Grammar Ninja

Abstract

Improving one's English language writing is a significant challenge without access to proficient teachers that can provide valuable feedback. Given the recent rapid acceleration in generative models ability to understand language, we aim to develop a model/fine-tune a model to provide an interface that will generate feedback given text as an input. Our goal is provide quantitative benchmarks for language proficiency in six different areas: cohesion, syntax, vocabulary, phraseology, grammar, and conventions in a provided writing. Additionally, we also generate feedback at the inference layers to provide concrete feedback as to how the input text can be improved. To conclude, our project aims to apply the concepts learned in class to a real-world challenge, by providing a interface to acquire feedback on English writing. By focusing on key areas of language skills and providing model generated actionable feedback, we hope to contribute a somewhat practical tool.

Introduction

This project aims to develop a Natural Language Processing (NLP) based system that can automatically evaluate and provide feedback on student argumentative essays. The system will focus on several key aspects of writing, including the effectiveness of arguments, grammar, use of evidence, syntax, and tone. The feedback can be either quantitative, in the form of scores in various categories, or qualitative, as generated English feedback that offers specific guidance and suggestions for improvement.

To this end, we have decided to split our problem into three separate parts - predicting feedback scores based on the writing, identifying the different parts of argumentative writing (think Parts-Of-Speech tagging but for sentences as opposed to words), and finally constructing a generative model to produce the corrected version of a given input sentence.

Methodology

In this section, we will discuss the various approaches we have taken to fine-tune the three models.

Feedback Scores (Bert-cased)

We used the English feedback dataset here. This was dataset we chose to use to identify scoring the dataset on the following features: cohesion, syntax, vocabulary, phraseology, grammar, conventions.

Data Exploration

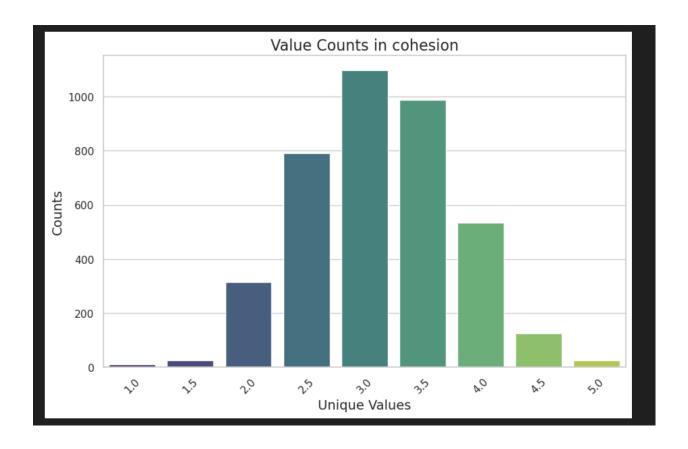
Given some input data X, we want to predict writing feedback scores y. The writing feedback scores range from 1 to 5 in 0.5 increments.



We noticed a few things, most importantly that the target classes are imbalanced:

```
cohesion
3.0
      1096
3.5
       988
2.5
      790
4.0
      534
2.0
      315
4.5
      125
1.5
    27
5.0
       26
1.0
        10
Name: count, dtype: int64
syntax
3.0
     1250
3.5
      867
2.5
      839
2.0
      410
4.0
      388
4.5
      100
1.5 29
5.0
      17
1.0
        11
Name: count, dtype: int64
vocabulary
3.0
     1503
3.5
     1007
. . .
       25
5.0
1.5
       20
1.0
       15
Name: count, dtype: int64
```

For example, cohesion:



Imbalanced Data - Applying Target Transformations

So, we now had to make a decision regarding how to re-balance the dataset, so the model can learn equally well on less represented data as well. Here, we considered some strategies:

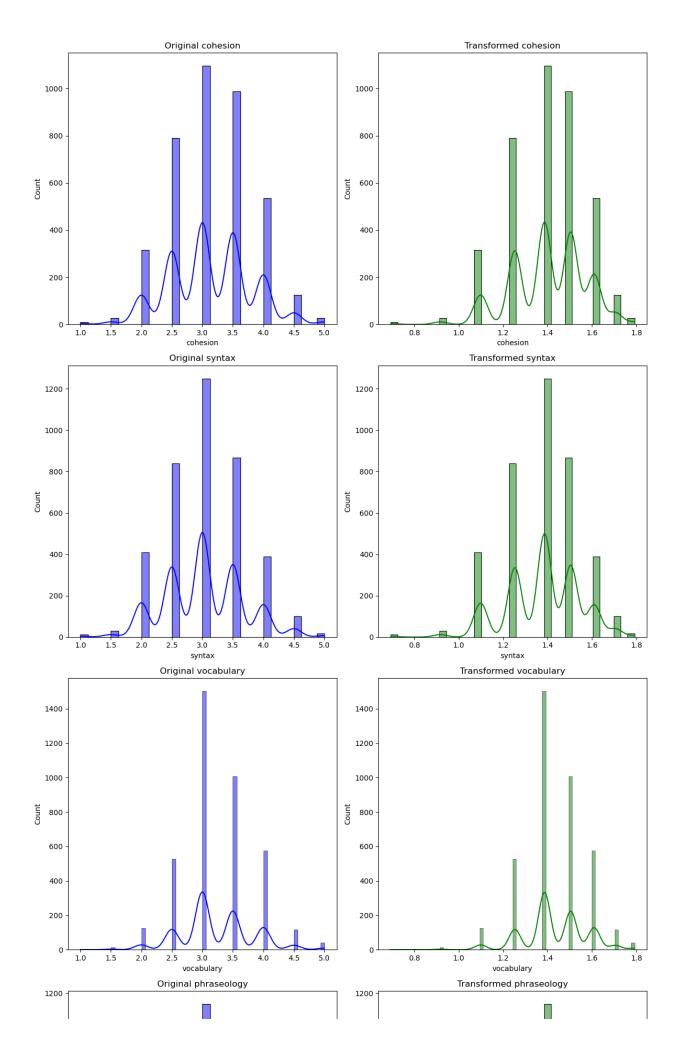
- Log Transformation: Useful for right-skewed distributions. However, it can't be applied to zero or negative values directly.
- Square Root Transformation: Less aggressive than log transformation and can be used for right-skewed distributions.
- Box-Cox Transformation: A more general form of transformation that includes log and square root as special cases. It can only be applied to positive data.
- Yeo-Johnson Transformation: Similar to Box-Cox but can be applied to data with both positive and negative values.

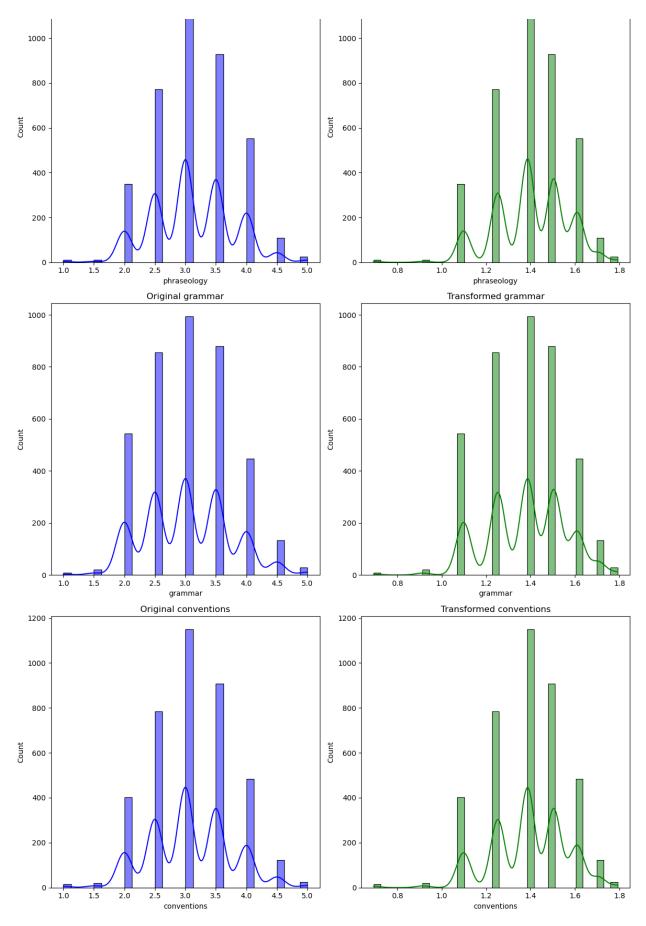
Alternate Strategies:

- Under-sampling; Delete some data from rows of data from the majority classes. In this case, delete 2 rows resulting in label B and 4 rows resulting in label C.
 - Limitation: This is hard to use when you don't have a substantial (and relatively equal) amount of data from each target class.
- Copy rows of data resulting minority labels. In this case, copy 4 rows with label A and 2 rows with label B to add a total of 6 new rows to the data set.
 - Limitation: copying current data and you don't really present anything new. You will get better models, though.
- SMOTE Synthetic Minority Oversampling Technique

- Synthetically generate new data based on implications of old data. Basically, instead of deleting or copying data, you use the current inputs to generate new input rows that are unique but will have a label based on what the original data implies. In the case above, one simple way to think of this idea would be to add 4 rows with label A to the data where the inputs represent total or partial similarities in values to current input features. Repeat this process for 2 rows of label B as well.
- Limitation: If two different class labels have common neighboring examples, it may be hard to generate accurate data representing what each unique label may look like from the input side and therefore SMOTE struggles with higher dimensionality data.

