CS190C Lec6

Training and Autoregressive Decoding

Overview

- Model Training: Macro System Architecture
- Training modules
  - Data Sampling Module
  - AdamW Optimizer
  - Training Auxiliary Modules
  - Checkpoints
- Training and Decoding
  - Start of Training: Input Parameter Decoding
  - Complete script
  - Autoregressive Decoding

PART1 - Model Training: Macro System Architecture

Overview of Model Training Process

- Determine the total number of iterative optimization steps. For each step:
  - Sample input text and standard output.
  - Forward.
  - Calculate loss function value.
  - Calculate gradients and clipping.
  - Schedule the learning rate.
  - Update parameters.
- It is recommended to output real-time status during the process, such as the current loss function value.

Modules to Implement

- Input & Standard output data $\Rightarrow$ `Data Sampling` Module
- Loss calculation $\Rightarrow$ `Cross-Entropy Loss` Module
- Calculate and process Gradients $\Rightarrow$ `Gradient Clipping` Module
- Parameter Optimization $\Rightarrow$ `Learning Rate Scheduler` Module, `AdamW` Optimizer Module
- `Checkpoint` Module, `Pre-decoding` Module

PART2: Data Sampling Module

Memory Issues with Very Large Corpus

- The corpus will initially encoded to a long string of numbers using BPE.
- To sample a part of it as a "sentence", a naive method is to read the whole long string into memory, and sample hundreds of numbers.
- If the corpus is extremely large, reading all of them will bring disaster to memory.

Memory Issues with Very Large Corpus

- Is there a solution that reads only specific parts needed from disk to memory?
- `numpy.memmap`: A list that interacts directly with disk space for reading and writing, and highly similar to a regular list in other aspects.
- We can implement `class Memmap_Manager`, containing `def save_as_memmap` and `def load_by_range`, for disk-oriented read/write operations.
- The core idea is: Store the whole long string in the disk, only read a small part of it to memory according to our needs.

Ideas

We can realize this idea roughly……

- Suppose the whole long string contains `500,000,000` numbers.
- We can cut it into small chunks, each chunk's length= `500,000`.
- Store these `10000` chunks into the disk.
- Suppose we need to read `2000th-8000th` number, what chunks should we load to memory? `490,000th-1,050,000th`?
- These two phases correspond to `def save_as_memmap` and `def load_by_range`.

1. `def save_as_memmap`

- Use `BPE_Tokenizer`'s `encode_iterable`, reading only a single number each time.
- Continuously read numbers into a buffer list.
- When the buffer reaches a certain size (e.g., 500,000 numbers), write it as a whole block to disk.
- Clear the buffer and continue reading and writing blocks until the corpus is finished.
- Count the total number of integer codes in the corpus during the process.

That is: **saving as encoding**.

1. `def save_as_memmap`

```
Code here
```

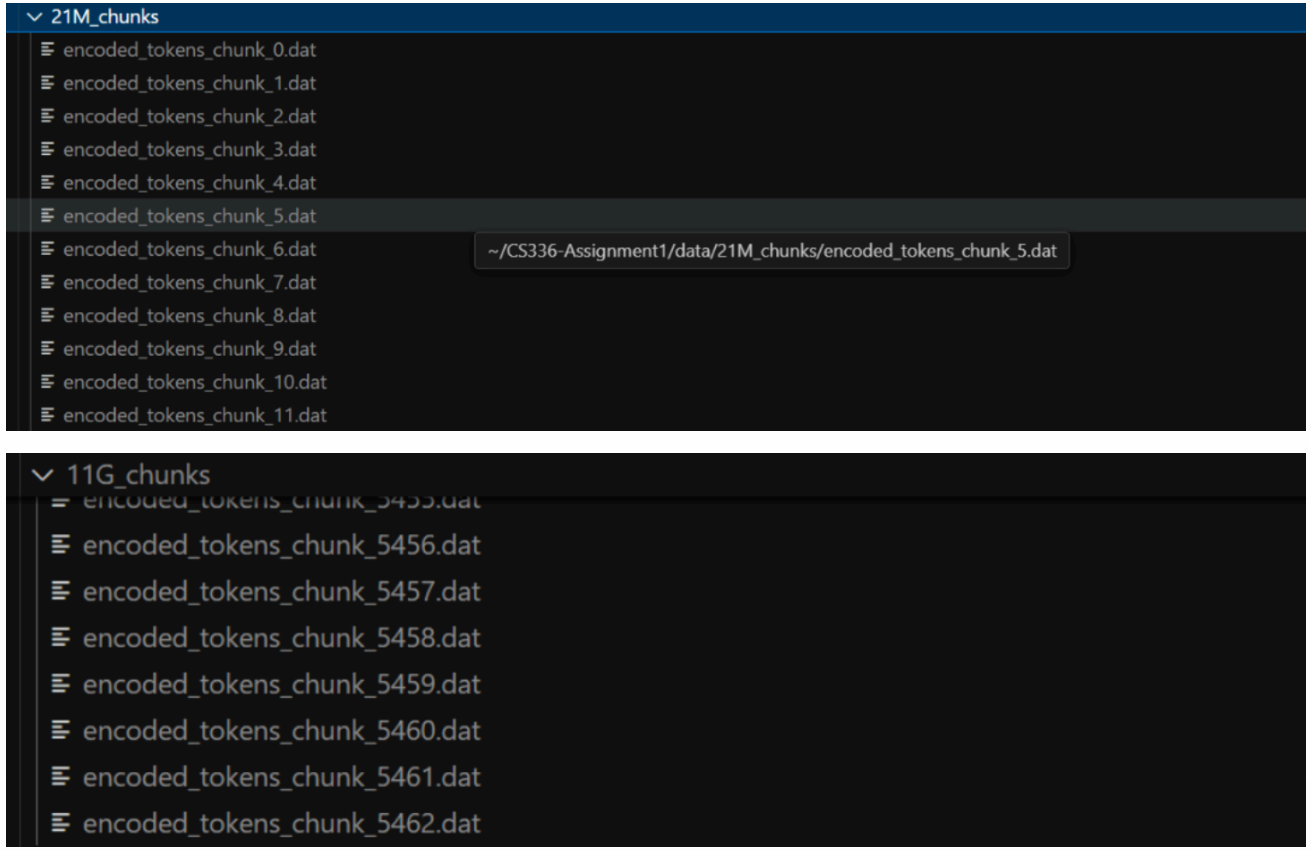
1. `def save_as_memmap`

```
Code here
```

For np.memmap:

- Define array type and shape in advance.
- Operate like a regular array, still stored in memory.
- Explicit flush operation writes to disk, clearing previous memory automatically.

