

ALMA MATER STUDIORUM · UNIVERSITÀ DI BOLOGNA

DEPARTMENT OF COMPUTER SCIENCE AND ENGINEERING
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Optimizing Small Language Models: An Experimental Investigation in Compressing Distilled LLaMA Architectures

Supervisor:
Prof. Francesco Conti

Presented by:
Angelo Galavotti

Co-supervisor:
Luca Bompani

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*Alla migliore madre del mondo,
al miglior padre del mondo,
e al miglior fratello del mondo.*

Abstract

Pending

Contents

| | | |
|----------|---|----------|
| 1 | Introduction | 1 |
| 1.1 | The Social Impact of LLMs | 2 |
| 1.2 | The Environmental Impact of LLMs | 3 |
| 1.3 | Scope | 4 |
| 1.3.1 | The Main Objective | 4 |
| 1.3.2 | How Can Optimization Techniques Help? | 6 |
| 1.4 | Document Structure | 7 |
| 2 | Background and Related Work | 9 |
| 2.1 | A Brief Overview of Early Language Models | 9 |
| 2.1.1 | Tokenization | 11 |
| 2.2 | The Transformer Architecture | 12 |
| 2.2.1 | The Attention Mechanism | 12 |
| 2.2.2 | Scaled Dot-Product Attention | 12 |
| 2.2.3 | Multi-Head Attention | 13 |
| 2.2.4 | Positional Encodings | 14 |
| 2.2.5 | Encoder-Decoder Transformers | 15 |
| 2.2.6 | Decoder-Only Transformers | 15 |
| 2.3 | Introducing the LLaMA Family of Models | 16 |
| 2.4 | A Look at the Target Hardware | 16 |
| 2.5 | Related Work | 19 |
| 2.5.1 | Knowledge Distillation | 19 |
| 2.5.2 | Pruning Approaches | 20 |

| | | |
|----------|--|-----------|
| 3 | Methodology and Implementation | 23 |
| 3.1 | Preliminary Research: Franken-LLaMA | 23 |
| 3.2 | An Overview of the Pipeline | 25 |
| 3.3 | Pruning techniques | 27 |
| 3.3.1 | Depth Pruning | 27 |
| 3.3.2 | Width Pruning | 29 |
| 3.4 | Low-Rank Adaptation (LoRA) | 31 |
| 3.5 | Quantization | 33 |
| 3.5.1 | Eigenspace Low-Rank Approximation (EoRA) | 35 |
| 3.6 | Evaluation | 37 |
| 3.6.1 | WikiText-2 Perplexity Evaluation | 37 |
| 3.6.2 | TriviaQA Question-Answering Evaluation | 38 |
| 4 | Experimental Results and Analysis | 41 |
| 4.1 | Preliminary observation of the results | 41 |
| 4.2 | Results on TriviaQA | 42 |
| 4.3 | Results on WikiText | 42 |
| 5 | Conclusion and Future Work | 43 |
| 5.1 | Future work | 43 |
| 5.2 | Final remarks | 43 |
| | References | 45 |

List of Figures

| | | |
|-----|--|----|
| 1.1 | Comparison of the energy consumption of various LLMs . . . | 5 |
| 2.1 | LSTM cell architecture | 11 |
| 2.2 | Transformer architectures comparison | 17 |
| 2.3 | The architecture of Siracusa | 19 |
| 3.1 | Comparison of Depth and Width Pruning | 28 |
| 3.2 | Advantages of Structured Sparsity | 30 |
| 3.3 | LoRA Overview | 31 |
| 3.4 | EoRA Overview | 36 |

Chapter 1

Introduction

It is no secret that, in the last three years, *Large Language Models* (LLMs) have fundamentally transformed our relationship with technology. Their impact rivals the most significant innovations of the past century, such as the internet and the smartphone. When people contemplate *Artificial Intelligence* (AI) today, they immediately think of ChatGPT or Claude, which have seamlessly integrated into our daily routines. Yet these powerful tools come with significant concerns. Their development and operation consume vast amounts of energy and water resource: modern data centers supporting these models require extensive cooling systems and electricity consumption that can rival small cities.

The computational complexity of these systems necessitates cloud-based deployment, which not only amplifies their environmental footprint but also fundamentally restricts user autonomy. This cloud dependency creates a concerning power dynamic where users have limited control over their tools, while simultaneously enabling extensive data collection practices and potential surveillance mechanisms that would be impossible with local, user-controlled alternatives.

In this more conversational opening chapter we will briefly examine the environmental and social impact of LLMs while highlighting the growing imperative for efficient, locally-deployable models that democratize access

without depleting our planet's resources. Afterwards, we will outline the scope of this project, which aims to explore the potential of compression techniques to make LLMs more efficient.

The future of AI depends not just on what these models can do, but how sustainably they can do it.

1.1 The Social Impact of LLMs

The widespread adoption of LLMs has created ripple effects across virtually every sector of society, fundamentally altering how we work, learn, and create. In education, these tools have sparked heated debates about academic integrity while simultaneously offering new possibilities for personalized learning and accessibility for students with disabilities [1]. The workplace has experienced perhaps the most dramatic shifts, with entire professions grappling with automation anxiety while others discover unprecedented productivity gains. Creative industries find themselves in particularly complex territory: writers, artists, and content creators must navigate between leveraging AI as a collaborative tool and protecting their intellectual property from being absorbed into training datasets without consent or compensation [2].

Perhaps what is most striking is how these models have democratized access to sophisticated capabilities that were once the exclusive domain of experts. A small business owner can now generate marketing copy that rivals professional agencies, students can receive tutoring in subjects where human expertise might be scarce, and non-programmers can write functional code with natural language instructions. Yet this democratization comes with a troubling caveat: it's entirely dependent on maintaining access to centralized, corporate-controlled systems. When OpenAI experiences an outage, millions of users worldwide suddenly lose access to tools they've integrated into their daily workflows. When pricing models change, entire business models built around AI assistance can become unsustainable overnight.

This dependency becomes even more concerning when we consider the data these systems collect. Every interaction, every query, every creative prompt may potentially become part of these companies' datasets, raising questions about privacy and intellectual property. As such, the need for transparency and user control over these systems has never been more urgent, and many believe that the future of AI must prioritize local, user-deployable models that empower individuals rather than centralizing power in the hands of a few corporations.

1.2 The Environmental Impact of LLMs

The computational demands of modern LLMs create an environmental footprint that grows exponentially with model size and usage. Training GPT-3, for instance, consumed an estimated 1,287 MWh of electricity, which is enough to power an average American home for over a century [3]. However, training represents only the tip of the iceberg; the real environmental cost lies in inference, where billions of daily queries across millions of users create a continuous drain on global energy resources. A recent study by Jegham et al. [5] estimates that OpenAI's GPT-4.5 requires 6.7 Wh of energy for a medium sized query (i.e. 100 input tokens, and outputting 300 tokens) to be processed. This figure grows to 20.5 Wh for a larger query (i.e. 1000 input tokens, and outputting 1000 tokens). To put this into perspective, this is equal to charging an average 40 Wh laptop battery to 50%. A graph comparing the energy consumption of different LLMs with different prompt sizes is shown in Figure 1.1.

Data centers housing these models consume approximately 1-2% of global electricity, a figure that's projected to reach 8% by 2030 if current trends continue [4]. The infrastructure supporting a single large-scale LLM requires thousands of high-performance GPUs running 24/7, each consuming as much power as several households.

Water consumption presents an equally pressing concern that receives far

less attention. Modern data centers require extensive cooling systems, with some facilities consuming millions of gallons daily. According to their sustainability report [6], Google’s water usage increased by 20% between 2021 and 2022, then by 17% from 2022 to 2023, and is largely attributed to AI inference operations [7]. In regions already facing water scarcity, this additional demand creates direct competition with human needs and agricultural requirements.

