NanoGPT Speedrun Living Worklog

How fast can I train GPT-2 on two RTX 4090 GPUs?

March 8, 2025

I've seen <u>some really awesome GPT-2</u> speedrun results from people like <u>Keller Jordan</u>, <u>Fern</u>, <u>Braden Koszarsky</u>, and others. I got a little inspired and wanted to see how fast I could train GPT-2 on my own hardware.

Technically, the NanoGPT speedrun is to train a neural network to 3.28 validation loss on FineWeb as fast as possible on an 8xH100 node. Keller Jordan maintains a leaderboard here. At the time of writing (Jan 16, 2025), the record is 3.14 minutes (!).

I have access to 2xRTX 4090 GPUs and I want to see how fast I can train GPT-2 on them by following the same rules as the NanoGPT speedrun. If I see some success, I may try to transfer my methods to an 8xH100 node for comparison with the main leaderboard.

I'll be documenting my progress here and updating this post as I go. Code can be found in this GitHub repo.

Progress so far

#	Description	Record time	Training Tokens	Tokens/Second	Date	Commit	Log
1	Initial baseline	8.13 hours	6.44B	221k	2025/01/16	b3c32f8	here
2.1	Architectural changes	7.51 hours	5.07B	188k	2025/01/18	<u>b7bb93f</u>	here
2.2	Muon optimizer	4.53 hours	3.04B	187k	2025/01/23	b91c2c0	here
2.3	Dataloading tweaks	4.26 hours	3.31B	216k	2025/02/18	<u>d59944d</u>	here
2.4	Logit Soft-capping at 30	4.01 hours	3.15B	218k	2025/02/23	12eab44	here
<u>3</u>	Longer Sequence Length	2.55 hours	1.88B	205k	2025/03/03	d982ed5	<u>here</u>

1. Initial setup and baseline

Part of the goal of this project is for me to learn as I go, so I am going to start at the beginning – with with Andrej Karpathy's <u>PyTorch GPT–2 trainer</u> from <u>llm.c</u>. This is the script that Keller Jordan used for <u>his initial baseline</u>. This trainer is very similar to the NanoGPT trainer with some minor modifications / simplifications (such as no dropout).

I have upstreamed some QOL improvements and basic tweaks to the training script from Keller's fork, but have not changed any of the core training / modeling logic. Specifically:

- 1. Implemented gradient accumulation so that my 2x24GB GPUs simulate the training experience of a 8xH100 machine.
- 2. Increased learning rate to 0.0015 and halved the batch size (total batch size is 262144 that is bs of 32/device * 2 devices * 1024 sequence length * 4 gradient accum steps).
- 3. Improved learning rate schedule (linear warmup then linear decay).
- 4. Removed all affine scale/bias parameters and switched to RMSNorm.
- 5. Padded the vocab size from 50257 to 50304 to make it a multiple of 128 (for better tensor core utilization).
- 6. Using Pytorch 2.5.1 (the switch from 2.4 to 2.5 gave ~9% speedup on the 8xH100 leaderboard).

Additionally, I added wandb logging for easy tracking of training progress - optimistically I may need to remove this one day as it slightly increases step time.

Commit with the initial setup is here: b3c32f8.

The baseline run time on my 2xRTX 4090 setup is 8.13 hours.

2. Implementing major improvements from the 8xH100 leaderboard

Waiting 8 hours for a result is too slow for effective experimentation, so I'm going to begin by implementing some of the notable improvements from the 8xH100 leaderboard.