1. `def save_as_memmap`



Attempting to load such a large complete list into memory is catastrophic!

2. `def load_by_range`

Requirement: Need to read all elements in the range `[start_idx, end_idx)` of the encoded list.

- Calculate: Which chunks covered?
- Load only corresponding intervals into memory.
- Operations are consistent with regular lists, except declaring the np.memmap type.

2. `def load_by_range`

`memmap_arr[idx_in_start:idx_in_end]`: Loads this interval from disk into memory.

```
Code here
```

3. Formal Sampling Operation: `class Batch_By_Memmap`

Problem1: How to sample input data?

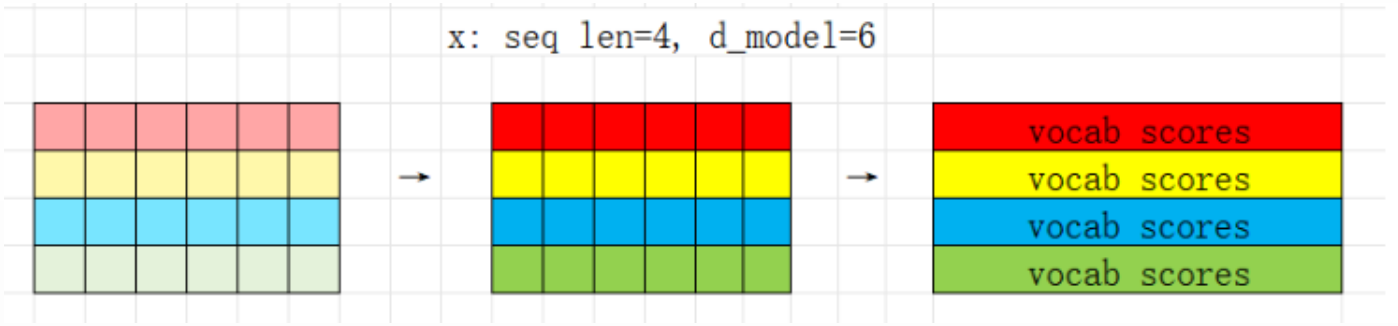
Assuming a corpus length of 10 and a single sampling sequence length of 4:

- Dense sampling: 1234; 2345; 3456; ...
  - Too much data. If the total number of training steps is small (i.e., 3 samples), the second half of the corpus might not be covered.
- Sequential sampling: 1234; 5678; ...
  - Some data is missing. E.g., cannot get 3456 as training input.
- Compromise: Random sampling
  - Randomly pick a start point (ensuring it's <= 7), then continuously sample 4 characters from the start point.
  - Usually sample `batch_size` sequences simultaneously, so randomly pick `batch_size` start points to sample.

3. Formal Sampling Operation: `class Batch_By_Memmap`

Problem2: What is the standard output?

- Review what LLM done: each vector is finally linearly projected to `vocab_size` scale weights to predict the next position's word (as the diagram below).
- So the standard output of position-i should be the $i+1$-th word in the corpus !
- That is: if the input sequence is 1234, the standard output should be 2345.



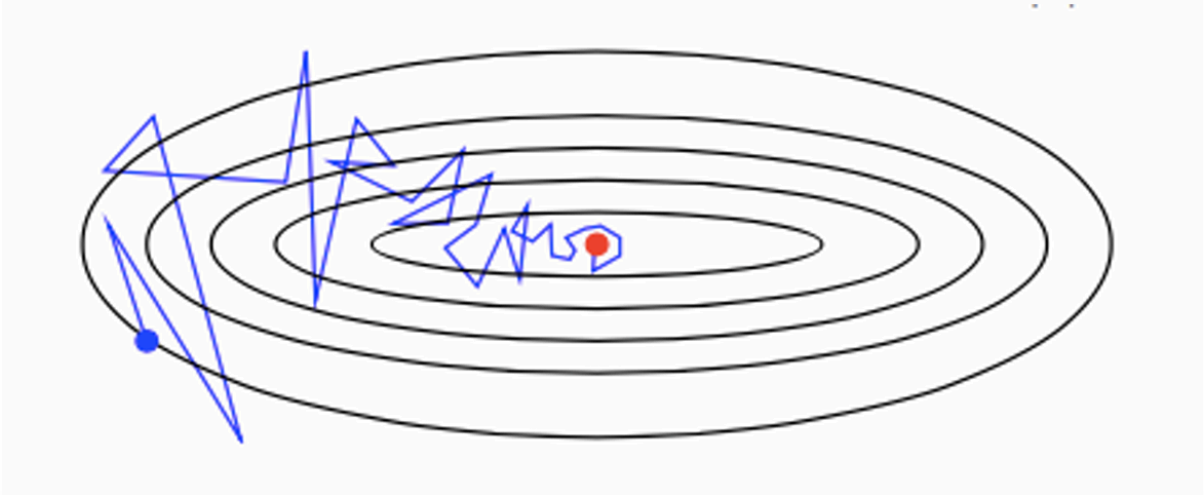
3. Formal Sampling Operation: `class Batch_By_Memmap`

```
Code here
```

PART3: AdamW Optimizer

1. From `SGD` to `AdamW`

SGD: $\theta = \theta - \alpha \nabla L(\theta)$



1. From `SGD` to `AdamW`

What's the defect of SGD?

- Zigzagging because the gradient may not stable.
- Share the same learning rate for updating of all parameters
  - Some parameter, such as word embedding of `the`, may update repeatedly and significantly, leading to possible oscillation.
  - Some parameter, such as word embedding of `pneumoconiosis`, may rarely update, leading to little optimization.

1. From `SGD` to `AdamW`

What we do in `AdamW`?