After some experimentation on small sample sizes with SMOTE, Box-Cox Transformation and Log Transformation we came to the conclusion that target rebalancing with logs would be the most appropriate to reconstruct an appropriate distribution.





We have transformed the each target with a log1 transformation, with cohesion as an example below:

df_transformed['cohesion'] = np.log1p(df['cohesion'])

We then compute the Shapiro-Wilk Test, Skewness, Kurtosis, giving rise to the following analysis:

Our transformation has correctly changed the distributions' characteristics, often reducing right skewness but introducing left skewness and altering the tailedness (kurtosis). After these changes, the data now more similarly conforms to a normal distribution according to the Shapiro-Wilk test.

Model Architecture

After this, we decided to choose a pre-trained generative model that understands semantics of language, and then further fine-tuning with our dataset. And additionally from above, we have, the following constraint - we want a model that is robust to non-normality.

After some research and considering a few generative models GPT-3.5/4, Llama, we decided to use BERT instead. This is because BERT has both a cased and uncased version. So, we can experiment in more ways with tokenization.

The following is the model architecture:

- We are using the BERT model. We then added a Dropout and Linear Layer as well. We add these layers to ensure the model is able to regularize and classify the data better.
- In the forward loop, there are 2 outputs from the BERT model layer.
- The output of this, pooled_output is passed through the Dropout layer and then the Linear layer.
- We set the number of dimensions in the Linear layer to be equal to the number of classes we have in the dataset.

This gives is the following output:

```
BERT_Classifier(
  (bert): BertModel(
    (embeddings): BertEmbeddings(
      (word_embeddings): Embedding(28996, 768, padding_idx=0)
      (position_embeddings): Embedding(512, 768)
      (token_type_embeddings): Embedding(2, 768)
      (LayerNorm): LayerNorm((768,), eps=1e-12, elementwise_affine=True)
      (dropout): Dropout(p=0.1, inplace=False)
    (encoder): BertEncoder(
      (layer): ModuleList(
        (0-11): 12 x BertLayer(
          (attention): BertAttention(
            (self): BertSelfAttention(
              (query): Linear(in_features=768, out_features=768, bias=Tr
              (key): Linear(in_features=768, out_features=768, bias=True
              (value): Linear(in_features=768, out_features=768, bias=Tr
              (dropout): Dropout(p=0.1, inplace=False)
            (output): BertSelfOutput(
              (dense): Linear(in_features=768, out_features=768, bias=Tr
              (LayerNorm): LayerNorm((768,), eps=1e-12, elementwise_affine
              (dropout): Dropout(p=0.1, inplace=False)
            )
          )
          (intermediate): BertIntermediate(
            (dense): Linear(in_features=768, out_features=3072, bias=Tru
            (intermediate_act_fn): GELUActivation()
          )
          (output): BertOutput(
            (dense): Linear(in_features=3072, out_features=768, bias=True
            (LayerNorm): LayerNorm((768,), eps=1e-12, elementwise_affine
            (dropout): Dropout(p=0.1, inplace=False)
          )
        )
      )
    (pooler): BertPooler(
      (dense): Linear(in_features=768, out_features=768, bias=True)
      (activation): Tanh()
    )
  )
  (drop): Dropout(p=0.0, inplace=False)
  (out): Linear(in_features=768, out_features=6, bias=True)
)
```

After reading latest consensus, experiments on Huggingface and local experimentation, we came to the conclusion that the following hyper-parameters are best for this model and data:

```
MAX_LEN = 200

TRAIN_BATCH_SIZE = 4

VALID_BATCH_SIZE = 4

EPOCHS = 45

LEARNING_RATE = 2e-05

tokenizer = BertTokenizer.from_pretrained('bert-base-cased')
```

Essay Dissection (Longformer)

This model automatically segments texts and classifies a range of argumentative and rhetorical elements. The primary elements identified by this model include:

- Position: Identifying the main stance or thesis of the essay.
- Lead: Recognizing introductory statements or premises.
- Rebuttal: Detecting responses to counterarguments.
- Claim: Identifying specific assertions or points made in support of the position.
- Evidence: Recognizing data, quotations, or other factual information supporting claims.
- Counterclaim: Detecting opposing viewpoints or arguments.
- Concluding Statement: Identifying concluding remarks or summaries of the argument.

