Conneau et al., 2022) is a multilingual speech dataset with 102 languages, each with 12 hours of speech. It supports tasks like ASR, language identification, and translation. We use the Arabic subset, which features MSA speech with Egyptian accents.

[3] https://huggingface.co/openai/whisper-large-v2

Common Voice. Common Voice (CV) (Ardila et al., 2020) is a volunteer-driven multilingual dataset with 124 languages and 31,176 hours of speech. The Arabic subset includes 156 hours of primarily MSA speech.

Multi-Genre Broadcast. The MGB dataset (Ali et al., 2019a, 2017, 2019b) consists of MGB2 (1,200 hours of MSA-dominant Aljazeera Arabic speech), MGB3 (6 hours of Egyptian dialectal speech), and MGB5 (14 hours of Moroccan dialectal speech).

SADA. The Saudi Audio Dataset for Arabic (SADA) (Alharbi et al., 2024) includes 668 hours of Arabic speech (435 labeled), featuring Saudi dialects, MSA, and speech from the Levant, Yemen, and Egypt. Statistics are provided in Appendix B 2.1.

4.2 Baselines

We evaluate several multilingual and monolingual speech recognition models on different Arabic varieties, including standard and accented MSA, as well as various dialects. The models are grouped as follows:

4.2.1 Supervised Models and Commercial Systems.

We evaluate three baselines: Wav2Vec2-XLS-R (Conneau et al., 2020; Babu et al., 2021) trained on CV8.0, HuBERT (Hsu et al., 2021) trained on MGB-3 (Ali et al., 2017) and a 5.5-hour Egyptian Arabic corpus, and a fine-tuned Whisper-large-v2 model trained on CV11.0 and MGB-2. These models have been used off the shelf and been trained by others.

4.2.2 Zero-Shot Models

Whisper. Whisper (Radford et al., 2023) is a versatile multilingual speech model designed for both speech recognition and translation, including Arabic. We assess four Whisper variants: *whisper-small* (W-S), *whisper-medium* (W-M), *whisper-large-v2* (W-L-v2), and *whisper-large-v3* (W-L-v3), using their default decoding parameters with a maximum sequence length of 225 tokens.

SeamlessM4T. SeamlessM4T models excel at generating high-quality transcripts across languages (Communication et al., 2023), but evaluations on them typically focus on English. To bridge this gap, we test three of its versions – *seamless-m4t-medium* (SM4T-M), *seamless-m4t-v1-large* (SM4T-L-v1), and *seamless-m4t-v2-large*

(SM4T-L-v2) – under a zero-shot setting on Arabic ASR, using the default parameters from the model's inference pipeline.

4.2.3 Distilled Models

To compare our results with an equivalent supervised setup, we evaluate two distilled model variants of different sizes. We extend the evaluation to SADA's *test* and *valid* splits, using 16-16 and 32-16 models trained with a WER threshold of 80% on both 100K and 500K segments. Additionally, we include models trained without filtering as a lower bound. For fair comparison with supervised data filtering, we follow an experimental setup identical to prior work (Waheed et al., 2024).

4.3 Experimental Setup

During evaluations, we adhere to the default decoding parameters specified by the original model implementations unless specified otherwise. Throughout our experiments, we maintain a maximum sequence length of 225 and use WER and CER as the primary metrics for quantitative evaluation of the models.

For distillation, we select roughly 27% of the top segments based on the metrics, to match the amount of data used in a supervised setup when $\lambda = 80\%$. We do not conduct hyperparameter search due to computational constraints, and instead apply the configuration detailed in Gandhi et al. (2023). The distillation process and its key parameters are detailed in Table 6 in Appendix C.

5 Results and Discussion

We report the baseline results from Waheed et al. (2024) and evaluate all models on SADA's test and validation splits. We distil two variants of student models. The first student has 16 of each of the encoder and decoder blocks (UDW-16-16) and the second one has 32 encoder and 16 decoder blocks (UDW-32-16). We experiment with 100K and 500K segments. We report our main results in Table 1, and the average across different evaluation setups in Table 2. Results on the top five varieties[4] of SADA are provided in Table 3. The reported results are computed on normalized text that includes removing diacritics. Results on the text with no normalization can be found in Appendix 3.2.

In zero-shot models, we compare our results with MMS (Pratap et al., 2023), three variants of SeamlessM4T family models (medium, large-v1, large-v2), and four Whisper models. We extend the evaluation to the SADA data for these models as well. While larger variants of both SeamlessM4T and Whisper perform well on standard benchmark setups like Common Voice and Fleurs, they perform poorly on novel and challenging settings, such as dialectal speech. For example, the best zero-shot model (SeamlessM4T-large-v2) has a 7.6% WER on Fleurs compared to 65.0% on the SADA test set as shown in Table 1. This underscores the inadequate evaluation and limited generalization capability of zero-shot models in unseen and challenging settings.

As demonstrated in the reported results, the best-distilled model DW-32-16++ outperforms all other models including its teacher (large-v2), the best Whisper (large-v3), and the best baseline (SeamlessM4T-large-v2).

We train two versions of teacher models using datasets of size 100K and 500K segments, respectively. Approximately 27% of the examples were removed from each dataset based on specific filtering metrics, matching the amount of data used in the supervised setup when the filtering threshold is 80.