On the other hand, one has to keep in mind the embedded carbon cost of the hardware itself. Each GPU cluster supporting LLM operations represents significant emissions from manufacturing, shipping, and eventual disposal. The rapid pace of AI advancement drives frequent hardware upgrades, creating electronic waste streams that the recycling industry struggles to process effectively.

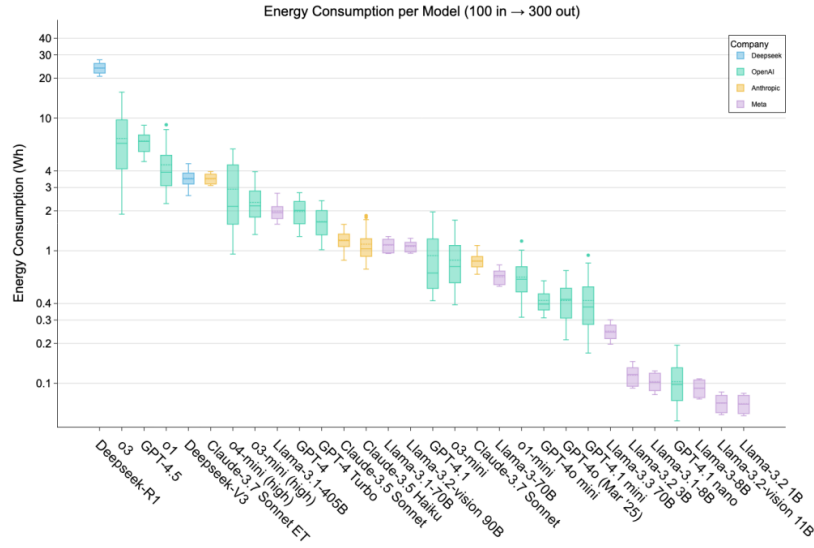
These environmental costs scale directly with model size and usage frequency, creating a fundamental tension between AI capabilities and sustainability.

1.3 Scope

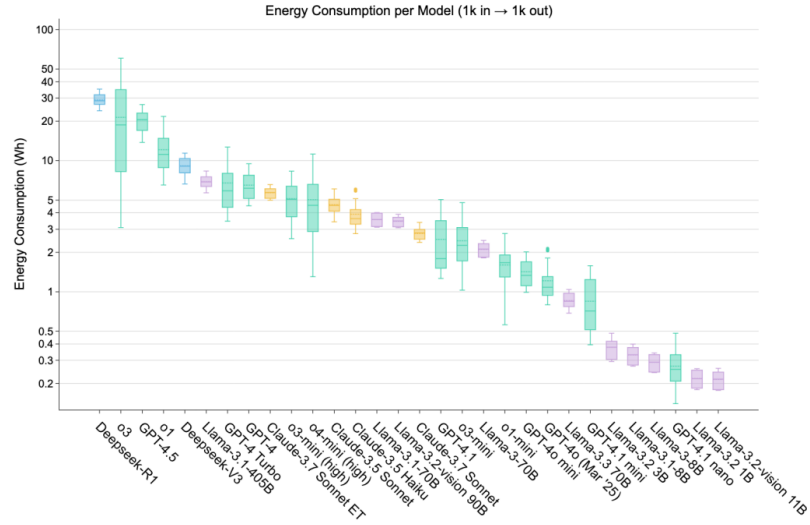
While we’ve examined several critical challenges facing LLMs, it’s worth emphasizing that these models hold significant potential for positive impact. As such, what was outlined in the previous sections points instead toward an urgent need for alternatives to the current paradigm of massive, centralized LLMs. A promising solution lies in compression techniques that can dramatically reduce model size while preserving core functionality. Through compression, billion-parameter server-sized models can be scaled down to more manageable ones that run much more efficiently.

1.3.1 The Main Objective

Given the context, the objective of this project is to push the boundaries of memory efficiency for small-scale LLMs by applying a targeted suite of



(a) Energy consumption for a small query (100 input tokens, 300 output tokens).



(b) Energy consumption for a large query (1000 input tokens, 1000 output tokens).

Figure 1.1: Comparison of the energy consumption of different LLMs for small and large queries. The data highlights the significant increase in energy requirements as query size grows. These graphs have been sourced from [5].

compression techniques. We begin with a distilled model, a 1B parameter variant of LLaMA 3 [28], which already represents a significantly reduced footprint compared to full-scale LLMs. From there, we explore and implement further optimizations, including depth and width pruning, *Low-Rank Adaptation* (LoRA), and quantization.

Pruning allows us to remove redundant layers or neurons from the network architecture, trimming excess capacity without substantial loss of capability. Quantization reduces the bit-width of model weights and activations, decreasing both memory usage and compute requirements. LoRA, meanwhile, introduces a lightweight, parameter-efficient training mechanism that reduces the cost of fine-tuning and adaptation without (necessarily) modifying the base model weights. In this way, the model can be adapted to new tasks or domains with minimal overhead.

The reason behind the focus on maximizing memory efficiency is related to the fact that memory represents the most expensive constraint in our target deployment scenario, where the intended hardware platform consists of a low-power RISC-V SoC with severely constrained memory resources (further details on the target hardware can be found in Section 2.4). Crucially, optimizations that reduce memory footprint also translate directly into computational complexity improvements, creating a dual benefit for resource-constrained environments.

1.3.2 How Can Optimization Techniques Help?

Optimization techniques tackle the environmental, social, and infrastructure problems that come with large-scale LLMs. When models run more efficiently, they need far less power for each query. Smaller models can actually run on edge devices and other energy-efficient hardware, cutting down on electricity usage substantially. This efficiency boost also extends the useful life of older devices: hardware that might otherwise be considered obsolete can suddenly run modern LLMs, which helps reduce electronic waste and makes better use of existing resources.

There's also the autonomy angle. Compact models that run locally allows users to be independent from centralized servers when using AI. People can run LLMs right on their own devices without any internet connection, which means no data gets sent to third parties and there's no risk of surveillance. This approach supports decentralization and opens up AI access to areas with poor connectivity or limited economic resources.

In other words, optimization is a pathway toward environmentally sustainable, privacy-respecting, and widely accessible AI.

1.4 Document Structure

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Chapter 2

Background and Related Work

Understanding the methodology behind this thesis requires first examining how Transformer architectures and Language Models have evolved. This chapter explores that evolution and briefly introduces the LLaMA family as it will be relevant to this project (see Chapter 3). The discussion then turns to the hardware limitations which influenced design decisions in this context. Finally, the chapter reviews research and literature on LLM optimization techniques relevant to this work.

2.1 A Brief Overview of Early Language Models

Before the emergence of the Transformer architecture, language models predominantly relied on *Recurrent Neural Networks* (RNNs). These networks represent an evolution of the *Multi-Layer Perceptron* (MLP), and incorporate cyclic connections within their architecture to create recurrent circuits. Such design enables the model to maintain a form of memory, allowing it to consider previous inputs when generating predictions and thereby capturing long-range dependencies inherent in sequential data such as text.

Despite their theoretical advantages, RNNs suffer from a critical limitation known as the vanishing gradient problem [10]. During backpropagation,

gradients from earlier time steps decay exponentially as they flow backward through the network layers. This degradation severely hampers the network’s ability to be trained effectively, as the influence of distant past information becomes negligible in the parameter updates.

To address these limitations, Hochreiter and Schmidhuber introduced *Long Short-Term Memory* (LSTM) networks [11] [12], which revolutionized sequence modeling through their sophisticated gating mechanisms. LSTMs employ specialized memory cells designed to selectively retain, update, and output information across extended sequences. The architecture incorporates three fundamental gate types that regulate information flow:

- The *input gate* determines the extent to which new information from the current input should be incorporated into the cell state. It evaluates the relevance of incoming data and controls its integration with existing memory.
- The *forget gate* governs the retention of information from previous time steps, deciding which aspects of the historical cell state remain relevant and should be preserved.
- Finally, the *output gate* regulates how much of the current cell state should be exposed to subsequent layers, effectively controlling what information propagates forward in the network.

