



Training Data Selection Strategies for Multi-Speaker Text-to-Speech Synthesis in Lithuanian

Master's Thesis Defense

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Context & Problem Statement

- **Text-to-Speech (TTS)** systems (also known as speech synthesis) are widely used in virtual assistants, visual aids, and other applications.
- State-of-the-art **Neural TTS** typically requires 10–20 hours of high-quality *single-speaker* data.
- **Low-resource languages** like Lithuanian rarely possess such datasets.
- *Liepa-2* corpus contains 939 hours of annotated Lithuanian speech, but it is **fragmented** across 2,621 speakers.
- **Multi-speaker TTS** models can leverage such data.
- **Question:** What is the optimal strategy for composing multi-speaker datasets to maximize synthesis quality?

Aim & Hypothesis

Research Aim

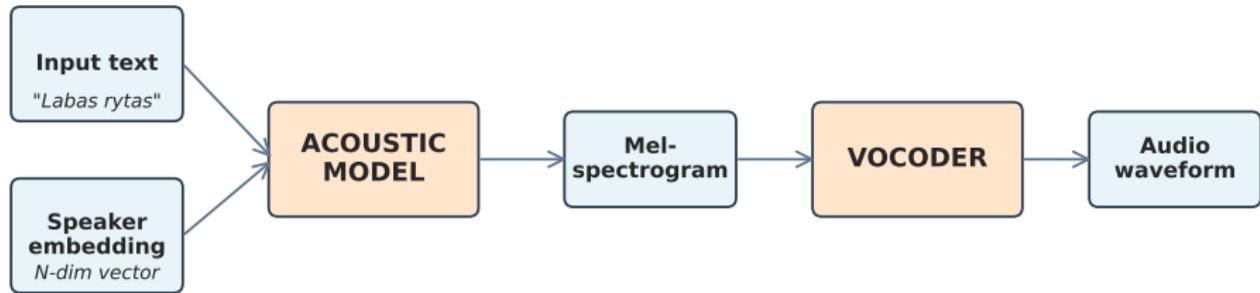
Investigate how varying training dataset **breadth** (number of speakers) and **depth** (duration per speaker) affects the synthesis quality of multi-speaker TTS models under a constant total data budget.

Hypothesis: Data *depth* is the critical factor. Synthesis quality will degrade as the number of speakers increases with the total training duration held constant.

Research Objectives

- ① Prepare 3 subsets of *Liepa-2* with varied breadth/depth but constant total duration.
- ② Configure and train two distinct acoustic model architectures — one autoregressive (AR), one non-autoregressive (NAR).
- ③ Evaluate the synthesis quality of the trained models using established objective evaluation metrics.
- ④ Develop a subjective evaluation application and conduct listening tests to assess naturalness.

TTS Synthesis Pipeline

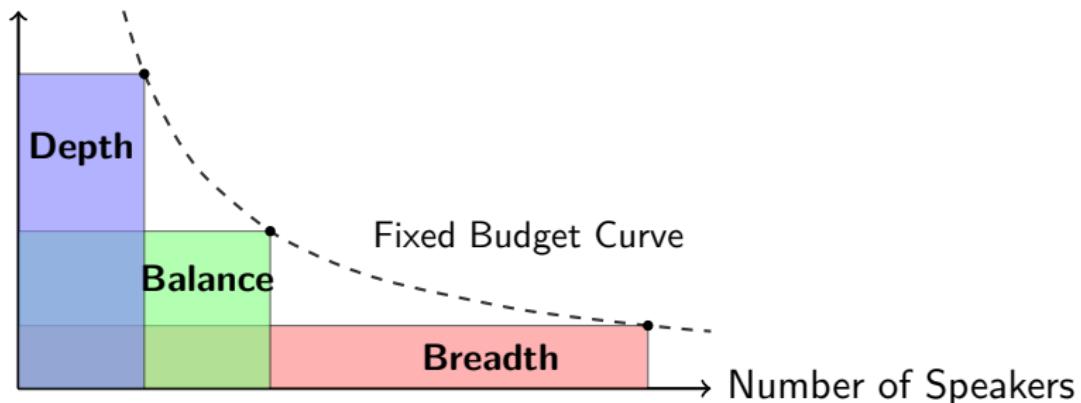


TTS Synthesis Pipeline

A typical neural TTS pipeline. An **Acoustic model** generates Mel-spectrograms from input text, which are then converted to raw audio waveforms by a **Vocoder**.

Experimental Design: Constant “Data Budget”

Duration per Speaker



Total “data budget” is fixed. The three strategies represent different points along the breadth-depth trade-off curve.

Experimental Design: Constant “Data Budget”

Total “data budget” fixed at **22.5 hours**.

Speakers are nested ($30 \subset 60 \subset 180$) and gender-balanced to ensure fair comparison.