- Use first moment `m` (we called "momentum") to record historical gradients information.
- Use second moment `v` to record historical gradients fluctuation information.
- Use first and second moment to adjust parameter update process.

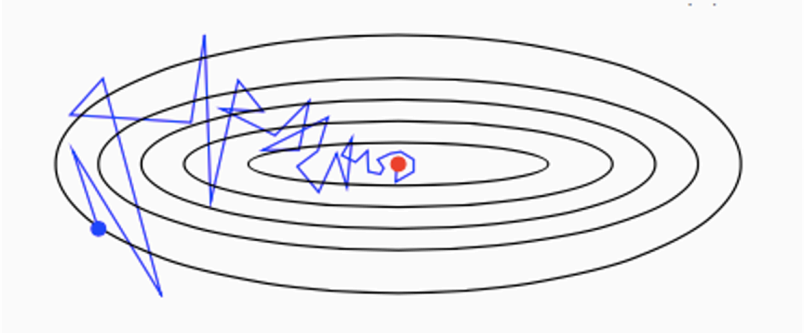
1. From `SGD` to `AdamW`

Let gradient of step t be g_t:

$$m_t = \beta_1(m_{t-1} - g_t) + g_t$$

It means: $m_t = (1-\beta_1)\sum_{\tau=1}^{t}\beta_1^{t-\tau}g_\tau$, it is the exponential weighting of historical gradients.

We use m_t as the "gradient" of step t instead of g_t, reducing zigzagging of gradient. m_t now can hardly fluctuation so serious.



1. From `SGD` to `AdamW`

$$v_t = \beta_2(v_{t-1} - g_t^2) + g_t^2$$

It means: $v_t = (1-\beta_2)\sum_{\tau=1}^{t}\beta_2^{t-\tau}g_\tau^2$, it is the exponential weighting of historical square of gradients.

And also, it can understand as: Variance given `mean=0` and `exponential weights`. But what if `mean≠0`?

Suppose a sequence of gradients with the same mean value: `2,2,2,2` `3,1,3,1`, we find `3,1,3,1` still leads to larger value of v. So it can reflect the fluctuation of historical gradients.

1. From `SGD` to `AdamW`

So how do we use these two information to adjust $\theta_t = \theta_{t-1} - \alpha g_t$?

- For first moment, change g_t to m_t
- How to make use of secomd moment?
  - If v is large: It means the gradient of parameter fluctuate seriously, such as word embedding of high frequency words. The update step should be smaller.
  - If v is small: It means the parameter rarely update, for example, the history gradient are `0,0,0,1,0,0`. The update step should be larger.

$$\theta_t = \theta_{t-1} - \alpha \frac{m_t}{\sqrt{v_t} + \epsilon}$$

1. From `SGD` to `AdamW`

$$\theta_t = \theta_{t-1} - \alpha \frac{m_t}{\sqrt{v_t} + \epsilon}$$

Is there any hidden problem of it?

When t is small……

- $m_1 = (1-\beta_1)g_1, m_2 = \beta_1(1-\beta_1)g_1 + (1-\beta_1)g_2$……
- Suppose all $|g_i|$ share the similar scale, so $|m_1| = (1-\beta_1)|g|, |m_2| = (1-\beta_1^2)|g|$ $, \ldots, |m_t| = (1-\beta_1^t)|g| \Rightarrow$ The scale is distortion comparing with $|g|$
- The same with $|v_t|$: $|v_t| = \sqrt{1-\beta_2^t}|g|$

1. From `SGD` to `AdamW`

Fix the scale distortion: $m_t \Rightarrow \frac{m_t}{1-\beta_1^t}$, $v_t \Rightarrow \frac{v_t}{\sqrt{1-\beta_2^t}}$

That is:

$$\theta_t = \theta_{t-1} - \alpha \frac{\sqrt{1-\beta_2^t}}{1-\beta_1^t} \frac{m_t}{\sqrt{v_t}+\epsilon}$$

We can implement it in engineering:

$$\alpha_t = \alpha \frac{\sqrt{1-\beta_2^t}}{1-\beta_1^t}$$

$$\theta_t = \theta_{t-1} - \alpha_t \frac{m_t}{\sqrt{v_t}+\epsilon}$$

1. From `SGD` to `AdamW`

So far: It is the adjustment of SGD without regularization. What if we consider $L2$-regularization?

$L_{total}(\theta) = L_{task}(\theta) + \frac{\lambda}{2}\|\theta\|_2^2$
$\Rightarrow \theta = \theta - \alpha\nabla L_{task}(\theta) - \alpha\lambda\theta$

The same to `AdamW`:

$$\theta_t = \theta_{t-1} - \alpha\frac{\sqrt{1-\beta_2^t}}{1-\beta_1^t}\frac{m_t}{\sqrt{v_t}+\epsilon} - \alpha\lambda\theta$$

$$= \theta_t = \theta_{t-1} - \alpha_t\frac{m_t}{\sqrt{v_t}+\epsilon} - \alpha\lambda\theta$$

1. From `SGD` to `AdamW`

What the order of magnitude of scale of updating?

SGD:$|\alpha g_t| = \Theta(\alpha|g|)$

AdamW: $|\alpha_t \frac{m_t}{\sqrt{v_t}+\epsilon}| = \Theta(\alpha_t \frac{\sum_{k=1}^{t} \beta_1^k |g|_{t-k}}{\sqrt{\sum_{k=1}^{t} \beta_2^k |g|_{t-k}^2}})$

When the training process enter the middle and late stages,

$\alpha_t \to \alpha$
$|g|$ can be considered as a sequence with `mean` $= 0$ and `variation` $= \sigma^2$
$\Rightarrow |g| \to \Theta(\sigma)$

1. From `SGD` to `AdamW`

AdamW: $|\alpha_t \frac{m_t}{\sqrt{v_t}+\epsilon}| = \Theta(\alpha_t \frac{\sum_{k=1}^{t} \beta_1^k |g|_{t-k}}{\sqrt{\sum_{k=1}^{t} \beta_2^k |g|_{t-k}^2}})$

$|m_t| = \sum_{k=1}^{t} \beta_1^k |g|_{t-k} = \Theta(\sigma)$

$|\sqrt{v_t}| = \sqrt{\sum_{k=1}^{t} \beta_2^k |g|_{t-k}^2} = \Theta(\sqrt{\sigma^2}) = \Theta(\sigma)$

$|\alpha_t \frac{m_t}{\sqrt{v_t}+\epsilon}| = \Theta(\alpha)$, which is only related to base learning rate.