All of these gates can be clearly observed in Figure 2.1. This gating mechanism allows LSTMs to maintain gradient flow over much longer sequences compared to vanilla RNNs. Thus, they can capture dependencies spanning hundreds of time steps, making them particularly effective for tasks requiring long-term context understanding such as language modeling, machine translation, and text summarization.

While LSTMs significantly improved the ability to model sequential data, they still faced challenges in terms of parallelization and context understanding, especially when compared to newer architectures such as Transformers

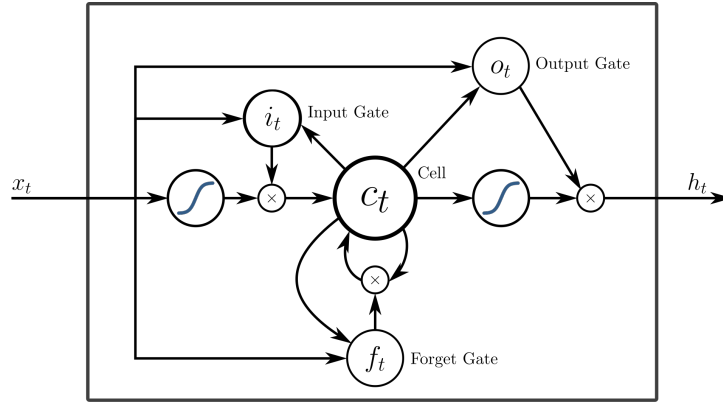


Figure 2.1: Architecture of an LSTM cell showing the three gating mechanisms. The input gate (left) controls information incorporation, the forget gate (center) manages memory retention, and the output gate (right) regulates information flow to subsequent layers. The cell state (horizontal line) maintains long-term memory throughout the sequence. The image was retrieved from [12].

(see Section 2.2). Nevertheless, LSTMs are still widely used for various *natural language processing* (NLP) tasks, including text generation [13].

2.1.1 Tokenization

Neural networks operate exclusively with numerical data, thus textual input needs to be converted into vector representations before processing. This is achieved through tokenization, which segments raw text into discrete units called tokens that are then mapped to high-dimensional vector embeddings.

The tokenization process typically involves breaking text into subword units rather than complete words, which enables models to handle both common vocabulary and previously unseen terms through compositional understanding. Modern language models predominantly employ algorithms such as *Byte Pair Encoding* (BPE) [14], which iteratively merges frequently occurring character pairs to create an optimal vocabulary. Each resulting token corresponds to a learned vector embedding that captures semantic and syntactic properties, enabling the neural network to process linguistic information through mathematical operations.

2.2 The Transformer Architecture

The Transformer architecture, introduced by Vaswani et al. [15], fundamentally changed how sequence modeling is approached by abandoning recurrent connections entirely in favor of the attention mechanism, allowing to capture long-range dependencies. Moreover, instead of that processing sequences step-by-step like LSTMs, the Transformer can examine all elements in a sequence simultaneously, and thus it can be easily parallelized during training.

2.2.1 The Attention Mechanism

Attention allows a model to dynamically focus on different parts of the input sequence when generating each output token, similar to how humans selectively concentrate on relevant information when reading a book. This is achieved by directly comparing and relating any two positions in a sequence, regardless of their distance (i.e. self-attention).

As such, the information bottleneck created when sequence models compress an entire input sequence into a single fixed-size vector is partially dealt with. This bottleneck becomes particularly problematic for long sequences, where important information can be lost or diluted.

2.2.2 Scaled Dot-Product Attention

The attention mechanism is computed using three matrices derived from the input:

$$\text{Attention}(Q, K, V) = \text{softmax}\left(\frac{QK^T}{\sqrt{d_k}}\right)V \quad (2.1)$$

Each component serves a specific purpose:

- **Queries (Q):** Generated by $Q = XW_Q$ where X is the input and $W_Q \in \mathbb{R}^{d_{\text{model}} \times d_k}$. Each query vector can be seen as asking “what information am I looking for?”

- **Keys (K):** Generated by $K = XW_K$ where $W_K \in \mathbb{R}^{d_{\text{model}} \times d_k}$. Keys act as indexed labels for each position's content.
- **Values (V):** Generated by $V = XW_V$ where $W_V \in \mathbb{R}^{d_{\text{model}} \times d_v}$. Values contain the actual information to be retrieved.

The attention computation proceeds in four steps:

1. **Similarity Computation:** QK^T produces a $n \times n$ attention matrix where entry (i, j) represents how much each token at position i is correlated to a token at position j . The dot product naturally measures similarity between query and key vectors.
2. **Scaling:** Division by $\sqrt{d_k}$ (where d_k is the dimensionality of the keys) prevents the dot products from becoming too large. Without scaling, large dot products push the softmax function into regions with extremely small gradients. For example, if $d_k = 512$, dot products could reach magnitudes of ± 20 or more, causing softmax to output distributions close to one-hot vectors.
3. **Normalization:** The softmax function converts similarity scores into a probability distribution: $\text{softmax}(x_i) = \frac{e^{x_i}}{\sum_{j=1}^n e^{x_j}}$. This ensures attention weights sum to 1 and creates a differentiable selection mechanism.
4. **Weighted Aggregation:** The attention weights are applied to the value vectors, producing a weighted sum that represents the attended information for each position.

2.2.3 Multi-Head Attention

Rather than applying attention once to the full representation, the Transformer divides the input representation (i.e. Q, K, V) into h blocks and applies the Scaled Dot Product Attention independently to each block. Each of these independent blocks is called an attention head, and they capture

different types of relationships simultaneously (some heads might track syntax, others semantics, positions, etc.). To achieve this, the input matrices Q, K, V are projected h times using different learned linear projections:

$$(QW_i^Q, KW_i^K, VW_i^V) \quad (2.2)$$

Given this, each head is computed as:

$$\text{head}_i = \text{Attention}(QW_i^Q, KW_i^K, VW_i^V) \quad (2.3)$$

As such, the final equation for multi-head attention is:

$$\text{MultiHead}(Q, K, V) = \text{Concat}(\text{head}_1, \dots, \text{head}_h)W^O \quad (2.4)$$

Each head uses different projection matrices $W_i^Q \in \mathbb{R}^{d_{\text{model}} \times d_k}$, $W_i^K \in \mathbb{R}^{d_{\text{model}} \times d_k}$, and $W_i^V \in \mathbb{R}^{d_{\text{model}} \times d_v}$, where typically $d_k = d_v = d_{\text{model}}/h$ for h heads.

Multi-head attention essentially gives the model multiple different perspective of the same input, and it is a vital system that contributes to make this architecture so powerful.

2.2.4 Positional Encodings

Since the attention mechanism computes similarity scores between every pair of tokens based solely on their content representations (i.e. is permutation invariant), it treats the input as an unordered set rather than a sequence. This means that rearranging the input tokens would produce identical attention weights, making it impossible for the model to distinguish between different orderings of the same words. To address this limitation, the Transformer requires explicit positional information.

The original work uses sinusoidal positional encodings:

$$PE_{(\text{pos}, 2i)} = \sin(\text{pos}/10000^{2i/d_{\text{model}}}) \quad (2.5)$$

$$PE_{(\text{pos}, 2i+1)} = \cos(\text{pos}/10000^{2i/d_{\text{model}}}) \quad (2.6)$$

These encodings create unique representations for each position that enable the model to learn the relative location of tokens through linear combinations of the sinusoidal functions.

2.2.5 Encoder-Decoder Transformers

Transformer-based models generate text by predicting the next token in a sequence, conditioned either on an external input (encoder-decoder models) or on the sequence generated so far (decoder-only models). The former is what the original Transformer architecture [15] follows.

- The **encoder** processes the full input sequence in parallel, producing a series of contextual embeddings that capture the meaning and structure of the input tokens.
- The **decoder** generates the output sequence token by token, using the embeddings from the encoder.

This design enables the model to generate outputs that are grounded in the input sequence, rather than generating new text from scratch. Thus, such design is well-suited for sequence-to-sequence tasks such as machine translation.