I'll start with the most impactful/easiest changes first:

- 1. Architectural changes and training tweaks
- 2. Muon optimizer
- 3. Dataloading tweaks
- 4. Logit Softcapping

2.1 Architectural changes and training tweaks

There are some basic architectural changes and modernizations that can be made to the model that will speed up training. These changes are general improvements to the transformer decoder architecture that have been generally adopted since the original GPT-2 paper. The changes are:

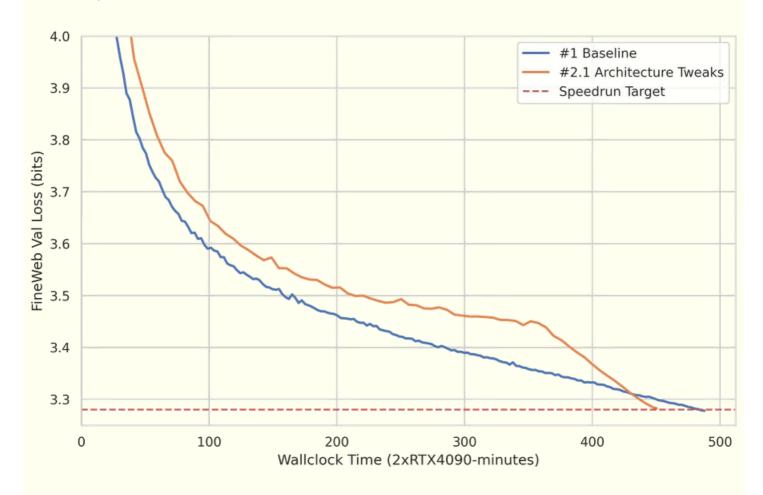
- 1. <u>RoPE (Rotary Positional Embeddings)</u>. There are <u>many good</u> explanations of RoPE out there so I won't go into detail here.
- 2. <u>ReLU^2 Activation</u>¹. Many activations that are better than GeLU have been proposed since GPT-2. ReLU^2 is a simple one that has been shown to be effective in decreasing training time required to reach a certain validation loss.
- 3. No gradient clipping. Gradient clipping can help stabilize training but it also slows down training. Since we are speed-running, we will remove gradient clipping. This also eliminates a hyperparameter that needs to be tuned.
- 4. <u>Trapezoidal learning rate schedule</u>. While cosine learning rate schedules are the de-facto standard, they can be difficult to work with since changing the number of training steps changes the entire schedule. Trapezoidal learning rate schedules are often easier to reason about / tune around, and they have been show to match the performance of cosine schedules.

In addition, learning rate and batch size have been tuned.

Once again, many of these changes are <u>downstreamed</u> from the <u>modded-nanogpt</u> repository / 8xH100 speedrun. Its not efficient to reinvent the wheel, and I want to get training time down as fast as possible in the beginning.

After implementing these changes (commit <u>b7bb93f</u>), the new run time is **7.51 hours**. This run was more data-efficient than the baseline, requiring only 5.07B tokens. However, the tokens/second increased, likely due to the larger batch size (more gradient accumulation

steps which tends to translate to lower throughput) and the architectural changes, such as the inclusion of RoPE. Once I have a shorter run time, I will be able to tune more effectively and see if I can remove gradient accumulation.



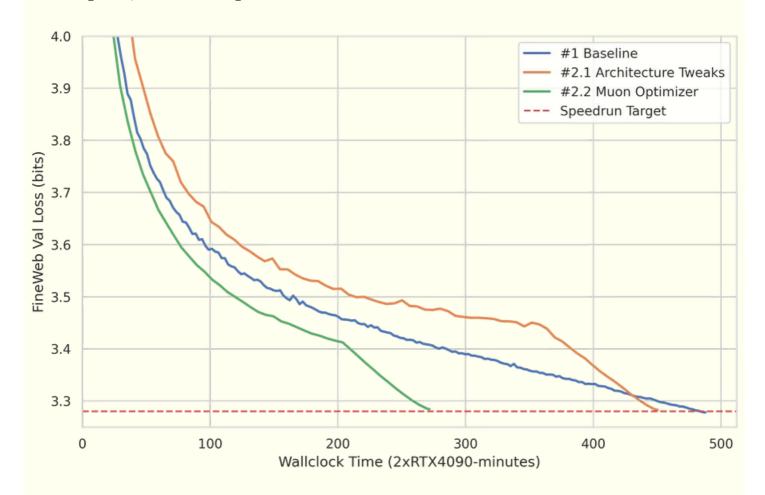
2.2 Muon Optimizer

The <u>Muon Optimizer</u> is a new optimizer developed with and for the NanoGPT speedrun by Jordan et al. It is a variant of SGD with Momentum that applies a postprocessing step to the gradient updates to approximately orthogonalize each update matrix. Muon has <u>some</u> <u>connections</u> to approximate second-order optimizers ² like <u>Shampoo</u>.