2. `torch.optim.Optimizer` Base Class

Any implementation of a custom optimizer inherits from the `torch.optim.Optimizer` base class.
The Optimizer base class provides two-level management for all model parameters:

- `self.param_groups`: Group parameters. Each group can set different features like learning rate. (First-level management: parameter grouping, sharing states within the group)
- Each `param_groups` contains at least the default key-value pair: "params", corresponding to a list of parameters. Other key-value pairs like "lr" can also be customized.
- For each parameter in the parameter list, various states can also be set, e.g., iteration step `"step"`. (Second-level management: finer-grained states for individual parameters)

2. `torch.optim.Optimizer` Base Class

```
Code here
```

3. `class AdamW_Optimizer`

Algorithm 1 AdamW Optimizer

init(θ) (Initialize learnable parameters)
$m \leftarrow 0$ (Initial value of the first moment vector; same shape as θ)
$v \leftarrow 0$ (Initial value of the second moment vector; same shape as θ)
for $t = 1, \ldots, T$ **do**
  Sample batch of data B_t
  $g \leftarrow \nabla_\theta \ell(\theta; B_t)$ (Compute the gradient of the loss at the current time step)
  $m \leftarrow \beta_1 m + (1 - \beta_1) g$ (Update the first moment estimate)
  $v \leftarrow \beta_2 v + (1 - \beta_2) g^2$ (Update the second moment estimate)
  $\alpha_t \leftarrow \alpha \frac{\sqrt{1-(\beta_2)^t}}{1-(\beta_1)^t}$ (Compute adjusted α for iteration t)
  $\theta \leftarrow \theta - \alpha_t \frac{m}{\sqrt{v+\epsilon}}$ (Update the parameters)
  $\theta \leftarrow \theta - \alpha \lambda \theta$ (Apply weight decay)
end for

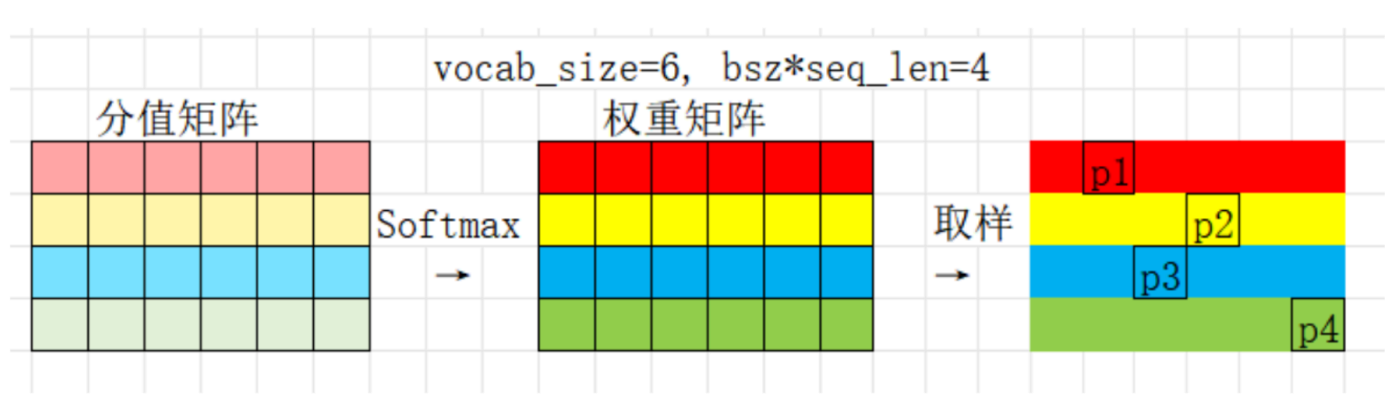
```
Code here
```

PART4: Training Auxiliary Modules

1. `class Cross_Entropy_Calculator`

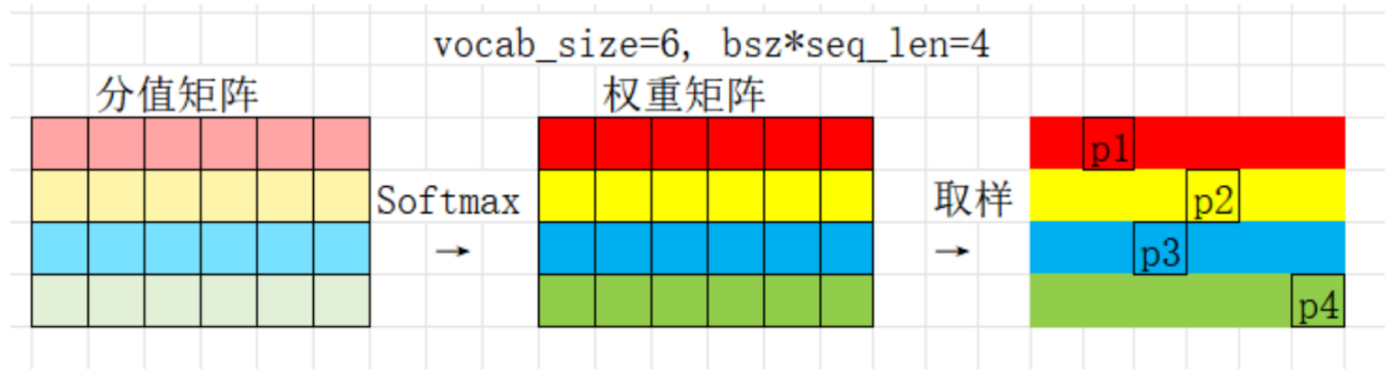
Review: What LLM done, and what's the meaning of it.

- Transformer output shape is `[bsz, seq_len, vocab_size]`. I.e., there are `(bsz*seq_len)` output predictions, each prediction is "the possibility score s_i for each word at the next position" $(1 \leq i \leq \text{bsz*seq\_len})$, same below.
- Perform Softmax to get probability weight distribution p_i of next word.