From our experiments, we observe that two of our best filtering measures, *sonar-sim* and *proxy-ref*, perform similar to or surpass the equivalent supervised setups. For instance, DW-32-16 trained on 100K examples with a filtering threshold of 80 achieves a WER of 35.3% on the benchmark test split, 66.6% on the SADA test split, and 61.1% averaged on five novel dialects. By comparison, our best-performing model in a similar configuration, UDW-32-16-sonar, achieves 35.5% WER on the benchmark test split, 58.7% on the SADA test, and 60.6% on five dialects. This reflects close to 3% improvement in average score compared to the supervised setup (51.6 vs 54.3) and more than a 7% improvement over the no-filter setup, as detailed in Table 2.

In addition to that, our experiments show that the smaller student model (UDW-16-16) performs better with *proxy-ref* as a filtering measure. More specifically, the UDW-16-16, experimented with 100K examples and utilizing *proxy-ref* to filter out low-quality pseudo labels, and matches the results of supervised setups (58.5% vs 57.6%) and shows over 6% improvement compared to setups where pseudo-labels are not filtered (64.2%).

Scaling the Data.

[4]We refer labels in SADA as varieties as not all labels are necessarily dialects.

Model	Size	CV15.0	MGB2	MGB3	MGB5	Fleurs	In-house Data					SADA
							ALG	JOR	PAL	UAE	YEM	
Amazon	-/-	-/-	-/-	-/-	-/-	-/-	83.6/70.2	45.5/25.6	52.4/29.0	58.8/40.8	64.7/43.5	-
XLS-R	0.96	89.7/39.4	97.6/53.1	98.7/61.6	99.5/68.0	94.9/43.9	99.7/67.0	99.1/61.4	99.1/61.1	99.4/64.6	99.5/63.6	99.5/68.3
HuBERT	0.31	55.2/18.9	49.6/17.3	**25.2/9.5**	92.4/45.5	34.9/10.9	96.8/44.3	65.2/23.3	73.8/27.9	83.0/36.7	90.5/38.8	75.6/34.5
W-FT	1.5	35.8/21.9	**15.3**/8.1	48.9/26.9	101.4/62.3	9.8/3.4	115.5/69.6	67.8/37.2	69.6/35.4	105.9/69.1	107.1/64.8	92.3/65.7
MMS-all	1.0	106.4/80.9	39.3/13.4	75.3/34.6	89.7/45.9	23.8/6.3	100.2/78.0	89.8/55.4	99.9/75.1	100.1/78.1	100.2/76.6	77.6/38.0
SM4T-M	1.2	16.3/5.7	19.5/9.0	41.4/21.7	83.8/46.6	8.7/3.6	81.1/39.7	46.3/15.9	55.2/20.1	59.8/24.7	68.9/29.5	65.2/39.3
SM4T-L-v1	2.3	19.8/7.3	21.8/10.5	44.4/22.6	89.9/52.1	11.1/5.1	87.9/47.8	50.7/18.8	57.5/23.1	61.8/27.4	72.2/32.5	64.9/37.8
SM4T-L-v2	2.3	**11.3/3.5**	17.3/8.7	36.2/18.6	89.1/53.7	**7.6**/4.0	92.1/52.0	**41.5**/14.6	49.5/17.2	55.9/23.3	69.7/30.7	65.0/41.8
W-S	0.24	40.3/16.4	46.8/24.7	81.4/51.9	226.5/164.8	28.2/8.7	130.7/84.7	68.6/32.9	73.8/36.3	97.8/59.7	107.1/66.7	139.9/103.6
W-M	0.77	29.8/13.2	33.1/18.5	64.3/39.5	127.7/88.3	16.4/5.1	103.7/69.9	50.5/21.1	58.7/24.7	82.5/52.6	86.8/52.0	99.3/74.1
W-L-v2	1.5	19.6/7.8	26.5/15.3	53.0/33.0	99.2/68.9	11.4/3.6	106.4/71.7	42.3/17.0	51.1/22.3	63.8/38.2	77.3/45.5	69.8/51.2
W-L-v3	1.5	15.8/5.2	15.9/**7.6**	35.7/17.3	79.8/44.6	9.7/3.2	101.9/65.4	43.6/16.3	53.4/22.7	63.4/32.7	76.1/38.9	67.2/45.6
DW-16-16	0.80	22.1/7.2	26.0/10.8	50.5/25.1	82.4/43.3	18.8/6.6	83.0/38.5	50.4/18.2	61.0/23.3	64.6/27.7	72.7/31.6	66.7/38.9
DW-32-16	1.12	18.8/5.9	21.1/8.9	43.8/21.4	78.9/**40.4**	14.2/4.8	79.5/33.4	44.4/14.7	55.0/19.5	58.1/22.8	68.5/28.1	72.3/49.2
DW-16-16++	0.80	19.2/6.2	23.0/10.2	47.2/24.8	79.0/42.6	15.0/5.2	79.0/39.0	46.7/17.2	56.4/21.6	60.4/26.8	69.1/31.5	69.9/40.6
DW-32-16++	1.12	17.1/5.5	19.7/8.8	40.7/20.3	**76.6**/40.6	11.1/**3.1**	4.6/**33.3**	41.6/**13.4**	51.4/18.8	53.5/21.1	**63.5/26.8**	60.3/35.8
No-filter												
– DW-16-16	0.80	22.8/7.6	26.1/11.2	54.1/29.7	95.1/59.1	17.6/6.0	93.4/51.6	51.8/20.2	64.7/27.3	68.0/34.0	78.3/38.8	78.4/49.6
– DW-32-16	1.12	21.2/7.3	22.8/10.4	51.3/30.8	90.5/58.8	14.9/4.9	87.3/63.2	47.6/20.0	57.1/24.9	63.9/35.6	73.0/50.9	70.7/53.6
UDW-16-16 (0.80)												
– nnll		23.59/8.15	26.43/11.26	54.37/27.98	92.25/55.25	18.49/6.26	82.44/41.4	63.33/25.7	52.97/20.52	72.03/35.96	93.4/49.2	84.3/55.0
– pesq		24.28/8.41	28.26/12.11	54.16/27.69	93.72/54.88	19.71/6.89	81.54/40.34	64.92/27.41	52.94/20.16	69.67/32.55	86.96/44.17	79.7/50.1
– entropy		23.84/8.1	27.41/12.17	57.29/31.24	94.75/56.64	18.16/6.4	90.05/48	54.76/22.81	64.26/27.67	73.72/37.85	96.35/52.56	92.0/61.8
– conf		23.47/7.83	27.2/11.87	52.66/27.85	87.79/50.73	17.66/6.12	89.15/43.94	53.12/20.29	63.44/25.52	69.03/31.75	79.11/39.27	78.6/49.2
– proxy		22.54/7.48	25.48/11.39	49.41/26.36	84.75/49.97	17.59/7.5	74.53/42.15	61.07/23.69	50.5/19.66	65.43/30.93	84.12/41.94	68.6/46.2
– sonar		24.06/8.04	28.34/11.86	55.09/28.66	85.59/49.21	20.42/7.06	86.93/43.48	56.36/22.61	65.4/27.43	70.7/32.89	76.48/36.13	74.9/45.6
UDW-32-16 (1.12)												
– nnll		18.92/6.24	22.63/10.12	48.08/25.39	94.07/55.53	13.29/4.47	75.43/35.85	55.24/20.88	44.69/16.04	63.71/30.49	83.81/41.38	73.4/46.1
– pesq		21.7/7.5	24.2/10.6	49.0/26.4	87.0/51.0	16.1/5.3	88.8/43.2	47.5/17.2	58.5/22.7	67.0/30.9	75.5/36.4	73.3/48.1
– entropy		19.09/6.53	21.82/10.34	46.93/28.71	87.11/66.87	13.29/4.34	90.99/84.02	46.68/21.07	58.16/31.08	64.94/44.23	76.35/54.25	68.3/52.3
– conf		20.27/6.46	22.76/9.79	46.6/23.42	83.43/48.61	14.53/4.73	**73.42**/36.08	57.25/21.99	61.06/25.82	**47.5/17.7**	85.71/42.82	66.2/41.3
– proxy		19.04/6.17	21.51/9.87	44.29/22.45	82.64/46.05	14.23/6.23	80.36/36.11	45.46/15.6	56.01/20.88	61.63/25.69	69.52/29.49	64.6/41.9
– sonar		17.82/5.62	21.1/8.98	45.29/23.4	80.39/46.97	13.07/4.41	78.97/37	44.3/15.11	54.34/19.56	58.76/24.29	66.82/28.24	58.7/35.6
UDW-16-16++ (0.80)												
– proxy		15.5/4.8	23.3/10.4	46.8/24.3	84.6/48.0	14.2/4.8	82.9/41.6	46.4/16.3	57.0/21.1	61.1/27.5	70.8/33.0	60.2/35.3
– sonar		18.9/6.1	22.8/9.8	47.2/24.2	82.0/47.1	14.9/5.0	81.5/38.9	47.7/17.2	57.4/21.4	62.1/26.2	69.2/29.5	59.4/33.96
UDW-32-16-++ (1.12)												
– proxy		18.1/5.8	20.8/9.2	43.1/21.1	82.3/44.1	12.6/4.2	80.8/37.1	43.1/14.5	54.4/20.1	57.7/23.1	68.5/29.2	**55.1/31.4**
– sonar		17.0/5.3	21.9/9.9	44.1/22.4	79.8/44.2	12.8/3.9	78.2/34.6	54.2/19.2	**44.4/15.1**	58.2/23.2	66.6/27.0	61.6/33.4