2.2.6 Decoder-Only Transformers

In decoder-only Transformers, there is no separate encoder module. The model predicts each token based only on the previously generated tokens, modeling the joint distribution $p(x_1, x_2, \dots, x_n)$ autoregressively. This means that the model outputs one token at a time, appends it to the input, and repeats until a stopping condition is met.

These architectures are simpler and more scalable, and they are the ones employed in large language models such as GPT [8] and, of course, LLaMA [16].

Decoder-only Transformers are ideal for open-ended text generation and other tasks where the output is not conditioned on a separate input sequence. Figure 2.2 shows a comparison of encoder-decoder and decoder-only transformer architectures.

2.3 Introducing the LLaMA Family of Models

The *Large Language Model Meta AI* (LLaMA) family [16], developed by Meta AI, comprises a series of transformer-based autoregressive language models designed as competitors to OpenAI’s GPT model family. The original LLaMA models were introduced in early 2023, followed by LLaMA 2 [17] later that year and LLaMA 3 [18] in 2024.

The models vary in the number of parameters and capabilities, with the newer versions offering improved performance over the older ones. In particular, with LLaMA 3, Meta introduced models ranging from 8B to 405B parameters with major upgrades in both training scale and performance over LLaMA 2, while approaching results to OpenAI’s GPT-4 [9] [18].

The main feature of LLaMA, however, is its open-weight release strategy. Unlike most proprietary models, Meta provides access to the model weights under a research-friendly license, enabling the broader research community and industry practitioners to experiment with, fine-tune, and deploy large-scale language models without relying on closed systems. This openness has led to a proliferation of derivative models, and thus to the making of the project central to this thesis.

2.4 A Look at the Target Hardware

The hardware constraints of an ideal target deployment scenario provide an additional compelling motivation for this compression work, beyond the environmental and social concerns outlined in Chapter 1. The envisioned

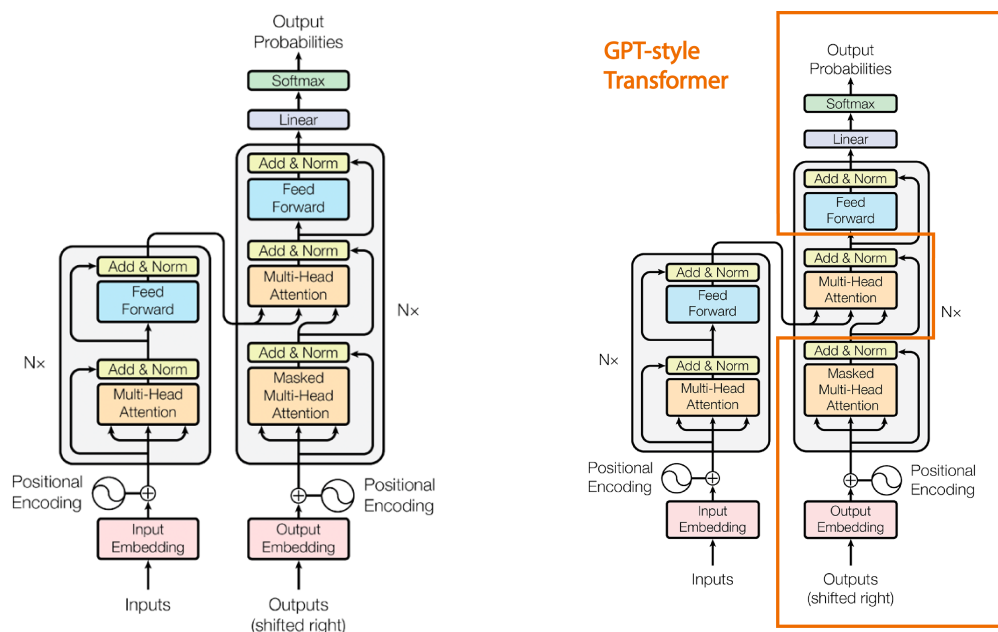


Figure 2.2: On the left, a standard Transformer architecture, which consists of an encoder and a decoder. On the right, we can observe the same architecture, while highlighting the only the layers used by a GPT-style model, which is a decoder-only transformer. The main difference is that the decoder does not attend to the encoder's output, allowing for auto-regressive generation. The image is a modified version of the one found on Vaswani et al. research paper [15].

platform is Siracusa [19], a heterogeneous RISC-V *System-on-Chip* (SoC) fabricated in 16 nm CMOS technology, which exemplifies the stringent resource constraints typical of edge XR devices designed for *Extended Reality* (XR) applications. While this project does not involve actual deployment or testing on this specific hardware, the severe memory limitations and computational constraints characteristic of such platforms make LLM compression not merely an environmental imperative, but a fundamental requirement for any practical edge deployment. A visual representation of Siracusa’s architecture can be found in Figure 2.3.

Siracusa’s memory hierarchy presents the primary constraint for the optimization efforts. The system features a three-tier memory organization: 256 KiB of L1 *Tightly-Coupled Data Memory* (TCDM) SRAM organized into 16 word-interleaved banks, 4 MiB of SRAM tile memory, and 4 MiB of *magnetoresistive memory* (MRAM) for weight storage, totaling 8.5 MiB. The memory hierarchy uses specialized allocation: MRAM stores weights while tile memory handles activations. However, it may be obvious to the reader that this amount of storage is rather limiting: exceeding the 4 MiB weight memory capacity forces reliance on slower off-chip transfers that can increase inference latency by orders of magnitude.

The computational architecture features an octa-core RISC-V cluster paired with the N-EUREKA neural processing engine, which provides substantial parallel processing capability with peak performance reaching 1.95 TOp/s. However, this performance is only achievable when models fit entirely within the on-chip memory hierarchy. The 92 Gbit/s bandwidth between MRAM and the neural engine enables efficient weight streaming, but only when the model’s memory footprint aligns with the available storage capacity.

These specifications illustrate the type of resource constraints that drive the need for aggressive compression techniques in edge AI deployment scenarios, and the compression pipeline developed in this work represents a promising approach toward achieving the level of optimization required for

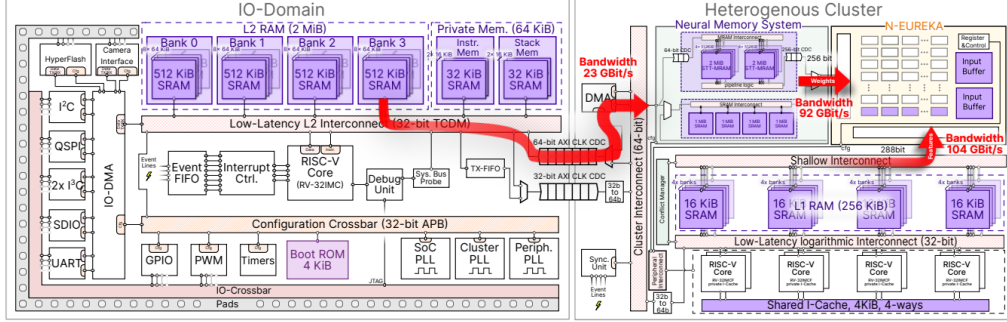


Figure 2.3: A visual representation of Siracusa’s architecture. On the upper right we can see the N-EUREKA accelerator, while on the lower right the RISC-V cluster is visible. This image was sourced from the original paper [19].

practical deployment on such platforms.

2.5 Related Work

This section examines some techniques that have proven effective for compressing large language models, providing an overview of the current available research and inspiration for novel methods that could extend the project’s scope (further explored in Section 5.2).

2.5.1 Knowledge Distillation

Knowledge distillation [20] has emerged as a fundamental technique for compressing pre-trained language models by transferring knowledge from a large “teacher” model to a smaller “student” model. In particular, the core principle involves training the student to match not only the ground truth labels but also the soft predictions of the teacher model, in order to capture richer information about the decision boundaries learned by the larger network.