I highly recommend reading the original <u>Muon blog post</u> for more details, as well as checking out the optimizer comparison for GPT-2 speedrunning that Keller Jordan put to gether <u>here</u>. For those interested in a more step-by-step walkthrough of Muon, check out <u>this excellent post</u> by Jeremy Bernstein.

Muon is designed to work on *Linear* layers, so it is not quite a drop-in replacement for AdamW (e.g. it isn't meant to optimize Embedding layers). However it can be used to optimize all of the hidden layers of our GPT-2 model. The output 1m_head layer and the token embeddings will still be optimized with AdamW.

Just like on the 8xH100 leaderboard, we observe a massive speedup when switching to Muon. The new run time is 4.53 hours, requiring only 3.04B tokens. The tokens/second is also very similar to the previous run, which is a good sign that we are not losing throughput by switching optimizers.



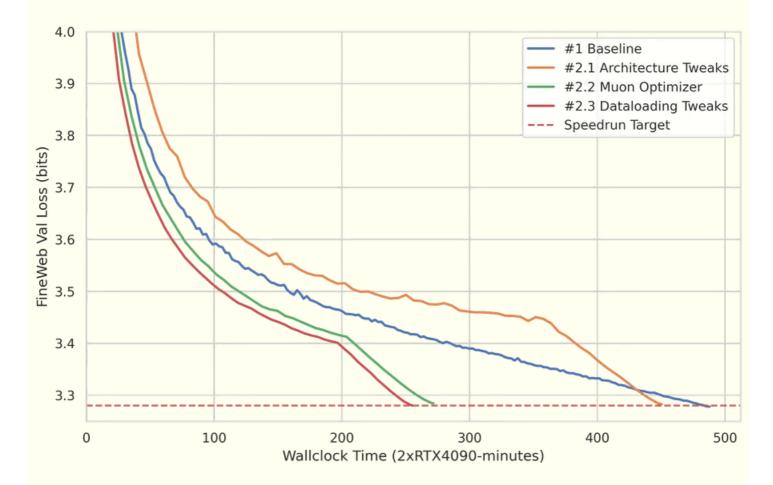
2.3 Dataloading Tweaks

As we have improved our data efficiency via architecture tweaks and an optimizer change, our training throughput has dropped from 221k tokens/second to 187k tokens/second. That is a ~15% drop in throughput. Recovering most of that throughput could provide a significant improvement to our run time. An obvious place to start is with our dataloading and gradient accumulation logic.

Up until now, we have loaded a full-batch of data on each device and then split that full batch into smaller chunks (micro-batches) for each gradient accumulation step (recall that we are doing 8 accumulation steps per gradient update). We can instead make a minor tweak to our logic to load only the next micro-batch at each step of the dataloader, and then step the dataloader for each gradient accumulation step.

We also increase our torch version from 2.5 to 2.6 (which was recently released), and, in accordance with the <u>new official rules</u> designated on 2025/02/01, we have removed the use of torch._inductor.config.coordinate_descent_tuning.

These tweak brings our throughput back up to 216k tokens/second. In order to make runs more consistently hit the 3.28 validation loss target³, we have also slightly increased the total number of training steps, so now 3.31B tokens are consumed. The new run time is **4.26 hours**, and the changes can be found at <u>d59944d</u>.



At this point, we code that can train GPT-2 almost twice as fast as the baseline.