1. `class Cross_Entropy_Calculator`

- Take $-\log$ of the probability weight to get $l_i = -\log(p_i)$, the mean of all l_i is the **cross-entropy loss**.
- That is: $\text{Loss} = \text{Mean}\{-l_i\} = \text{Mean}\{-\log p_i\} = \text{Mean}\{-\log\text{Softmax}(s_i)\}$



1. `class Cross_Entropy_Calculator`

Numerical hazard: What if p_i is too small and approximated as 0?

- Suppose token1's 6-word weight scores are `100, -1, 1, 5, 2, 1000`, and the standard output is word2, with score `-1` and weight `0`.
- Calculation result: $\log 0 = \mathrm{NAN}$

Formula derivation:

$$\begin{aligned}
&\mathrm{logSoftmax}(s_i) = \log \frac{\exp(s_i)}{\sum \exp s_k} \\
&= \log \frac{\exp(s_i - s_{max})}{\sum \exp(s_k - s_{max})} \\
&= (s_i - s_{max}) - \log(\sum \exp(s_k - s_{max}))
\end{aligned}$$

1. `class Cross_Entropy_Calculator`

```
Code here
```

1. `class Cross_Entropy_Calculator`

```
Code here
```

2. `class Gradient_Clipper`

Suppose the gradient of parameter is too large?

- SGD: $|\alpha g_t| = \Theta(\alpha|g|)$. It may lead to failure of optimization.
- AdamW: $|\alpha_t \frac{m_t}{\sqrt{v_t}+\epsilon}| = \Theta(\alpha)$. It seems no problem?

Review: v_t of AdamW: $v_t = (1-\beta_2)\sum_{\tau=1}^{t}\beta_2^{t-\tau}g_\tau^2$

- v_t becomes extremly large during next hundreds of steps
- $\alpha_t \frac{m_t}{\sqrt{v_t}+\epsilon} \to 0$, which means the optimization stops abnormally.

So when the gradient is too large, we need to scale it to a small enough level.

2. `class Gradient_Clipper`

Suppose all parameters of a model come from 5 Linear_Transform (5D to 5D), then the model has 5 parameter tensors (all 5*5 shape).
After one round of backpropagation, suppose gradient of all 5 parameters are equal to:

```
tensor([[1., 1., 1., 1., 1.],
        [1., 1., 1., 1., 1.],
        [1., 1., 1., 1., 1.],
        [1., 1., 1., 1., 1.],
        [1., 1., 1., 1., 1.]])
```

Let the parameter tensors be p_1, p_2, p_3, p_4, p_5. $\|\nabla p_i\|_2 = \sqrt{25 * 1} = 5$

So the grad norm is $\sqrt{\sum_{i=1}^{5} \|\nabla p_i\|_2^2} = \sqrt{125} = 11.18$

2. `class Gradient_Clipper`

- Assuming the acceptable upper limit is `max_norm` $=0.01$:
- All parameters must be multiplied by the scaling factor $\frac{\text{max\_norm}}{g+\epsilon}$ to ensure the total L2 norm size of gradients is reasonable.
- After clipping, the gradients of each parameter are $11.18/0.01$ times smaller:

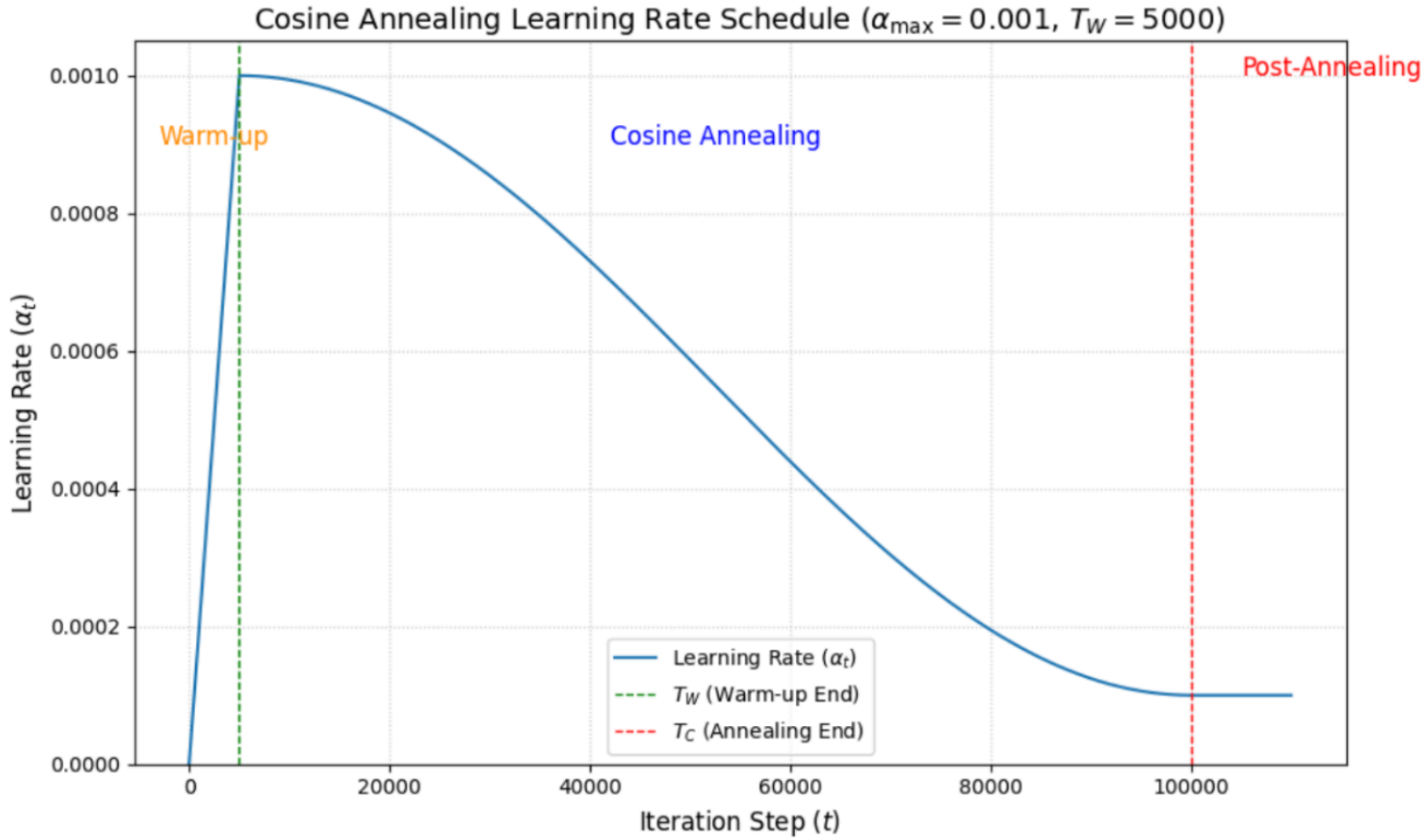
```
tensor([[0.0009, 0.0009, 0.0009, 0.0009, 0.0009],
        [0.0009, 0.0009, 0.0009, 0.0009, 0.0009],
        [0.0009, 0.0009, 0.0009, 0.0009, 0.0009],
        [0.0009, 0.0009, 0.0009, 0.0009, 0.0009],
        [0.0009, 0.0009, 0.0009, 0.0009, 0.0009]])
```