Traditional distillation approaches can be categorized into task-specific and task-agnostic variants. Task-specific distillation first fine-tunes the teacher

model on a particular downstream task before distillation, while task-agnostic distillation maintains a general-purpose student that can be efficiently adapted to various tasks.

A significant challenge in conventional distillation arises from the substantial capacity gap between teacher and student models. When the student model has significantly fewer parameters than the teacher, the prediction discrepancy can become too large for effective knowledge transfer, particularly over massive amounts of open-domain training data. This limitation has motivated more sophisticated initialization and training strategies.

Homotopic Distillation (HomoDistil) [21] addresses this challenge through an innovative approach that combines iterative pruning with distillation. Rather than initializing a small student model from scratch, HomoDistil begins with the full teacher model and gradually prunes neurons while simultaneously distilling knowledge. This maintains a small prediction discrepancy throughout training, ensuring effective knowledge transfer even at high compression ratios.

The key insight of HomoDistil is that starting from the teacher model and iteratively removing the least important components prevents the dramatic performance drops typically seen with aggressive compression. At each training iteration, the method computes importance scores for individual neurons and removes those with minimal impact on the loss function. This pruning strategy is very similar to one adopted in this document for removing whole layers (i.e. depth pruning), as shown in Section 3.3.1. This gradual reduction maintains network connectivity and allows the distillation loss to effectively guide the compression process.

2.5.2 Pruning Approaches

Pruning is an optimization technique that reduces model complexity by systematically removing network components. The approach operates through two primary strategies: width pruning, which removes individual neurons or layers within transformer blocks, and depth pruning, which elim-

inates entire submodules. Given its central importance to this work, pruning techniques are examined in detail in Section 3.3.

Nevertheless, it is worth reviewing relevant work in this area to position our contribution within the broader research landscape. *Two-Stage Structured Pruning* (2SSP) [22] combines width and depth pruning optimally using a two stage approach. This method applies width pruning to feed-forward networks in the first stage, followed by depth pruning of attention modules in the second stage.

For width-pruning, importance scores are computed based on the magnitude of neuron activations across calibration samples, which are then used to remove entire neurons. On the other hand, the second stage targets attention modules which are iteratively removed based on their impact on model perplexity. This coarser-grained approach proves effective because attention modules often exhibit high redundancy.

A critical innovation in 2SSP is the automatic balancing of sparsity between the two stages. The method uses an adaptive formula that considers the relative sizes of feed-forward and attention components to determine optimal allocation of parameter reduction across stages. This ensures that neither component becomes a bottleneck while maintaining overall performance.

While some similarities can be found, the work of this thesis differs from the pruning approach adopted in 2SSP in two fundamental ways.

- The width-pruning targets differ significantly: while 2SSP focuses on FFN components, the approach presented in this thesis targets all attention weight matrices. Additionally, the structured width-pruning strategies are entirely distinct: 2SSP removes complete rows and columns by slicing weight matrices rather than using ratio-based structural pruning (see Section 3.3.2).
- The method for computing layer importance during depth-pruning employs a compound metric rather than relying solely on perplexity-based measures (see Section 3.3.1).

Another promising approach to structured pruning is demonstrated by *Sheared LLaMA* [23], which takes a fundamentally different perspective by viewing pruning as an initialization step rather than a final compression technique. This method employs targeted structured pruning to compress LLaMA2-7B down to 1.3B and 2.7B parameter models, followed by substantial continued pre-training to recover performance.

In particular, Sheared LLaMA combines two techniques: first, a constrained optimization approach that prunes models to precisely match target architectures (derived from existing well-optimized models), and second, dynamic batch loading that adjusts training data composition based on domain-specific loss recovery rates. This addresses a critical observation that pruned models retain different levels of knowledge across domains (e.g. maintaining better performance on code repositories while losing more capability on natural language text).

Chapter 3

Methodology and Implementation

Having covered the importance of LLM optimization and the technical architecture of transformers in previous chapters, we now turn to the practical challenge of model compression.

The following sections begin with preliminary experiments on layer manipulation that revealed crucial insights about architectural redundancy in transformer models and shaped the main compression strategy. It then presents what is possibly the heart of this research: a multi-stage pipeline that combines several optimization techniques in a structured sequence. Each stage is examined with both theoretical foundations and key implementation considerations. Finally, the evaluation framework is outlined along with the chosen metrics and benchmarks.

3.1 Preliminary Research: Franken-LLaMA

Before embarking on systematic compression techniques for our target 1B parameter LLaMA model, preliminary research was conducted to understand the behavior of transformer architectures under structural changes. This exploratory phase consisted on the “Franken-LLaMA” project [24],

which involved experimenting with architectural modifications on the larger LLaMA2-7B-Chat model [17] to gain a first insight into which components of the transformer architecture are most critical for maintaining model performance.

The approach centered on modifying the standard transformer execution flow by selectively including, excluding, or repeating attention blocks within the 32-layer architecture. Implementation was carried out using PyTorch [25], with modifications to Hugging Face’s transformers library [26] to enable dynamic layer skipping and repetition at runtime. The repetition strategy was particularly attractive as it could theoretically reduce memory footprint by reusing the same layer weights multiple times rather than storing distinct parameters for each position. This weight sharing approach aligned directly with the target hardware constraints outlined in Section 2.4.

25 different layer configurations were tested, each of which was evaluated through qualitative text generation tasks and quantitative assessment on the HellaSwag dataset [27]. The results revealed that conservative modifications often maintained reasonable performance while reducing computational overhead. In particular, skipping layers towards the middle-end of the model rather than the beginning or final layers resulted in lower output quality degradation. For instance, the configuration that skipped layers 23-27 achieved a HellaSwag score of 0.38 compared to the baseline’s 0.34 (the baseline consisted on the unmodified LLaMA 2 model), suggesting that certain middle layers may contribute less to final performance than expected.

However, more aggressive modifications typically led to a much more severe degradation in output quality. Configurations involving extensive layer repetition or using only sparse layer selections often produced incoherent text with non-ASCII characters and semantic breakdown. This behavior indicated that while some redundancy exists in the transformer architecture, there are clear limits to how extensively the model can be modified before fundamental language capabilities deteriorate.

These preliminary findings informed the subsequent approach to system-

atic compression, as they revealed that strategic layer removal could maintain acceptable quality levels while reducing computational overhead, suggesting that pruning techniques might offer promising avenues for optimization. At the same time, Franken-LLaMA proved that layer repetition was not a viable strategy and caused heavy performance degradation.

3.2 An Overview of the Pipeline

The core result of this research is the compression pipeline, which implements a sequential approach that combines multiple optimization techniques in a carefully orchestrated manner. As previously mentioned, the target model for these optimizations is LLaMA 3.2 1B, a distilled version of the larger LLaMA 3.2 model that has already undergone some level of pruning during its creation, according to the model documentation on Hugging Face [28].

The choice of this particular model as the starting point is both strategic and practical. While incorporating distillation directly into the pipeline would have been ideal given its effectiveness as a compression technique [21], the computational requirements make it rather prohibitive: in terms of complexity, distillation is essentially equivalent to training a model from scratch, thus requiring extensive GPU resources that far exceed the computational budget available for the project. In light of this, an existing pre-distilled model was utilized instead which already provides a rather compact foundation.

Following the approach established in the preliminary experiments (Section 3.1), the pipeline is implemented using PyTorch [25] for model modifications and Hugging Face’s transformers library [26] for model loading and inference. The majority of configurations were developed using the Instruct version of LLaMA 3.2 1B [29], as these instruction-tuned models are specifically optimized for dialogue and question-answering tasks that constitute the primary evaluation benchmarks in this work (see Section 3.6),

thereby minimizing the need for additional prompt engineering during testing phases. Nevertheless, the compatibility with the original model is kept, and the framework can produce configurations built upon the vanilla LLaMA 3.2 1B variant [28] by specifying a flag before execution.