2.4 Logit Soft-capping

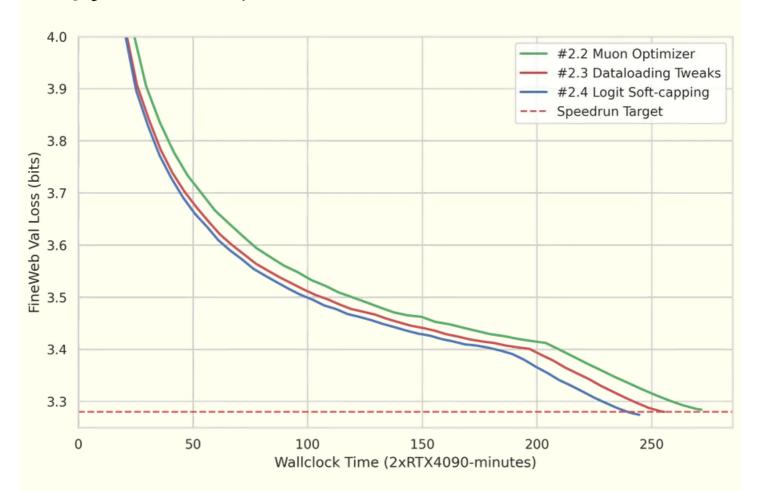
Logit soft-capping is a technique popularized by <u>Gemma 2</u> and initially used to improve the NanoGPT speedrun by <u>@Grad62304977</u>.

Soft-capping is essentially a smooth and differentiable version of clipping ⊕:

$$\operatorname{softcap}(\operatorname{x},\operatorname{cap}) = \operatorname{cap} \cdot \operatorname{tanh}\left(\frac{\operatorname{x}}{\operatorname{cap}}\right)$$

Logit soft-capping prevents logits from growing excessively large by scaling them to a fixed range, which seems to help improve training dynamics. One could argue that this is imposing an inductive bias - and since we're in a relatively small model/low data regime that this is helpful.

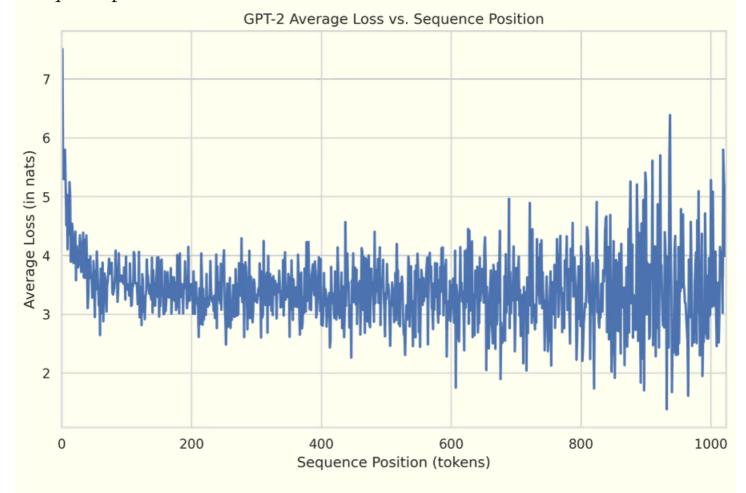
After implementing logit soft-capping with a cap of 30 (and doing some learning-rate tuning), the new run time is 4.01 hours, requiring 3.15B tokens (commit 12eab44). Throughput remained steady at ~218k tokens/second.



3 Longer Training and Evaluation Sequence Length

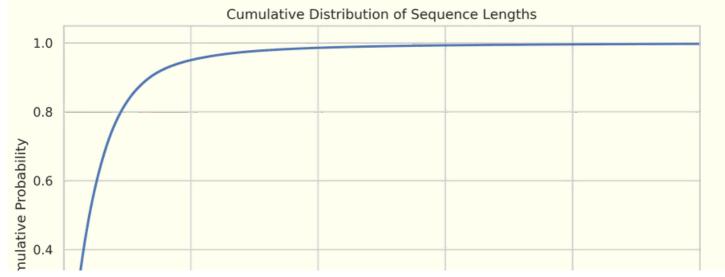
So far, we've been training and evaluating on sequences of 1024 tokens. We also haven't been particularly clever about how those sequences are processed. At each step, we simply load the next 1024 tokens into an element of the batch without regard for where the document starts or stops. That means much of the time we are starting in the middle of a document and cutting that document off before it reaches its end. We are also attending to tokens *across documents* since we're just using a simple causal mask.