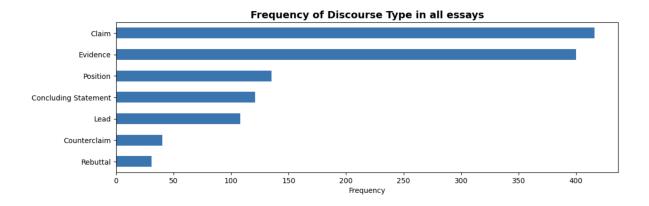
The training data used is taken from the writing feedback dataset here.

Data Exploration

The first step was to explore the initial two datasets. We found the smaller dataset mostly had duplicates of the first one, so we continued with just the larger dataset.

	id	discourse_id	discourse_start	discourse_end	discourse_text	discourse_type	discourse_type_num	predictionstring
0	423A1CA112E2	1.622628e+12	8.0	229.0	Modern humans today are always on their phone	Lead	Lead 1	1 2 3 4 5 6 7 8 9 10 11 12 13 14 15 16 17 18 1
1	423A1CA112E2	1.622628e+12	230.0	312.0	They are some really bad consequences when stu	Position	Position 1	45 46 47 48 49 50 51 52 53 54 55 56 57 58 59
2	423A1CA112E2	1.622628e+12	313.0	401.0	Some certain areas in the United States ban ph	Evidence	Evidence 1	60 61 62 63 64 65 66 67 68 69 70 71 72 73 74 75
3	423A1CA112E2	1.622628e+12	402.0	758.0	When people have phones, they know about certa	Evidence	Evidence 2	76 77 78 79 80 81 82 83 84 85 86 87 88 89 90 9
4	423A1CA112E2	1.622628e+12	759.0	886.0	Driving is one of the way how to get around. P	Claim	Claim 1	139 140 141 142 143 144 145 146 147 148 149 15

The dataset was structured as shown above. There were overall about 15k documents/texts in the training dataset (which we also had access to). The split into the respective rhetorical parts is shown in the visual:



This problem could have been treated as sequence classification problem or a token classification problem. However, we decided to structure it as a token classification similar to POS tagging.

• Using the id, discourse type and prediction string columns we concatenated the texts on their id, each word was tagged with a discourse type.



 Now this text had to be converted to tokens suitable for our model. Since the first layer in our model was going to be a base pre-trained Longformer transformer, we went with the corresponding tokenizer.

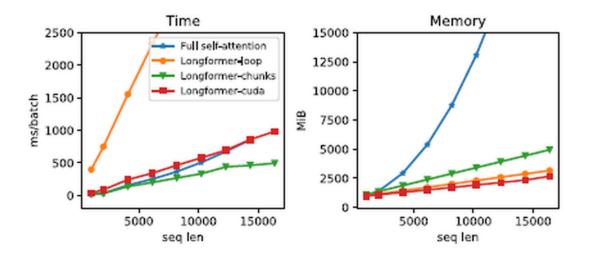
After converting the text to tokens, we needed to carefully align the tags with the tokenized text since some words would be split into multiple tokens.



These tokens would eventually be encoded by the tokenizer along with the label before being passed into the model.

Model Architecture

The model relied on the pre-trained Longformer transformer, with a context-window of 4096 tokens. The Longformer is an advanced version of the Transformer model, optimized for processing long documents. It overcomes the limitations of traditional Transformer models like BERT or GPT, which struggle with lengthy texts due to their quadratic self-attention mechanism. The Longformer uses a "sliding window" attention mechanism, which reduces computational complexity from quadratic to linear by focusing each token's attention on a nearby window of tokens. This design allows the Longformer to efficiently handle texts much longer than standard models, making it ideal for tasks involving large documents, such as legal analysis, long-form summarization, and detailed document review.



The image above shows the difference in memory usage between full self-attention models and Longformer models. Notice the exponential increase in the memory usage of the full-self attention as the sequence length increases. Comparatively, Longformer performs much better.