The compression pipeline follows a five-stage progression, with each stage building upon the previous one. The particular ordering was chosen based on the complementary nature of these techniques and their relative impact on model structure.

1. The first stage implements depth-wise pruning (Section 3.3.1), where entire transformer layers are removed based on importance metrics computed during a calibration phase. This coarse-grained approach eliminates redundant blocks, and ultimately redefines the skeleton of the architecture.
2. Width-wise pruning follows as the second stage (Section 3.3.2), applying the WANDA algorithm [31] to remove less critical weights within the remaining layers. This is a finer-grained approach compared to the depth-pruning, as it operates at the parameter level.
3. The third stage introduces *Low-Rank Adaptation* (LoRA) [33] (Section 3.4) to recover performance lost during the pruning phases, while also allowing for fine-tuning on specific downstream tasks.
4. The fourth stage applies 4-bit quantization using GPTQ [36] (Section 3.5), reducing the memory footprint of individual parameters.
5. The final stage includes an optional *Eigenspace Low-Rank Approximation* (EoRA) [40] (Section 3.5.1) step to improve the performance of the quantized model.

Detailed descriptions of each stage are provided in the following sections. Each step yields an intermediate model suitable for independent evaluation, except when explicitly stated otherwise. Additionally, the pipeline features a

comprehensive set of tuning options, such as the ability to skip specific stages or experiment with different parameter configurations without modifying the core implementation. This allows to conduct ablation studies more easily and adapt the framework to specific hardware constraints.

3.3 Pruning techniques

Pruning reduces neural network complexity by removing components according to importance criteria. The technique operates through two distinct main approaches that differ in their granularity and organization:

- **Unstructured pruning** eliminates individual weights throughout the network randomly or based on an importance criteria such as magnitude.
- **Structured pruning** removes entire architectural units such as groups of neurons, attention heads, or complete layers.

Additionally, they can be further categorized based on the spatial dimension of the operation. In particular:

- **Width pruning** targets components within layers, such as individual neurons inside an attention block.
- **Depth pruning** eliminates entire layers or transformer blocks from the architecture. It is generally regarded as a form of structured pruning.

A visual comparison between these two pruning techniques is shown in Figure 3.1.

3.3.1 Depth Pruning

Regarding depth pruning, the approach used in this project eliminates entire attention blocks following the methodology established by Kim et al. in their Shortened LLaMA work [30], in which they define three metrics to determine the importance of a transformer block:

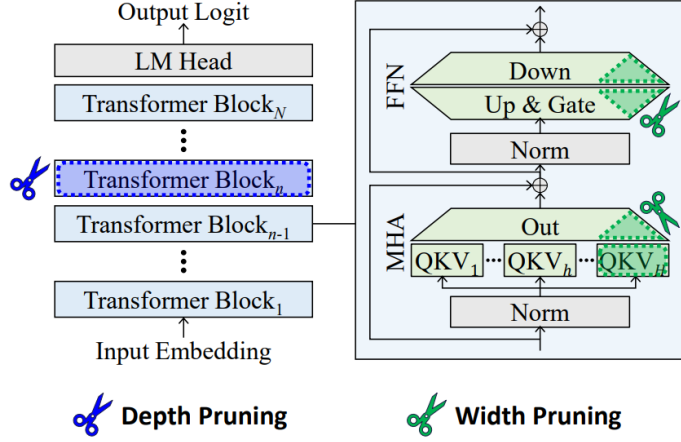


Figure 3.1: In depth pruning, we remove the single transformer layers (left). On the other hand, in width pruning we remove single neurons from the weight matrices (right). The image was retrieved from [30].

- **Magnitude-based importance:** computes L1 norms of weights within each block, assuming that layers with larger weight magnitudes contribute more significantly to model output.
- **Taylor expansion importance:** leverages gradient information to estimate removal impact through the approximation

$$L(W = 0) \approx L(W) + \nabla L \cdot (-W) \quad (3.1)$$

where the gradient-weight product indicates layer significance based on how loss would change if the layer were removed.

- **Perplexity-based importance:** temporarily removes each layer and measures performance degradation on calibration data through perplexity [42], providing a direct empirical assessment of layer contribution to model quality (details on this metric will be further explained in Section 3.6).

While the pipeline implementation incorporates all three importance ranking methods, in practice, perplexity served as the main importance metric

during testing due to its direct correlation with model performance degradation. In view of this, all layers are ranked according to their computed importance scores, with layers receiving lower rankings considered less critical and prioritized for removal during the pruning process.

Following insights from prior work in Section 3.1 and Shortened-LLaMA [30], the implementation protects the first four and last two layers from removal, as these positions prove critical for maintaining performance.

The pruning implementation creates a new model architecture with reduced depth by systematically copying weights from the original model while skipping the least important layers. The process begins by modifying the model configuration to reflect the reduced number of layers, then establishes a mapping between original and pruned layer indices. This mapping accounts for the gaps created by removed layers, ensuring that retained layers maintain their relative positioning and connectivity.

The resulting pruned model maintains the same computational pattern as the original but with fewer sequential operations, directly reducing inference latency and, most importantly, memory requirements.

3.3.2 Width Pruning

The width pruning approach in this pipeline exclusively employs structured pruning techniques to maintain compatibility with modern hardware accelerators. Specifically, the implementation targets structured sparsity patterns supported by NVIDIA’s sparse Tensor Cores in Ampere and Hopper architectures [32]. This sparsity pattern enables theoretical 2x compute throughput compared to dense matrix multiplication while maintaining efficient memory access. In particular, unlike unstructured pruning that creates irregular sparse matrices requiring specialized computation libraries, structured sparsity preserves regular tensor shapes that map directly to accelerator capabilities without additional software overhead. A visual representation of the advantage is shown in Figure 3.2.

Weight And activation-based pruning (WANDA) [31] serves as the core

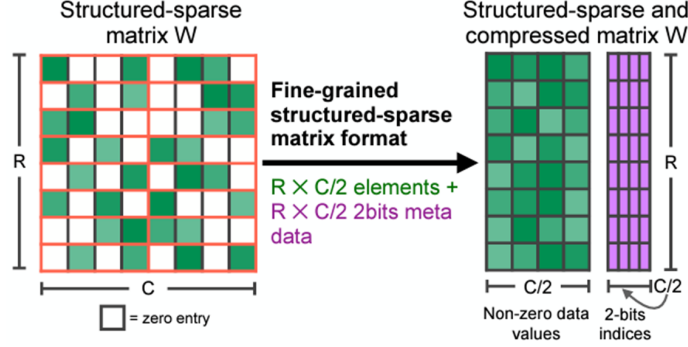


Figure 3.2: Visual representation of how the structured-sparse matrices are processed in NVIDIA’s Ampere and Hopper GPUs. The image was sourced from [32].

algorithm for implementing this structured approach. Originally developed for unstructured weight removal, WANDA’s official implementation includes extensions for structured pruning patterns that align with the hardware requirements. The method leverages the observation that large language models develop outlier activations, which are emergent features with magnitudes significantly larger than typical hidden state values and are crucial for performance. Consequently, the WANDA importance metric for each weight W_{ij} is defined as

$$S_{ij} = |W_{ij}| \cdot \|X_j\|_2 \quad (3.2)$$

where $|W_{ij}|$ represents the absolute weight magnitude and $\|X_j\|_2$ evaluates the L2 norm of the j -th input feature across calibration samples. This formulation addresses a key limitation of traditional magnitude-based importance, which is typically used by width pruning approaches, by accounting for input activations, which play an equally important role in determining neuron outputs.

The structured pruning process each layer by organizing weights into groups of M consecutive elements (i.e. consecutive columns of the same row of the weight matrix) and removing the N with lowest importance scores while retaining the rest. To achieve this, forward hooks collect activation statistics during calibration, with each linear layer wrapped to compute the

L2 norm of input activations across samples. After computing importance scores for each weight group, the implementation uses PyTorch’s scatter operations to efficiently mark low-importance weights for removal. The algorithm completes pruning in a single forward pass using the C4 corpus [39] as calibration dataset.