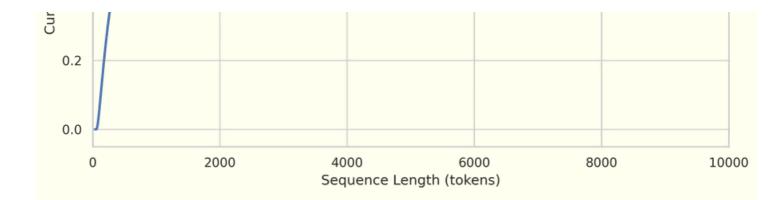
Cutting off documents in the middle is an especially large issue. See this plot of average loss vs sequence position:



Notice how the first twenty-five or so positions have a much higher average loss than the later positions. This is because at the beginning of the sequence the LLM has much less information with which to make informed predictions about the next token in the sequence. We want to avoid needlessly restarting documents/sequences in order to avoid this loss penalty!

A natural question to ask at this point is: how long are sequences in our dataset, on average?





The data reveals that approximately 20% of documents exceed our current 1024 token sequence length. By increasing the sequence length to >=8192 tokens, we can accommodate virtually all documents in our dataset without truncation.

To address the issues identified above, we'll implement two key improvements. First, we'll extend our sequence length to minimize document splitting across sequence boundaries. Taking this approach to its logical conclusion, we'll eliminate the traditional batch dimension entirely and instead maximize sequence length (effectively using a "batch size" of 1 that contains multiple concatenated documents). Second, we'll implement sophisticated attention masking that prevents cross-document attention while simultaneously leveraging the computational efficiency of sparse attention patterns.

Fortunately, <u>FlexAttention</u> provides an elegant solution that maintains the performance benefits of <u>FlashAttention</u> while enabling these improvements. One of FlexAttention's primary strengths is its ability to efficiently handle sparse, custom attention masks, making it ideal for our use case.

To implement FlexAttention, we need to define an appropriate attention mask that handles our specific requirements:

```
def make_attn_mask(idx, eot_token, window_size=1024):
    # Create a causal mask (only attend to past tokens)
    def causal_mask(b, h, q_idx, kv_idx):
        return q_idx >= kv_idx

# Track document boundaries using end-of-text tokens
    documents = (idx == eot_token).cumsum(dim=1)

# Only allow attention within the same document
    def document_mask(b, h, q_idx, kv_idx):
        return documents[b, q_idx] == documents[b, kv_idx]

# Limit attention to an N-token window for efficiency
    def sliding_window_mask(b, h, q_idx, kv_idx):
        return q_idx - kv_idx <= window_size</pre>
```

Let's break down each mask:

- 1. Causal Mask: Standard in autoregressive language modeling. Ensures that tokens can only attend to previous tokens in the sequence, preventing information leakage from future tokens.
- 2. Document Mask: This restricts attention to tokens within the same document. By tracking document boundaries using end-of-text tokens, we prevent tokens from attending across different documents, which helps the model maintain coherent context within a single document.
- 3. Sliding Window Mask: This limits attention to a fixed window of tokens before the current position. This approach balances efficiency with context retention with a clear tradeoff: smaller windows are more efficient but may miss long-range dependencies, while larger windows capture more context at the expense of resources.

In order to build intuition about the individual component masks, we visualize them below:

When combined with the and_masks function, these three masks⁴ work together to create an efficient attention pattern that respects document boundaries, maintains causality, and limits computational overhead for long sequences.

After incorporating FlexAttention with these masks, and increasing our sequence length to 32768 tokens, we observe a massive speedup⁵. The new run time is **2.55 hours**, requiring only 1.88B tokens (a huge data-efficiency improvement). Our throughput dropped slightly to ~205k tokens/second. See commit d982ed5 for the full details.

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