Table 1: WER/CER scores after normalization and removing diacritics. All baseline distilled models (dw-) are trained with a filtering threshold of 80 if not specified. Best results are shown in **bold**. Second best results are underlined. We report the score on the test split of each dataset. Abbreviations. **W** - Whisper, **FT** - Finetuned, **M** - Medium, **L** - Large, **S** - Small, **U** - Unsupervised, **D** - Distil.

We scale our dataset from 100K to 500K segments and we train two models, UDW-16-16 and UDW-32-16, employing *proxy-ref* and *sonar-sim* as filtering criteria. As shown in Table 2, our UDW-32-16++ model, trained with 500K segments, performs comparably with its supervised counterpart DW-32-16++. For instance, UDW-32-16++ with *proxy-ref* filtering achieves a WER of 51.4%, similar to DW-32-16++'s 50.0%. Notably, our smaller student model (UDW-16-16++) trained on 500K segments results in superior performance compared to models trained under supervised filtering. For example, UDW-16-16++ with *sonar-sim* gives 53.9% WER compared to 56.2 for DW-16-16++. These results showcase the efficacy of our data filtering methods.

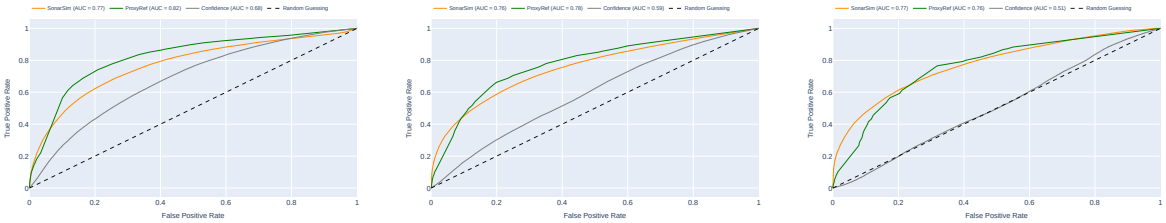
Generalization to Unseen Dialects.