3.4 Low-Rank Adaptation (LoRA)

Low-Rank Adaptation (LoRA) [33] represents a parameter-efficient fine-tuning technique that addresses the computational and memory constraints associated with full model retraining. Rather than updating all model parameters during adaptation, LoRA introduces trainable low-rank decomposition matrices that capture the essential changes needed for task-specific performance. Figure 3.3 provides an overview of this process.

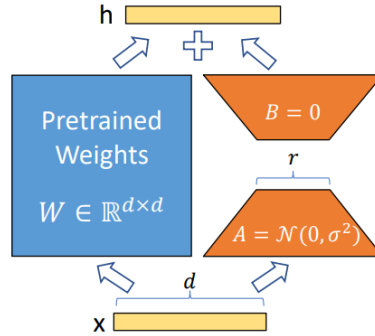


Figure 3.3: A summary of the LoRA process. We can see how the original pre-trained weights and the low-rank adapter are summed when the output activation is computed. The image was retrieved from the original LoRA paper [33].

The fundamental insight behind LoRA stems from the hypothesis that weight updates during adaptation have a low intrinsic rank, even when the full rank of the weight matrices is very high. For a pre-trained weight matrix $W_0 \in \mathbb{R}^{d \times k}$, LoRA constrains the update by representing it through a low-rank decomposition:

$$W_0 + \Delta W = W_0 + BA \quad (3.3)$$

where $B \in \mathbb{R}^{d \times r}$, $A \in \mathbb{R}^{r \times k}$, and the rank $r \ll \min(d, k)$. During training, W_0 remains frozen while only A and B contain trainable parameters.

The modified forward pass becomes:

$$h = W_0 X + \Delta W X = W_0 X + BAX \quad (3.4)$$

The matrices are initialized such that A follows a random Gaussian distribution while B is initialized to zero, ensuring that $\Delta W = BA$ starts at zero so that the model begins with its original pre-trained behavior.

LoRA offers substantial reductions in trainable parameters while maintaining comparable performance to full fine-tuning. Its linear design allows to merge the low-rank matrices with frozen weights during deployment, introducing no additional inference latency. According to the original paper [33], in Transformer architectures LoRA is most effective when applied to attention weights, particularly the query and value projection matrices (W_q and W_v), with very low ranks ($r = 1$ or $r = 2$) often proving sufficient for downstream tasks.

Within the compression pipeline, LoRA serves as the third stage, following the pruning phases. This positioning is strategic: LoRA helps recover performance degraded by the aggressive structural modifications while simultaneously adapting the model for specific downstream tasks.

The implementation utilizes Hugging Face’s *Parameter-Efficient Fine-Tuning* (PEFT) library [34], which provides optimized LoRA integration with Hugging Face transformer models. The configuration targets the attention projection matrices with a rank of 8 and dropout rate of 0.1. Training employs mixed-precision optimization with gradient accumulation to maintain computational efficiency while adapting to the reduced parameter space.

3.5 Quantization

Quantization represents a fundamental compression technique that maps continuous floating-point values to a discrete, finite set of representations, typically using low precision formats such as 8-bit or 4-bit integers. This approach is widely used on architectures such as Convolutional Neural Networks [35], as these models are often deployed on edge devices and quantization offers a substantial memory reduction with controlled impact performance.

The quantization process follows the mapping:

$$\text{quant}(x) = \text{round} \left(\frac{x - z}{s} \right) \quad (3.5)$$

where x represents the original floating-point value, s denotes the scale factor and z is the zero-point offset. Subtracting the offset and dividing by the scale normalizes the floating-point value to the quantization grid, so that it is accurately represented within the discrete range of the target precision format.

However, this straightforward round-to-nearest approach often results in significant accuracy degradation. The GPTQ technique proposed by Frantar et al. [36] addresses these limitations through a sophisticated post-training quantization algorithm that leverages second-order information from the Hessian matrix of the layer’s reconstruction error to minimize quantization impact. Building upon the *Optimal Brain Quantization* (OBQ) framework [37], GPTQ formulates quantization as a layer-wise optimization problem where for each linear layer with weight matrix W and calibration inputs X , the algorithm seeks quantized weights \hat{W} that minimize the reconstruction error:

$$\arg \min_{\hat{W}} ||WX - \hat{W}X||_2^2 \quad (3.6)$$

The core innovation of GPTQ lies in its three-step algorithmic approach that enables scalability to billion-parameter models.

The first step is based on the idea that arbitrary quantization order suffices. Unlike OBQ, which greedily selects weights based on quantization

error, GPTQ demonstrates that quantizing weights in any fixed order yields comparable results for large, heavily-parametrized layers. This insight allows all rows of the weight matrix to be quantized in the same column order, reducing the overall computational complexity.

The second step consists on the implementation of lazy batch updates. To address memory bandwidth bottlenecks, GPTQ processes weights in blocks of $B = 128$ columns simultaneously. The algorithm exploits the observation that quantization decisions for column i are only affected by updates to that specific column, enabling “lazy batching” where updates are accumulated within blocks before global application:

$$\delta_F = -(w_Q - \text{quant}(w_Q))([H_F^{-1}]_{QQ})^{-1}(H_F^{-1})_{:,Q} \quad (3.7)$$

$$H_{-Q}^{-1} = [H^{-1} - H_{:,Q}^{-1}([H^{-1}]_{QQ})^{-1}H_{Q,:}^{-1}]_{-Q} \quad (3.8)$$

where Q denotes the set of quantized weight indices within the current block, F represents remaining full-precision weights, δ_F is the compensation update applied to unquantized weights to minimize reconstruction error, and H_{-Q}^{-1} is the updated inverse Hessian matrix with rows and columns corresponding to quantized weights Q removed.

The final step is the Cholesky reformulation. For numerical stability at scale, GPTQ precomputes Hessian inverse information using Cholesky decomposition rather than iterative matrix updates. The Hessian matrix is computed as:

$$H = 2XX^T + \lambda I \quad (3.9)$$

where λ represents a damping factor (1% of average diagonal value) for numerical stability. The Cholesky decomposition $H^{-1} = LL^T$ provides numerically stable access to required matrix rows while avoiding the accumulation of errors from repeated matrix inversions that plagued the older approaches when applied to large models.

The quantization procedure processes each layer independently, iterating through weight columns while maintaining compensation updates to remain-

ing unquantized weights, and is applied to the attention matrices and linear layers of the Transformer.

The compression pipeline places quantization as the fourth stage, applied after pruning and LoRA adaptation phases, to strategically leverage the reduced parameter count from pruning. For the practical implementation, the GPTQModel toolkit [38] was employed, which provides optimized implementations of the GPTQ algorithm with support for various model architectures and quantization configurations. This toolkit handles the complex numerical computations and memory management required for large-scale quantization, allowing focus on the integration aspects within the broader compression pipeline. The quantization process utilizes 4-bit precision with a group size of 128, which means that groups of 128 consecutive weights share the same quantization parameters (scale and zero-point).

3.5.1 Eigenspace Low-Rank Approximation (EoRA)

Following quantization, Eigenspace Low-Rank Approximation (EoRA) [40] can be optionally applied to recover accuracy lost during compression. EoRA is a rather novel technique which provides a training-free method to enhance performance of compressed models through low-rank matrix compensation by projecting compression errors into task-specific eigenspaces. Figure 3.4 summarises the process in a schematic way.

Despite their similar names, EoRA differs fundamentally from Low-Rank Adaptation (LoRA). While LoRA adapts pre-trained models to new tasks through gradient-based optimization of low-rank matrices, EoRA focuses on compensating compression-induced errors without training. Nevertheless, both methods employ additive low-rank corrections to weight matrices which are formulated as adapters.