Strategy	Speakers	Depth/Speaker	Total Budget
Depth	30	45.0 min	22.5 h
Balance	60	22.5 min	22.5 h
Breadth	180	7.5 min	22.5 h

Data Preparation

- **Source:** *Liepa-2* corpus — 939 hours of read Lithuanian speech from 2,621 speakers.
- **Segmentation:** Utterance-level audio segments sliced using provided timestamps.
- **Filtering:** Adult (18+ years) speakers only, read speech (not spontaneous).
- **Text:** Grapheme normalization done using rule-based preprocessing. Kirčiuoklis-based accentuation applied (where accents are unambiguous, 82% of total words).
- **Audio:** Resampled to 22.05 kHz to match pre-trained vocoder.

Model Architectures

Tacotron 2

- Autoregressive.
- Seq2seq (encoder-decoder) architecture.
- Uses Dynamic Convolution Attention.

Glow-TTS

- Non-autoregressive (parallel).
- Flow-based generative model.
- Uses Monotonic Alignment Search.

Speaker Embeddings: 512-dimensional learnable embeddings, jointly trained with the acoustic models.

Vocoder: *HiFi-GAN* (pre-trained on VCTK). Frozen during training to isolate acoustic model performance.

Model Training

- **Hardware:** Personal high-performance workstation with a 48-core CPU, 256 GB of RAM, and an NVIDIA RTX 3090 GPU.
- **Hyperparameters:** Based on default configurations, with adjustments to Mel-spectrogram parameters (to match vocoder), batch size (due to hardware constraints), learning loss schedules (for a balance between convergence speed and optimal loss values), Lithuanian grapheme set, and other minor tweaks empirically found to reduce validation loss.
- **Training:** On each dataset, from scratch, until validation loss convergence.
 - Tacotron 2 for 90k steps (\approx 22 hours) each.
 - Glow-TTS for 180k steps (\approx 25 hours) each.

Evaluation Setup

Objective Metrics:

- Mel-Cepstral Distortion (MCD): measures spectral distortion in dB.
- Fundamental Frequency RMSE (F_0 RMSE): measures pitch contour error in Hz.

Subjective Evaluation:

- Mean Opinion Score (MOS) test
- 6 randomly selected speakers
- 60 identical test sentences for each speaker
- Latin Square design

Evaluation Setup: Custom MOS Application

TTS Rating System

Rate Audio Statistics Logout

Rate This Audio Sample

Rate the **naturalness** of the audio sample

Please use headphones for accurate evaluation

Progress: 53/60 completed (7 remaining)

Text Prompt

šiandien iš leto keblindamas prie savo automobilio kažkodėl pagalvojo

Audio Sample

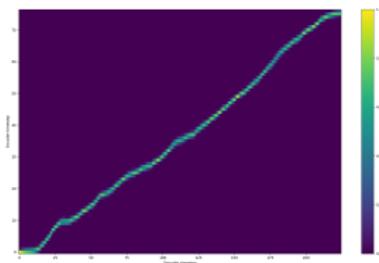
▶ 0:04 / 0:04

Bad	Poor	Fair	Good	Excellent
Very annoying / Unintelligible	Annoying	Slightly annoying	Perceptible but not annoying	Imperceptible from real speech
1	2	3	4	5

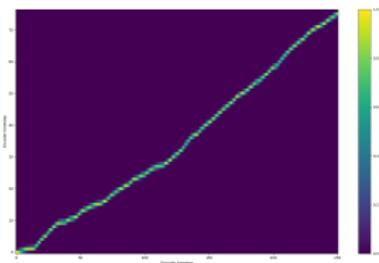
Submit Rating

Results: Alignment Convergence Examples

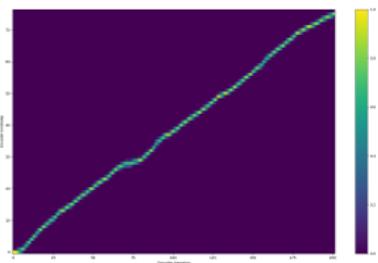
All models achieved successful **alignment convergence**.



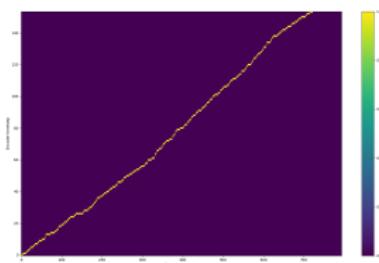
(a) Tacotron 2, 30 spk



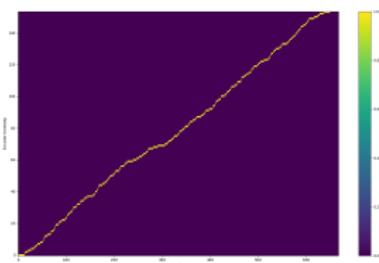
(b) Tacotron 2, 60 spk



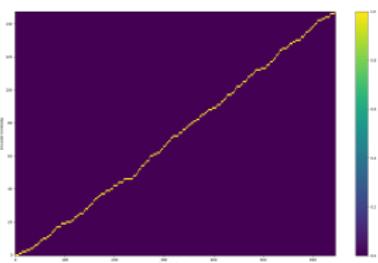
(c) Tacotron 2, 180 spk



(d) Glow-TTS, 30 spk



(e) Glow-TTS, 60 spk



(f) Glow-TTS, 180 spk

Results: Objective Evaluation

Objective metrics calculated on a 60-sentence test set (excluded from training data) by comparing synthesized audio to ground truth.
Lower is better for both metrics.