We evaluate our models under various unseen conditions, including five novel dialects that our models have not encountered before. Additionally, we compute category-wise [5] error rates for both the SADA *test* and *valid* splits. We present results on the top five categories. As detailed in Table 3, our best model UDW-32-16++ consistently outperforms all others when averaged across these novel dialects.

Furthermore, UDW-32-16++, demonstrates superior performance over DW-32-16++ in the top five SADA categories. For instance, when using *proxy-ref* as a filtering measure, UDW-32-16++ achieves 58.06% WER, compared to DW-32-16++'s 59.42% averaged across top (with most utterances) five categories in the SADA *test* split. This demonstrates our ability to (1) distill smaller models from larger Whisper models, (2) maintain or improve performance, and (3) reduce model size without relying on labeled data.

Effectiveness of Unsupervised Metrics to Filter Low-Quality Pesudo Labels. We investigate the effectiveness of two of our best metrics for filtering low-quality pseudo-labels, specifically targeting instances with a WER higher than 80%, 40%, and

[5] By *categories* we refer to dialectal and non-dialectal labels in the SADA data.



(a) WER > 80 (b) WER > 40 (c) WER > 20

Figure 1: Area under the curve (AUC) for detecting low-quality examples (WER > 20%, 40%, 80%). The Y-axis represents the true positive rate (TPR), and the X-axis represents the false positive rate (FPR).

Model	Bench		SADA2022		IH
	Test	Valid	Test	Valid	
HuBERT	51.4/20.4	54.4/22.7	75.6/34.5	73.7/31.9	81.9/34.2
W-FT	42.2/24.5	47.4/28.6	92.3/65.7	81.7/56.2	93.2/55.2
SM4T-v1	37.4/19.5	39.5/21.2	64.9/37.8	60.3/35.6	66.0/29.9
SM4T-v2	32.3/17.7	34.7/19.4	65.0/41.8	63.2/40.8	61.7/27.6
W-M	54.3/32.9	59.6/38.0	99.3/74.1	91.9/66.7	76.4/44.1
W-L-v2	41.9/25.7	46.8/29.7	69.8/51.2	66.5/48.5	68.2/38.9
W-L-v3	**31.4/15.6**	**33.0/17.1**	67.2/45.6	58.4/39.3	67.7/35.2
DW-16-16	40.0/18.6	42.2/20.5	66.7/38.9	65.2/37.8	66.3/27.8
DW-32-16	35.3/16.3	37.7/17.9	66.9/47.3	64.5/43.9	61.1/23.7
DW-16-16++	36.7/17.8	38.8/19.4	69.9/40.6	64.7/36.6	62.3/27.2
DW-32-16++	33.0/15.7	34.9/**17.1**	60.3/35.8	57.1/33.4	**56.9/22.7**
No-Filter					
– DW-16-16	43.1/22.7	46.5/25.1	78.4/49.6	72.5/45.9	71.2/34.4
– DW-32-16	40.1/22.5	43.6/25.8	70.7/53.6	66.0/45.0	65.8/38.9
UDW-16-16					
– proxy	40.0/20.5	42.3/22.1	68.6/46.2	65.4/42.2	67.1/31.7
– sonar	42.7/21.0	44.9/22.8	74.9/45.6	71.2/43.5	71.2/32.5
UDW-32-16					
– proxy	36.3/18.2	39.1/19.6	64.6/41.9	58.7/36.1	62.6/25.6
– sonar	35.5/17.9	37.8/19.7	58.7/35.6	56.4/34.7	60.6/24.8
UDW-16-16++					
– proxy	36.9/18.4	41.4/21.2	64.1/39.2	60.2/35.3	63.6/27.9
– sonar	37.2/18.4	39.3/19.9	61.7/35.4	59.4/34.0	63.6/26.6
UDW-32-16++					
– proxy	35.1/17.1	37.2/18.7	58.9/33.6	55.1/31.4	60.3/23.8
– sonar	35.4/16.9	37.7/18.5	**57.9/33.1**	**54.7/31.3**	60.9/24.8

Table 2: Average WER/CER across different evaluation datasets. Best results are shown in **bold**. Second best results are underlined. WER/CER scores after normalization and removing diacritics.