2. `class Gradient_Clipper`

```
Code here
```

3. `class Learning_Rate_Scheduler`

Throughout the model training process, the learning rate is not constant:



3. `class Learning_Rate_Scheduler`

```
Code here
```

PART5: Checkpoints

A hidden problem

Usually we need hours even days to train a large model……

Can we ensure there's nothing wrong during the whole process?

- For example, the computing power resource you applied for has been preempted, and everything crashed immediately.

To reduce the problem, we try to save to model every once in a while.

This is `checkpoint`.

What to save and load?

- Model parameters
- Optimizer parameters (such as m_t amd v_t)
- Step now

We need to get them first, and try to save them as files in the disk.

1. How to Get All Parameters?

- PyTorch has a built-in `state_dict()` function that stores all parameters of the PyTorch module in dictionary form.

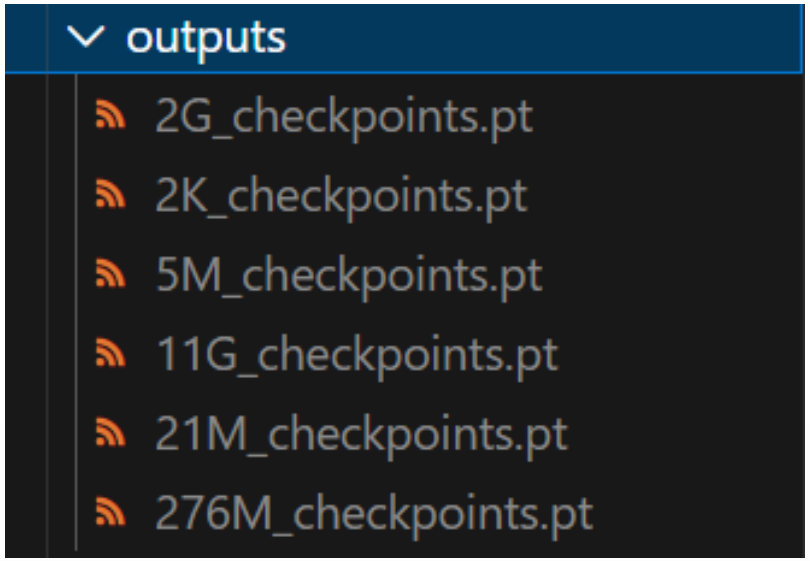
```
lm = Transformer_LM(
    d_model=512,
    num_heads=8,
    d_ff=2048,
    vocab_size=10000,
    num_layers=2,
    max_seq_length=128,
    theta=10000,
    dtype=torch.float32,
    device="cpu"
)

states = lm.state_dict()
for state_key in states:
    print(state_key, states[state_key].shape)
```

```
transformer_blocks.0.RMSNorm_Attn.g torch.Size([512])
transformer_blocks.0.RMSNorm_FF.g torch.Size([512])
transformer_blocks.0.Multihead_Attn.q_proj.linear_matrix torch.Size([512, 512])
transformer_blocks.0.Multihead_Attn.k_proj.linear_matrix torch.Size([512, 512])
transformer_blocks.0.Multihead_Attn.v_proj.linear_matrix torch.Size([512, 512])
transformer_blocks.0.Multihead_Attn.o_proj.linear_matrix torch.Size([512, 512])
transformer_blocks.0.Multihead_Attn.rope.cos_values torch.Size([1, 128, 32])
transformer_blocks.0.Multihead_Attn.rope.sin_values torch.Size([1, 128, 32])
transformer_blocks.0.Feed_Forward.linear_w1.linear_matrix torch.Size([512, 2048])
transformer_blocks.0.Feed_Forward.linear_w3.linear_matrix torch.Size([512, 2048])
transformer_blocks.0.Feed_Forward.linear_w2.linear_matrix torch.Size([2048, 512])
transformer_blocks.1.RMSNorm_Attn.g torch.Size([512])
transformer_blocks.1.RMSNorm_FF.g torch.Size([512])
transformer_blocks.1.Multihead_Attn.q_proj.linear_matrix torch.Size([512, 512])
transformer_blocks.1.Multihead_Attn.k_proj.linear_matrix torch.Size([512, 512])
transformer_blocks.1.Multihead_Attn.v_proj.linear_matrix torch.Size([512, 512])
transformer_blocks.1.Multihead_Attn.o_proj.linear_matrix torch.Size([512, 512])
transformer_blocks.1.Multihead_Attn.rope.cos_values torch.Size([1, 128, 32])
transformer_blocks.1.Multihead_Attn.rope.sin_values torch.Size([1, 128, 32])
transformer_blocks.1.Feed_Forward.linear_w1.linear_matrix torch.Size([512, 2048])
transformer_blocks.1.Feed_Forward.linear_w3.linear_matrix torch.Size([512, 2048])
transformer_blocks.1.Feed_Forward.linear_w2.linear_matrix torch.Size([2048, 512])
final_norm.g torch.Size([512])
final_layer.linear_matrix torch.Size([512, 10000])
```

2. How to Load All Parameters?

- PyTorch has a built-in `load_state_dict()` function that loads the values of all parameters of the PyTorch module from a state_dict dictionary.
- Prerequisite: Both must match completely!



