Thesis Structure

1. Introduction

- Background on TinyML and embedded AI
- The growing demand for on-device speech intelligence
- Challenges of deploying Python-trained models on ultra-low-power microcontrollers
- Motivation for a pure C implementation
- Contributions of this work:
 - o A practical conversion workflow
 - Quantization and implementation strategy
 - o Evaluation on a real embedded platform?

2. Related Work

- Brief review of:
 - o TinyML toolchains (TFLite Micro, CMSIS-NN, etc.)
 - o Previous efforts in low-power speech inference
 - Manual(framework-free, resource-minimal solution) vs framework-based deployment methods

3. Model and Task Description

- Overview of the speech task
- Model architecture
- Dataset used for training and evaluation

4. Conversion Workflow from Python to C

4.1. Model Training and Export

- Trained PyTorch model
- Exporting weights and bias from .pth to .h file
- Weight quantization (float $32 \rightarrow Q7.8$ or Q3.12)

4.2. C Inference Framework Design

- Data structures in C to represent layers and weights
- Fixed-point arithmetic choices (Q7.8, Q3.12, etc.)
- Forward-pass implementation for different layers

5. Experimental Setup

- Target platforms
- Evaluation metrics:

- Accuracy
- o Flash / RAM usage
- Inference latency
- o Power consumption

6. Results and Analysis

6.1. Accuracy Evaluation

- Quantized model vs original floating point model
- Comparison across quantization schemes (Q7.8 vs Q3.12)

6.2. Memory and Latency

- Inference latency on MCU (ms)
- Peak RAM and Flash usage

6.3. Error Sources and Trade-offs

- Sources of degradation (quantization, activations, fixed-point math)
- Trade-off: accuracy vs memory vs speed

7. Discussion

- Benefits and limitations of the manual C approach
- Use cases where this method is preferable over existing ML frameworks
- Reusability of the approach
- Potential for automation (e.g., scripts for weight export, model translation)

8. Conclusion

- Summary of key findings
- Practical implications for real-world embedded speech applications
- Future work:
 - o Loop unrolling / SIMD optimization
 - o Auto-code generation from Python