Model	Split	NJD	MTOS	KHLJ	HJZ	UNK
W-FT	Test	106.4/77.5	77.0/51.8	117.8/85.4	84.2/61.5	139.9/112.2
	Valid	77.1/52.6	63.4/41.1	139.4/100.3	119.1/89.7	140.3/107.6
SM4T-v1	Test	56.5/30.9	74.0/46.0	61.9/32.2	54.6/29.0	69.6/39.4
	Valid	51.9/28.1	68.7/44.0	61.7/31.5	54.2/30.9	62.3/35.2
SM4T-v2	Test	52.6/31.1	76.2/53.1	58.2/30.4	53.3/32.0	74.7/45.1
	Valid	52.2/30.7	75.8/53.7	65.1/35.3	51.1/30.3	59.8/34.4
W-M	Test	88.2/65.8	107.9/79.3	102.8/77.0	86.7/59.7	134.0/122.2
	Valid	80.4/56.9	102.8/75.1	89.5/62.9	72.9/52.0	127.7/106.5
W-L-v2	Test	56.6/39.9	74.0/57.4	80.2/54.4	63.4/39.6	98.2/80.7
	Valid	60.9/41.4	72.9/55.4	67.7/44.9	64.5/43.6	68.0/67.1
W-L-v3	Test	53.1/31.6	73.7/53.7	70.5/44.1	61.7/38.6	89.0/61.3
	Valid	49.3/30.2	65.5/47.7	67.5/39.2	**46.5**/27.2	67.7/49.2
DW-16-16	Test	58.2/30.8	74.9/47.6	65.1/32.7	58.2/30.7	68.4/39.8
	Valid	59.4/31.4	70.6/44.7	66.2/35.2	61.1/32.8	69.9/39.8
DW-32-16	Test	58.7/35.8	74.9/60.1	66.7/38.7	58.7/34.1	70.1/44.5
	Valid	58.3/34.8	69.7/54.1	67.5/37.6	62.7/38.2	68.3/40.1
DW-16-16++	Test	58.8/30.9	81.9/50.7	62.1/31.5	57.5/31.0	77.5/46.8
	Valid	56.8/29.8	72.0/43.8	62.3/31.8	60.2/33.0	75.2/41.0
DW-32-16++	Test	51.7/28.3	67.8/43.1	57.5/29.4	52.6/28.6	67.5/41.3
	Valid	50.3/27.3	61.8/38.3	62.3/34.5	53.7/28.1	66.4/43.0
No-Filter						
– DW-16-16	Test	61.6/34.8	90.4/59.7	72.9/41.4	68.5/42.3	88.8/63.0
	Valid	64.8/38.9	80.3/53.4	71.4/41.9	65.5/37.7	77.0/54.3
– DW-32-16	Test	58.2/42.8	82.3/63.9	67.3/47.0	60.4/45.9	76.6/63.2
	Valid	57.9/35.2	73.7/54.9	68.6/43.3	56.8/36.5	72.3/49.6
UDW-16-16						
– proxy	Test	60.2/35.5	75.6/55.6	65.7/38.9	62.1/39.6	75.6/52.0
	Valid	59.3/34.0	70.7/50.9	66.5/39.1	61.4/37.1	68.7/41.2
– sonar	Test	64.6/35.8	59.9/30.3	84.6/55.7	72.1/38.8	66.6/36.8
	Valid	64.8/35.9	67.5/39.3	78.1/52.4	69.9/38.7	65.3/36.7
UDW-32-16						
– proxy	Test	53.4/31.1	75.7/54.0	60.0/32.1	51.4/30.8	72.8/46.0
	Valid	51.0/29.3	65.9/44.3	58.1/29.1	53.7/28.6	64.9/36.6
– sonar	Test	**48.8**/25.4	52.6/**23.6**	67.5/45.9	58.3/29.9	**48.9/25.5**
	Valid	49.2/26.0	**51.6**/26.7	62.5/44.1	58.9/30.3	52.6/29.5
UDW-16-16++						
– proxy	Test	53.4/29.7	73.9/48.8	61.9/33.8	53.3/29.6	68.6/42.8
	Valid	52.7/27.8	66.6/42.0	62.0/34.3	55.7/32.2	67.3/41.7
– sonar	Test	53.7/28.4	69.4/43.3	61.4/30.8	53.7/27.5	64.6/37.0
	Valid	53.6/27.5	64.4/40.3	62.3/32.4	55.7/30.7	63.7/35.8
UDW-32-16++						
– proxy	Test	48.9/**25.3**	66.7/41.3	60.6/31.0	48.7/**24.6**	66.0/38.2
	Valid	49.0/**25.3**	60.4/37.4	57.6/30.1	50.7/25.6	60.7/37.4
– sonar	Test	50.5/26.3	65.3/40.8	**57.3/28.3**	49.1/24.8	61.8/34.5
	Valid	49.3/**25.3**	58.8/37.0	58.1/30.2	53.6/29.8	61.1/34.2

Table 3: Results on top five dialects/categories in SADA data. Best results are shown in **bold**. Second best results are underlined. WER/CER scores after normalization and removing diacritics.

20%. To assess their efficacy, we calculate the area under the curve (AUC) (as shown in Figure 1) for detecting low-quality examples. The results indicate that *sonar-sim* achieves an AUC of 0.77 for detecting examples with a WER > 80, demonstrating reasonably high discriminative power in identifying low-quality labels. The *proxy-ref* metric shows a slightly better performance, with an AUC of 0.82, indicating robust capability in distinguishing between high and low-quality pseudo-labels. In contrast, the confidence-based measure yielded an AUC of 0.68, which falls behind the other measures' discriminative power. These findings highlight sonar embeddings and the proxy reference-based measure as promising tools for improving the quality of pseudo-labels in scenarios where ground truth data is unavailable.

5.1 Experiments on Other Language.

To further validate the effectiveness of our approach, we conduct experiments on Swahili, a low-resource language. We collect over 100 hours of labeled speech data from a variety of sources, namely OpenBible (Meyer et al., 2022), CommonVoice

Evaluation	Dataset	W-L-v2	DW-16-16	DW-32-16	UDW-16-16-proxy	UDW-32-16-proxy
IID	**OpenBible**	101.33/44.43	59.14/13.95	**58.78/13.75**	59.21/13.96	58.86/14.13
	CommonVoice17	117.13/60.13	82.91/34.96	**69.8/24.77**	75.58/29.18	70.43/25.36
	ALFAA	217.11/143.19	78.24/28.23	74.4/**25.72**	76.82/27.2	**73.81**/26.53
OOD	**DVoice**	214.6/144.57	124.35/74.1	**110.19**/62.61	110.68/**62.4**	114.9/69.14
	AMMI-LigAikuma	**46.71/12.96**	60.1/18.02	51.77/14.37	60.44/18.47	52.19/14.42
	Fleurs	54.61/14.81	60.85/18.85	**51.58/14.8**	58.87/18.45	51.82/14.88

Table 4: WER/CER comparison on Swahili. Best results are shown in **bold**. Second best results are underlined.