3. `class Checkpoint_Manager`

- Objects to store: Model parameters, optimizer parameters, current iteration step.

```
Code here
```

3. `class Checkpoint_Manager`

- Objects to load: Model parameters, optimizer parameters, current iteration step.

```
Code here
```

PART6: Start of Training: Input Parameter Decoding

To This Point...

We have sorted out:

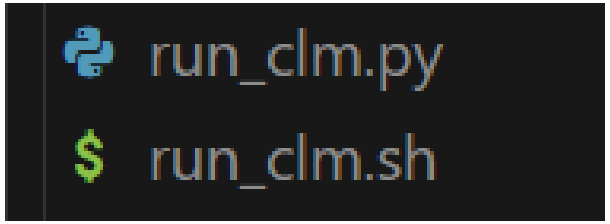
- The architecture and parameters to pass of each model module.
- The implementation and parameters to pass of each training module.

Regarding passing parameters...

How do we specify the specific values of these parameters, **effectively** modify and pass certain values to each module?

1. Starting Training Scripts

- We can directly assign values to various parameters in the training script.
- When the training script is very complex, various functional modules will be ambiguous.
- We usually use a `bash` script to carry the required parameter values and start the execution of the `python` training script.



1. Starting Training Scripts

```
#!/bin/bash
python /home/kuangph/CS336-Assignment1/cs336_basics/run_clm.py \
    --d_model 512 \
    --num_heads 8 \
    --d_ff 1344 \
    --vocab_size 32000 \
    --num_layers 8\
    --max_seq_length 256 \
    --seq_length 256 \
    --batch_size 48 \
    --theta 100000 \
    --device cuda \
    --num_epochs 5.5 \
    --lr 1e-4 \
    --lr_min 1e-5 \
    --warmup_ratio 0.05 \
    --warmfix_ratio 0.9 \
    --chunk_size 500000 \
    --vocab_path /home/kuangph/CS336-Assignment1/data/vocab_32000.txt \
    --merges_path /home/kuangph/CS336-Assignment1/data/merges_32000.txt \
    --special_tokens "<|endoftext|>" \
    --corpus_size "2G" \
    --log_interval 20 \
    --save_interval 500 \
    --weight_decay 0.01 \
    --betas 0.9 0.95 \
    --eps 1e-8 \
    --max_norm 1.0
```

- It is equal to enter a long command line in the terminal, containing information we need
- For `run_clm.py`, we need to properly decode information of command line.

2. Parameter Decoding

```
def parse_bash_args():
    parser=argparse.ArgumentParser()

    parser.add_argument("--d_model",type=int,default=512)
    parser.add_argument("--num_heads",type=int,default=8)
    parser.add_argument("--d_ff",type=int,default=1344)
    parser.add_argument("--vocab_size",type=int,default=32000)
    parser.add_argument("--num_layers",type=int,default=6)
    parser.add_argument("--max_seq_length",type=int,default=512)
    parser.add_argument("--seq_length",type=int,default=256)
    parser.add_argument("--batch_size",type=int,default=32)
    parser.add_argument("--theta",type=int,default=100000)
    parser.add_argument("--device",type=str,default="cuda")

    parser.add_argument("--num_epochs",type=float,default=10)
    parser.add_argument("--lr",type=float,default=1e-4)
    parser.add_argument("--lr_min",type=float,default=1e-5)
    parser.add_argument("--warmup_ratio",type=float,default=0.1)
    parser.add_argument("--warmfix_ratio",type=float,default=0.9)
    parser.add_argument("--corpus_size",type=str)
```

```
parser.add_argument("--chunk_size",type=int,default=500000)
parser.add_argument("--vocab_path",type=str,
                    default="data/vocab_32000.txt")
parser.add_argument("--merges_path",type=str,
                    default="data/merges_32000.txt")
parser.add_argument("--special_tokens",
                    type=str, nargs="*", default=["<|endoftext|>"])

parser.add_argument("--log_interval",type=int)
parser.add_argument("--save_interval",type=int)

parser.add_argument("--weight_decay",type=float,default=0.01)
parser.add_argument("--betas",type=float, nargs="*",
                    default=(0.9,0.95))
parser.add_argument("--eps",type=float,default=1e-8)

parser.add_argument("--max_norm",type=float,default=1.0)

args=parser.parse_args()
return args
```