As noted earlier, EoRA projects compression error into the eigenspace of layer-wise input activations before applying singular value decomposition. For a layer with original weights W and compressed weights \hat{W} , compression error is $\Delta W = W - \hat{W}$. Rather than directly decomposing this error,

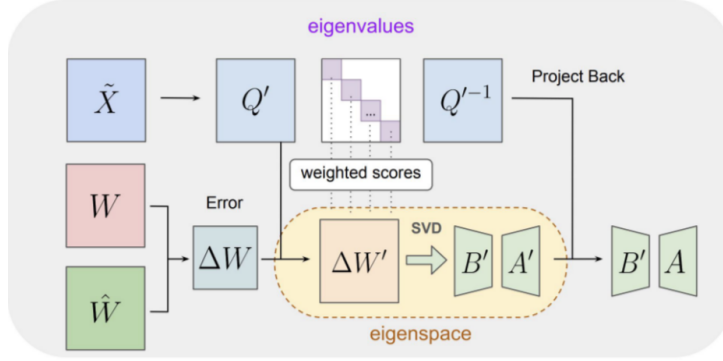


Figure 3.4: The image summarizes the EoRA process in a schematic way, courtesy of [40].

EoRA first computes the eigendecomposition $\tilde{X}\tilde{X}^T = Q\Lambda Q^T$, where \tilde{X} represents average input activations over the calibration dataset. This is done in order to derive the eigenspace projection matrix Q , whose columns are the eigenvectors, and Λ , which is a diagonal matrix containing the corresponding eigenvalues.

The projection of the compression error into the eigenspace is then computed using the transformation matrix $Q' = Q\sqrt{\Lambda}$:

$$\Delta W' = \Delta W \cdot Q' \quad (3.10)$$

This eigenspace projection amplifies compression errors along directions of high activation variance (i.e. high eigenvalues) while diminishing those along low-variance directions, causing the subsequent SVD operation to naturally allocate more of its limited representational capacity to approximating the amplified, task-relevant error components.

Finally, the projected error $\Delta W'$ is decomposed using SVD to obtain $\Delta W' \approx B'A'$, where B' and A' are low-rank matrices. The final compensation matrices are obtained by back-projecting to the original space: $A = A'(Q')^{-1}$.

Similarly to LoRA, the forward pass on the compensated model becomes:

$$h = \hat{W}X + B'A'(Q')^{-1}X = \hat{W}X + B'AX \quad (3.11)$$

In the default configuration of the pipeline, the EoRA calibration step

uses 64 samples from the C4 dataset [39]. This minimal calibration requirement is one of the main advantages of this method. Implementation once again leverages the GPTQModel toolkit [38], which integrates EoRA compensation with existing quantized models.

3.6 Evaluation

The evaluation framework employs two distinct benchmarking datasets to assess model performance across different dimensions of language understanding. WikiText-2 [41] serves as the primary benchmark for measuring language modeling capability through perplexity computation [42], while TriviaQA [43] evaluates both reading comprehension and factual knowledge through question-answering tasks.

The implementation leverages a custom evaluation framework which automatically detects model types and applies appropriate loading procedures.

3.6.1 WikiText-2 Perplexity Evaluation

WikiText-2 [41] represents a collection of high-quality Wikipedia articles that serves as a standard benchmark for language modeling tasks. The dataset consists of over 2 million tokens extracted from verified Wikipedia articles, providing a diverse corpus of well-structured text that spans multiple domains and writing styles.

Perplexity serves as the fundamental metric for language model evaluation, measuring how well a model predicts the next token in a sequence. Mathematically, perplexity is defined as the exponential of the cross-entropy loss:

$$\text{Perplexity} = \exp \left(\frac{1}{N} \sum_{i=1}^N -\log P(w_i | w_1, w_2, \dots, w_{i-1}) \right) \quad (3.12)$$

where N represents the total number of tokens, w_i denotes the i -th token, and $P(w_i | w_1, w_2, \dots, w_{i-1})$ represents the model's predicted probability for

the correct next token given the preceding context. Lower perplexity values indicate better language modeling performance, with the metric providing intuitive interpretation: a perplexity of X means the model is, on average, as confused as if it had to choose uniformly among X possibilities for each token.

The evaluation protocol processes the WikiText-2 test split by accumulating negative log-likelihoods across all valid tokens before computing the final exponential transformation. The implementation iterates through sequences, computing cross-entropy loss for each token while applying attention masks to exclude padding tokens from the calculation, ensuring that only meaningful content contributes to the perplexity measurement.

3.6.2 TriviaQA Question-Answering Evaluation

TriviaQA [43] constitutes a comprehensive reading comprehension dataset containing over 650,000 question-answer pairs sourced from trivia and quiz-bowl competitions. The dataset’s distinctive characteristic lies in its dual-mode evaluation framework that enables assessment of both knowledge retrieval and reading comprehension capabilities through closed-book and open-book testing scenarios.

The closed-book evaluation presents models with questions in isolation, requiring them to rely entirely on knowledge encoded in their parameters during training. This mode directly tests the factual knowledge retention of compressed models and reveals whether compression techniques have degraded the model’s ability to recall specific information. Questions use a simple prompt structure: “Question: [question text] Answer:”.

Conversely, the open-book evaluation provides relevant context passages alongside each question, which simulates a reading comprehension scenario where models must extract answers from provided text. The context is derived from TriviaQA’s associated search results, with the implementation selecting the shortest available context passage to minimize computational overhead while preserving essential information. Consequently, the prompt

format becomes: “Context: [context passage] Question: [question text] Answer:”.

Answer correctness evaluation employs a three-tier verification system designed to accommodate the variability inherent in natural language responses. The evaluation process begins with testing the exact matching of the answer, where the generated text is compared directly against all provided correct answer aliases after normalization. Normalization involves converting text to lowercase, removing punctuation, and standardizing whitespace to ensure fair comparison despite minor formatting differences.

When exact matching fails, the system applies substring matching, checking whether any correct answer appears as a contiguous substring within the generated response. This approach accounts for models that provide correct answers embedded within longer explanations or additional context, which is common behavior in instruction-tuned models.

The final verification stage implements fuzzy matching for cases where neither exact nor substring matching succeeds. For answers containing multiple words, the system computes the intersection between the set of words in the generated answer and each correct answer, considering a response correct if at least 80% of the words in the target answer appear in the generated text:

$$\text{Fuzzy Match} = \frac{|\text{Words}_{\text{generated}} \cap \text{Words}_{\text{correct}}|}{|\text{Words}_{\text{correct}}|} \geq 0.8 \quad (3.13)$$

Generation parameters are carefully controlled to ensure consistent evaluation conditions. Models generate between 1 and 50 tokens per answer using deterministic decoding (with temperature set to 1.0) to mitigate randomness that could affect reproducibility. The evaluation framework processes 1200 samples of TriviaQA’s validation split (alternatively, the quantity of samples can be tuned by the user), and the accuracy is computed separately for these two tasks.

Chapter 4

Experimental Results and Analysis

intro here

4.1 Preliminary observation of the results

Before examining the results based on In this section, we will present and comment on some examples of text generated by the

Before testing, a select group of configuration undergo LoRA fine-tuning using these two task-specific datasets. Then, the results were compared between baseline compressed models and their LoRA-enhanced counterparts isolate the effectiveness of parameter-efficient fine-tuning in recovering compression-induced degradation.

Additionally, cross-task evaluation examines the transferability of LoRA adaptations by testing TriviaQA-tuned models on WikiText-2 perplexity and WikiText-2-tuned models on TriviaQA accuracy. This cross-evaluation reveals the task-specificity of LoRA adaptations and their potential for broader performance improvements beyond their target domains.

4.2 Results on TriviaQA

4.3 Results on WikiText

Chapter 5

Conclusion and Future Work

5.1 Future work

- speak about Distillation
- speak about experimenting with other quantization methods
- experimenting with synergy between pruning methods
- speak about adding support for other llms
- kV cache compression
- QQQ compression algorithm

5.2 Final remarks

The objective of this project was to research and implement a methodology for compressing LLaMA based LLMs,

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Let's see what this "greatness" is all about.