(Swahili subset) (Ardila et al., 2020), ALFAA[6], DVoice (Gauthier et al., 2016), AMMI-LigAikum[7], and FLEURS (Swahili subset) (Conneau et al., 2022).

We distill two models, UDW-16-16 and UDW-32-16, using our best filtering method: *proxy-ref*. The training data includes the *train* splits of OpenBible, CommonVoice, and ALFAA, and we evaluate the models on their respective *test* splits. We also test the models on three out-of-distribution (OOD) datasets: DVoice, AMMI-LigAikum, and FLEURS, which were not included in the training data.

We compare our distilled models to the teacher model and evaluate the performance of our unsupervised approach. The results show that our unsupervised distillation models perform on par with, or better than the supervised setup. Additionally, our distilled models outperform the teacher model by a significant margin on both IID and OOD datasets, demonstrating the utility of our approach in extremely low-resource settings. Specifically, the UDW-32-16 model achieves a WER/CER of 58.86/14.13% on the IID OpenBible dataset, compared to the teacher model's 101.33/44.43%. On the OOD dataset FLEURS, UDW-32-16 attains a WER/CER of 51.82/14.88, significantly outperforming the teacher model's 54.61/14.81. Across various datasets, our distilled models consistently outperform the teacher, with UDW-32-16 showing the best results overall. Table 4 presents the WER and CER scores for the different models and datasets.

These findings highlight the strength of our unsupervised data filtering approach, particularly in low-resource scenarios, where labeled data is scarce but the distilled models still perform robustly.

[6] https://github.com/besacier/ALFFA_PUBLIC/tree/master/ASR/SWAHILI

[7] https://github.com/besacier/AMMIcourse

6 Conclusion

In this study, we explore methods for distilling large Whisper models into smaller, more efficient ones. Our proposed techniques for filtering out low-quality examples eliminate the need for labeled data, bridging a gap in prior research and facilitating the creation of compact and effective speech recognition models for settings with limited labeled data. We conduct a comprehensive evaluation of our trained models, comparing them against various strong zero-shot and distilled models. Our results demonstrate that our models can outperform both the teacher model and models distilled with supervised data filtering setups.

We evaluate our models across a diverse range of Arabic varieties and dialects, assessing their generalization and adaptability to linguistic diversity. From large multilingual speech models (Whisper), we distill language-specific models optimized for Arabic ASR. Our evaluation spans three different setups, encompassing a wide array of linguistic varieties (ie: dialects), and reveals that despite their reduced size (25-50%), these distilled models consistently outperform their base counterparts and can compete with state-of-the-art models twice their size—all achieved without the need for labeled data. We also apply our unsupervised distillation approach on Swahili, another low-resource language, which shows a similar performance and proves its efficacy in different setting.

Notably, our model-based unsupervised data filtering methods (proxy and sonar) demonstrate superior robustness across linguistic variations. Moving forward, we aim to explore model-free approaches to further enhance the efficacy of filtering low-quality pseudo labels. Additionally, our future research will extend to include extremely low-resource languages and domains.

7 Limitations

In this study, we distill small Whisper models from relatively large ones via pseudo-labeling and unsupervised data filtering. Our distilled models are computationally efficient and maintain a performance similar to or better than the base teacher model and models trained in a supervised data filtering setup. Unlike (Waheed et al., 2024; Gandhi et al., 2023), our approach does not utilize any labeled data in the distillation process, making it directly applicable in data-scarce settings. However, despite these advantages, we acknowledge several limitations in our work, which we outline below.

Efficiency. Our distilled models achieve 25-50% compute efficiency relative to their larger counterparts while maintaining comparable performance. However, the training of these models requires significant computational resources.

Our main approach relies heavily on a robust reference model to serve as a proxy for filtering lower-quality pseudo labels. Specifically, we utilize SeamlessM4T-large-v2, a state-of-the-art model with 2.3 billion parameters, to generate proxy references which is then used to filter out low-quality data points. For similarity-based measures, we use SONAR() to generate multimodal embeddings from speech and pseudo labels. These embeddings provide contextual similarity which is then utilized to discard low-quality pseudo labels. We use AceGPT (7B), to compute the log-likelihood of the pseudo labels which is leveraged to filter out low-quality examples.

Although these measures allow attaining a performance on par or better than the supervised setup, it's important to highlight that each of these methodologies entails additional computational overhead.

Multilinguality. We use SeamlessM4T-large-v2 for generating proxy references, SONAR for generating multimodal embeddings, AceGPT (7B) for computing log-likelihood, and XTTS-v2 for generating synthetic speech. The multilingual capabilities of these models are crucial for effectively applying our techniques to a wide range of languages and dialects. However, a significant limitation of our approach is that it is constrained to languages supported by these models. This dependency restricts our ability to extend our distillation process to languages beyond the scope of the models' multilingual capacities.