2. Parameter Decoding

```
Code here
```

```
Code here
```

PART7: Complete Script

Review of Model Training Process

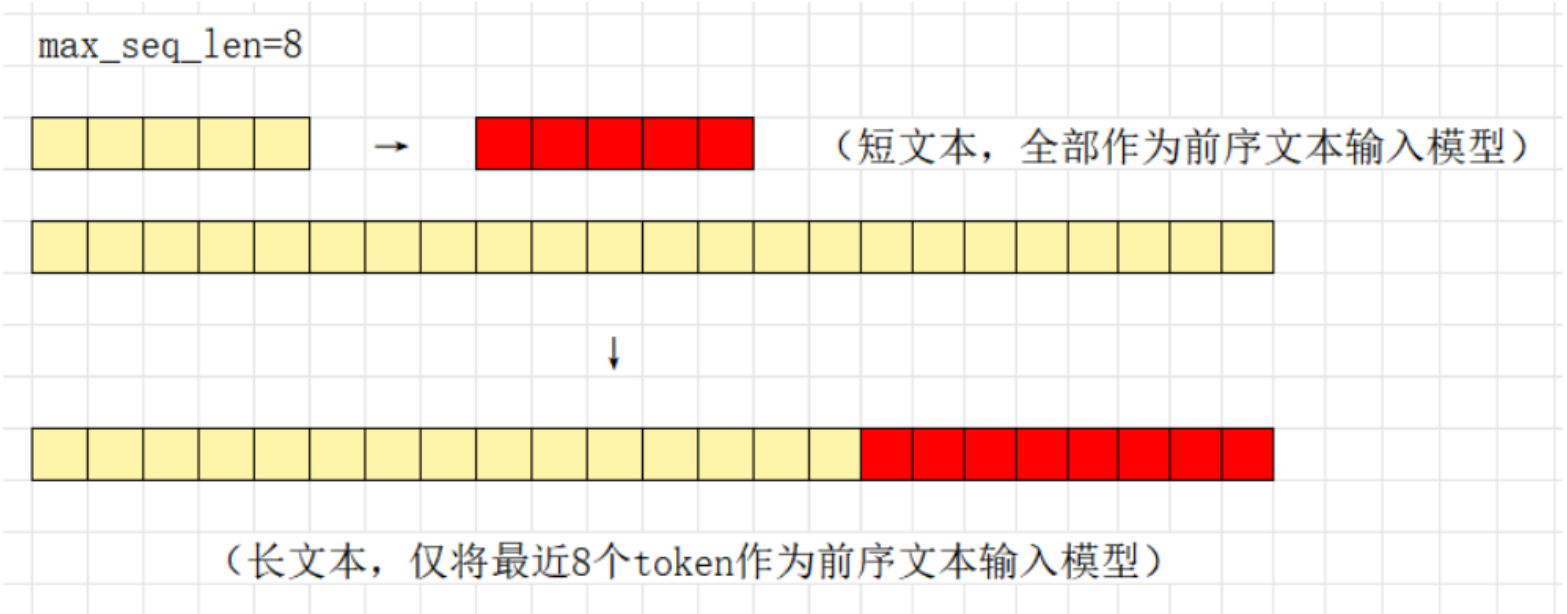
- Receive parameters, establish model and training modules.
- Determine the total number of iterative optimization steps.
- For each step:
  - Sample data.
  - Forward.
  - Calculate loss.
  - Calculate gradients and clip.
  - Schedule the learning rate.
  - Update parameters.
- It is recommended to output real-time status during the process, such as the current loss function value, current text prediction situation, etc.

Code Explanation

PART8: Autoregressive Decoding

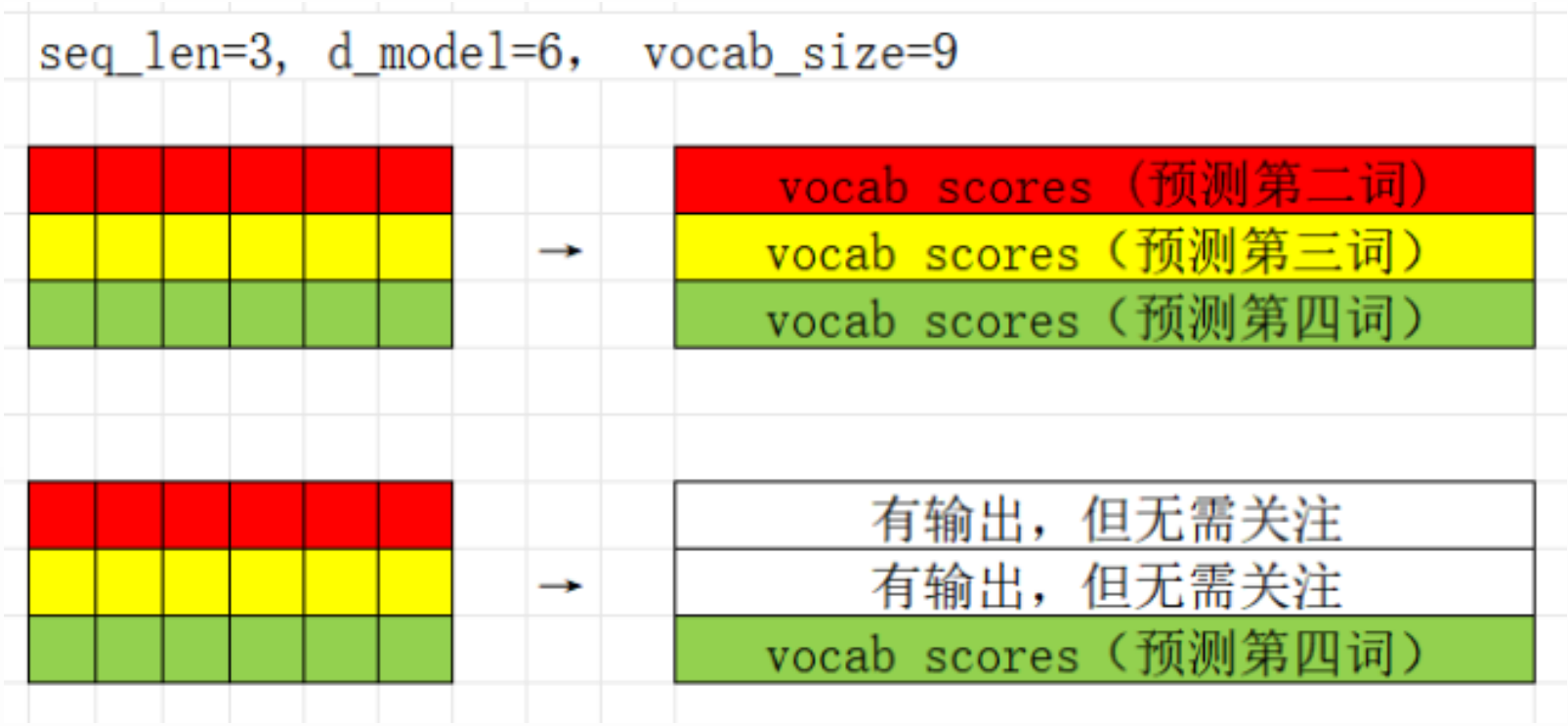
Determining Input Text

During training, fixed-length text is often sampled as input to the model. However, the total length of decoded text may be unpredictable.



Sampling Output Values

- During training, we need to know the output at all positions to calculate the loss value;
- During decoding, we only care about "what is the next word?"!



From Vocab Scores to Token Sampling

After receiving vocab scores:

- Repetition penalty: Reduce scores of recently appeared tokens to avoid repeating the same word.
- Temperature sampling: Scale all scores by the same factor to make softmax differences larger/smaller.
- Softmax: Convert scores into a probability distribution.
- Randomly sample the next token according to the probability distribution.

Autoregressive

The sampled output this step is a part of input next step.

Code Explanation and Practical Demonstration