Model	Speakers	MCD (dB)	F_0 RMSE (Hz)
Tacotron 2	30	9.58	31.28
	60	9.55	30.49
	180	9.63	31.06
Glow-TTS	30	9.90	37.86
	60	10.00	36.18
	180	9.98	35.69

Note: Minimal variation across strategies within each architecture.

Results: Subjective Naturalness (MOS)

Listening test with 21 native Lithuanian speakers (1,260 ratings).
Average MOS per model architecture and data strategy shown below:

Strategy	Tacotron 2	Glow-TTS
Depth ($30 \times 45.0 \text{ min}$)	3.11 ± 0.16	2.13 ± 0.12
Balance ($60 \times 22.5 \text{ min}$)	3.12 ± 0.17	2.18 ± 0.15
Breadth ($180 \times 7.5 \text{ min}$)	3.03 ± 0.18	2.03 ± 0.14
Ground Truth	4.84 ± 0.06	

Observations

- **No significant difference** between data selection strategies within any architecture.
- Hypothesis: **Rejected**.
- Tacotron 2 significantly outperforms Glow-TTS in naturalness.

Results: Speaker-Dependent Naturalness

Speaker ID	Tacotron 2 MOS	Glow-TTS MOS
AS009	4.17 ± 0.20	2.61 ± 0.23
IS031	3.26 ± 0.20	2.13 ± 0.19
IS038	3.48 ± 0.21	2.51 ± 0.19
MS052	2.26 ± 0.17	1.88 ± 0.16
VP131	2.43 ± 0.19	1.93 ± 0.16
VP427	2.92 ± 0.22	1.60 ± 0.14

Observations

AS009 consistently yields highest MOS scores across all models and strategies, while *MS052*, *VP131* consistently perform poorly.

Qualitative analysis revealed noticeable muffling and reverberation in *MS052*, *VP131* recordings, pointing to data quality issues.

Conclusions

- ① For Lithuanian multi-speaker TTS, within the tested range, the specific distribution of speakers does not meaningfully affect quality if the total budget is fixed.
- ② 7.5 minutes per speaker is sufficient for convergence of Tacotron 2 and Glow-TTS.
- ③ Tacotron 2 outperforms Glow-TTS for naturalness in Lithuanian, likely due to better prosody/pitch modeling.
- ④ Audio quality (reverberation, muffling) and/or speaker characteristics have a strong impact on synthesis naturalness.

Recommendations & Future Work

Recommendations:

- Prioritize maximizing **total data volume** and ensuring high per-speaker quality rather than optimizing the breadth-depth balance.
- Multi-speaker TTS models can be successfully trained from scratch using sparse data with 7.5 min/speaker.

Future Work:

- Investigate more architectures (e.g., VITS, FastSpeech 2).
- Test the “breadth versus depth” trade-off at higher budgets (e.g., 100+ hours).
- Implement automated data quality filtering to remove muffled/reverberant speakers.

Thank You!

Thank you for your attention!

Appendix 1: Reviewer's Questions

- **Q1:** What is the purpose of a pre-processing step audio resampling process?
- **Q2:** Where were the model parameters obtained or how were they selected? Were they optimal, suboptimal, or random?
- **Q3:** Why the explored models (Tacotron 2 and Glow-TTS) were based on different loss function? Are these models (with different loss functions) comparable?

Appendix 2: Loss Functions

Tacotron 2 optimizes Multi-component Regression Loss:

$$\mathcal{L}_{T2} = \mathcal{L}_{Dec(L2+SSIM)} + \mathcal{L}_{Post(L2+SSIM)} + \lambda_{attn}\mathcal{L}_{Guided} + \lambda_{stop}\mathcal{L}_{Stop} \quad (1)$$

Glow-TTS optimizes Negative Log-Likelihood:

$$\mathcal{L}_{Glow} = - \sum_{j=1}^{T_{mel}} \left[\log \mathcal{N}(z_j; \mu, \sigma) + \log \left| \det \frac{\partial z_j}{\partial x_j} \right| \right] \quad (2)$$

Appendix 3: Latin Square Design Example

Listener group (ID)	Presentation order (sequence)			
	Order 1	Order 2	Order 3	Order 4
Group 1	Model A Sentence 1	Model B Sentence 2	Model C Sentence 3	Model D Sentence 4
Group 2	Model B Sentence 3	Model A Sentence 4	Model D Sentence 1	Model C Sentence 2
Group 3	Model C Sentence 4	Model D Sentence 3	Model A Sentence 2	Model B Sentence 1
Group 4	Model D Sentence 2	Model C Sentence 1	Model B Sentence 4	Model A Sentence 3