Evaluation. Arabic is a linguistically rich and complex language with over 400 million speakers (Abdul-Mageed et al., 2021), resulting in its wide range of varieties and dialects. We evaluate all the models on ten different datasets representing different varieties, including five novel dialects collected and curated by native speakers and never seen before by any models. However, our varieties do not cover all Arabic-speaking regions. We aim to address this in future work by covering more varieties and dialects.

Distillation Training Data. We distilled four variants of student models using 100K and 500K segments of which approximately 25% are filtered. We see improvement going from 100K ($\approx$100 hours) to 500K ($\approx$500 hours) segments. As Gandhi et al. (2023) shows, going over 1000 hours results in a better model, we aim to study how distillation can be done under a low resource setting which is why we do not scale the data. Additionally, we also keep the WER threshold high (80) so that we remain close to a setting where no labeled data is available (even for filtering). It would be interesting, however, to see how distilled models may perform on unfiltered data in low-resource setting.

Nature of Speech Data. Despite putting together a never-seen dataset of under-represented Arabic dialects, we realize that sourcing our data from television series renders its nature distant from speech spoken *in the wild*. This type of content tends to be more "theatrical" and involves different elements such as background music and laughing tracks that do not accurately reflect regular conversational Arabic. Consequently, this could fail to accurately portray the performance of these models on real speech.

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A Appendix

B Dataset

2.1 SADA Dataset

Dialect	Test(S/D)	Valid (S/D)
Najdi	1703/2.0709	2249/3.3155
MTOS	1320/4.8044	1048/3.82
Khaliji	1150/1.1308	679/0.6317
Hijazi	809/1.1202	528/0.6423
Unknown	762/0.8325	489/0.4861
NA	167/0.1341	2/0.0004
MSA	157/0.5406	54/0.1682
Egyptian	96/0.0865	45/0.0524
Shamali	18/0.0243	-
Yemeni	7/0.0052	23/0.0349
Levantine	-	19/0.0137
Total	6189/10.75	5136/9.17

Table 5: SADA stats. S is the number of segments and D is the duration (in hours).

C Experiments

3.1 Training Parameters

3.2 Results

D Efficacy

Parameter	Value
warmup_steps	50
learning_rate	0.0001
lr_scheduler_type	constant_with_warmup
batch_size	128
max_label_length	225
gradient_accumulation_steps	1
dtype	bfloat16

Table 6: Training parameters. All the training parameters are the default ones provided in Huggingface Seq2SeqTrainingArguments unless specified otherwise in this table.

Model	Size	CV15.0	MGB2	MGB3	MGB5	Fleurs	In-house Data					SADA
							ALG	JOR	PAL	UAE	YEM	
Amazon	-/-	-/-	-/-	-/-	-/-	-/-	88.0/71.6	59.2/29.1	63.4/32.2	71.1/44.3	77.4/47.7	-
XLS-R	0.96	92.7/46.7	97.7/54.5	99.1/64.5	99.6/70.1	95.1/45.4	99.7/68.0	99.3/62.9	99.2/62.8	99.5/66.4	99.7/66.4	99.6/69.5
HuBERT	0.31	76.5/31.0	59.4/20.3	**43.3/16.5**	95.0/48.7	48.9/14.4	96.2/45.6	70.6/25.4	81.5/31.4	87.9/39.9	91.3/40.8	81.3/37.1
W-FT	1.5	70.0/33.8	29.4/10.9	60.1/32.2	105.0/64.3	28.7/7.3	114.5/70.3	75.1/39.0	81.3/38.7	113.7/70.9	110.1/65.6	101.4/67.6
MMS-all	1.0	106.0/82.5	40.3/14.0	77.7/38.1	90.4/48.5	28.8/7.8	100.2/77.8	91.5/56.2	100.0/75.8	100.1/78.4	100.1/76.8	79.8/39.1
SM4T-M	1.2	42.3/18.2	28.1/11.2	50.2/26.8	88.2/50.8	19.5/6.0	84.5/42.8	55.2/18.7	63.0/23.0	68.0/28.1	79.4/34.5	73.2/42.8
SM4T-L-v1	2.3	44.2/19.1	25.9/11.7	52.5/27.6	92.8/55.9	22.6/7.6	89.7/50.3	59.1/21.7	64.7/25.8	69.0/30.3	81.5/37.0	72.4/40.8
SM4T-L-v2	2.3	**37.7/15.8**	22.4/9.9	46.7/23.9	92.1/58.4	19.8/6.5	94.8/55.2	51.3/17.6	**58.5/20.1**	65.6/26.9	80.6/35.5	72.2/44.4
W-S	0.24	68.9/31.8	49.5/25.7	84.8/55.4	228.6/164.5	33.4/10.3	129.15/87.85	75.25/36.55	79.73/39.3	103.83/63	112.69/70.69	144.5/106.6
W-M	0.77	55.1/24.2	37.6/19.6	71.5/43.7	129.7/89.4	24.0/7.1	103.9/71.4	59.0/23.9	66.8/27.6	90.7/55.7	95.2/56.2	106.0/76.3
W-L-v2	1.5	46.9/19.6	33.7/16.9	60.6/37.7	101.1/71.1	19.7/5.6	106.9/74.6	51.2/19.6	60.2/25.2	73.2/41.2	86.9/50.1	78.0/53.5
W-L-v3	1.5	43.2/16.9	**20.4/8.6**	44.6/22.5	82.0/47.7	**16.4/4.8**	103.8/68.9	52.7/18.9	64.3/26.4	72.3/35.9	86.0/43.3	74.6/47.9
DW-16-16	0.80	48.0/18.9	33.2/12.5	57.1/29.6	84.1/46.2	26.2/8.5	83.8/40.2	57.8/20.5	68.2/26.2	72.0/31.0	80.0/35.6	72.0/40.9
DW-32-16	1.12	45.6/17.7	27.7/10.3	51.2/26.1	80.9/**43.4**	22.0/6.6	**80.5/35.1**	52.6/17.1	62.9/22.4	66.7/26.3	77.3/32.6	66.9/47.3
DW-16-16++	0.80	44.1/17.1	28.5/10.5	54.5/28.5	83.2/45.6	22.4/6.9	82.3/38.7	55.4/18.9	65.2/24.9	69.3/28.2	76.8/33.0	76.0/42.7
DW-32-16++	1.12	44.7/17.3	25.2/10.0	48.8/25.2	**79.0**/43.7	20.2/5.0	76.4/35.4	**50.0/15.9**	60.1/21.8	**63.2/24.7**	**73.5/31.5**	67.2/38.1
No-filter												
– DW-16-16	0.0	48.3/19.1	34.2/13.0	60.2/33.9	96.8/61.3	24.9/7.9	93.8/53.1	58.9/22.4	72.3/30.2	75.5/37.2	84.8/42.3	83.9/51.6
– DW-32-16	0.0	47.2/18.8	29.0/11.8	58.3/35.0	92.5/60.8	23.5/7.0	88.0/64.2	55.3/22.4	65.5/27.9	72.5/38.8	80.4/53.8	76.7/55.5
UDW-16-16												
– nnll	0.0	49.12/19.55	32.18/12.64	60.76/32.4	94.05/57.88	25.75/8.09	88.88/44.93	70.14/28.53	59.84/22.8	78.73/38.95	93.3/50.73	89.2/56.9
– pesq	0.0	49.46/19.73	34.27/13.55	60.58/31.99	95.63/57.22	26.8/8.64	87.91/43.83	71.92/30.19	60.01/22.47	76.7/35.72	87.51/45.81	84.7/52.1
– entropy	0.0	49.03/19.5	33.4/13.57	62.74/35.19	96.45/58.9	25.81/8.29	90.31/49.75	61.72/25.08	71.29/30.42	80.69/40.86	101.99/55.25	96.8/63.7
– conf	0.0	48.73/19.3	34.01/13.44	59.2/32.17	89.77/53.36	24.68/7.86	89.61/45.69	60.05/22.57	71.27/28.52	76.71/35.08	85.44/42.53	84.0/51.3
– proxy	0.0	47.9/18.76	30.96/12.7	56.03/30.8	86.44/52.66	24.69/9.25	81.7/45.74	68.43/26.62	58.08/22.07	72.96/34.2	84.6/43.64	74.1/48.2
– sonar	0.0	48.82/19.4	33.88/13.19	61/33.03	87.08/52.03	27.75/8.91	87.6/45.09	62.72/24.81	71.92/30.21	76.9/35.99	82.49/39.72	78.9/47.3
UDW-32-16												
– nnll	0.0	45.55/17.88	29.02/11.52	54.66/29.79	96.01/57.91	20.19/6.13	84.03/39.91	63.45/23.8	52.63/18.43	71.98/33.71	84.57/43.01	79.6/48.0
– pesq	0.0	47.5/19.0	32.0/12.4	55.6/30.8	88.9/53.8	23.5/7.1	89.3/44.9	55.2/19.6	66.4/25.6	75.0/34.2	83.3/40.4	79.0/50.0
– entropy	0.0	45.68/18.18	29.86/12.07	53.88/32.97	88.92/68.55	21.55/6.32	91.78/84.74	54.32/23.38	66.47/33.82	73.62/47.04	83.91/56.95	74.7/54.1
– conf	0.0	47.07/18.29	31.12/11.62	54.04/28.14	85.53/51.41	22.78/6.68	81.01/39.84	65.69/25	69.83/29.36	55.32/20.11	86.59/44.52	72.9/43.6
– proxy	0.0	45.38/17.71	28.22/11.32	51.47/27.06	84.54/48.86	21.41/7.93	81.25/37.94	53.42/18.05	63.9/23.76	70.19/29.16	78.38/34.03	71.0/44.1
– sonar	0.0	44.74/17.38	27.35/10.35	52.72/28.08	82.33/49.95	21.05/6.29	80.58/38.85	52.29/17.53	63.17/22.67	67.71/27.87	75.42/32.57	65.0/37.7
UDW-16-16++												
– proxy	0.0	52.0/22.5	28.4/11.6	53.8/28.8	86.5/50.6	21.9/6.7	83.6/43.2	54.2/18.8	64.9/24.1	69.3/30.9	78.9/37.2	66.2/37.3
– sonar	0.0	45.5/17.7	29.9/11.4	53.9/28.8	83.7/50.1	22.8/7.0	82.6/40.6	55.4/19.6	65.0/24.4	70.2/29.6	77.5/33.9	65.7/36.0
UDW-32-16-++												
– proxy	0.0	44.9/17.5	26.6/10.5	50.4/25.8	84.4/47.1	20.7/6.1	81.6/39.0	51.5/17.0	62.6/23.0	66.5/26.6	77.6/33.8	62.1/33.6
– sonar	0.0	44.2/17.1	29.0/11.5	51.2/27.0	81.7/47.2	20.6/5.7	79.5/36.5	62.3/22.2	52.5/17.5	67.1/26.8	76.0/31.8	54.7/31.3

Table 7: WER/CER scores on orthographic transcription. Average is the mean score across all the evaluation sets. All distilled models are trained with a filtering threshold of 80. We report the score on the test split of each dataset.