Our model used to classify long text documents such as essays consists of:

- A pre-trained Longformer layer that can process text sequences of up to 4096 tokens, making it well-suited for lengthy essays.
- A **linear layer** (**dense1**) that reduces the dimensionality from the Longformer's output size to 256.
- A ReLU activation to introduce non-linearity, allowing the model to capture complex patterns.
- A dropout layer to mitigate overfitting by randomly zeroing some fraction of the output units during training.
- A final linear layer (dense2) that maps the reduced representation to the number of target labels (num_labels).

The model outputs logits for each label, which can be converted into probabilities for essay classification. The **forward** method defines the data flow from input to output, utilizing **input_ids** and **attention_mask** to handle and process the input text.

Training Details

After extensive testing on smaller subsets of the data, we decided on the following hyper-parameters:

```
num_epochs = 10
max_seq_len = 800
batch_size = 16
optimizer = torch.optim.AdamW(model.parameters(), lr=1e-5)
loss_fn = nn.CrossEntropyLoss(ignore_index=-100)

# Scheduler to step learning rate
scheduler = StepLR(optimizer, step_size=2, gamma=0.75)

# Early stopping parameters
best_val_loss = float('inf')
epochs_no_improve = 0
patience = 2
```

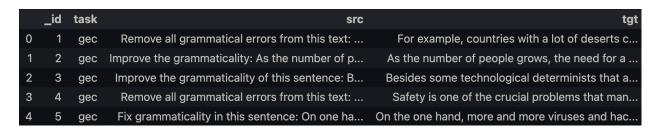
We also employed mixed precision training and data parallelism to decrease training time. You can find the training loss and validation loss over epochs below.

Grammar Correction (Mistral 7B)

Data Exploration

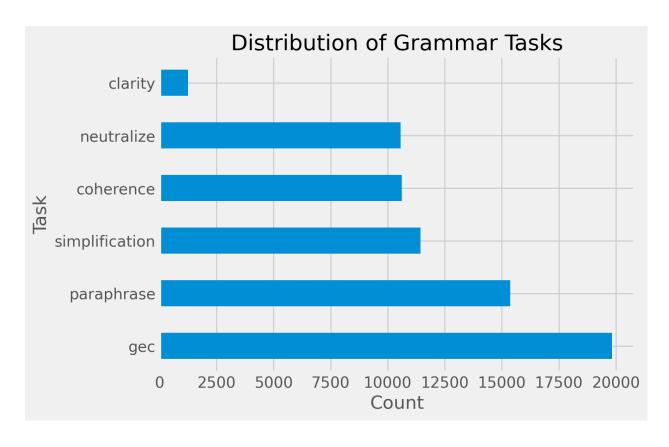
We used the CoEdIT dataset from grammarly for our grammar correction and writing improvement which can be found here.

Here is how the dataset looks like:



For this model, we have the following tasks: grammatical error correction, neutralize, simplification, paraphrase, coherence, and clarity improvement.

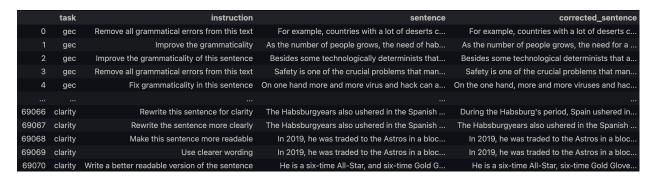
Based on availability of the particular task and relevance for grammatical and writing improvement we chose to only train on neutralize, coherence, and grammatical error correction tasks. The distribution of grammar scenes is as follows:



Here is an example of existing prompts in our dataset for different writing improvement tasks:

	task	prompt
68460	clarity	Use clearer wording
68655	clarity	Make the sentence clear
57765	coherence	Fix coherence in this sentence
67787	coherence	Improve the consistency of the text
4979	gec	Fix errors in this text
13127	gec	Remove grammar mistakes
28548	neutralize	Make this paragraph more neutral
26584	neutralize	Neutralize this sentence
55342	paraphrase	Write a paraphrase for the sentence
47067	paraphrase	Rewrite this text
34231	simplification	Rewrite this sentence for simplicity
35500	simplification	Make this easier to understand

We apply pre-processing to separate the dataset into instruction, input, and response format to prepare it for instruction fine-tuning. After preprocessing the dataset looks like this:



Next, to format the text for a generation pipeline we draw inspiration from the Alpaca Dataset formatting:

Below is an instruction that describes a task, paired with an input that provides further context. Write a response that appropriately the request.

```
### Instruction:
{instruction}

### Input:
{sentence}

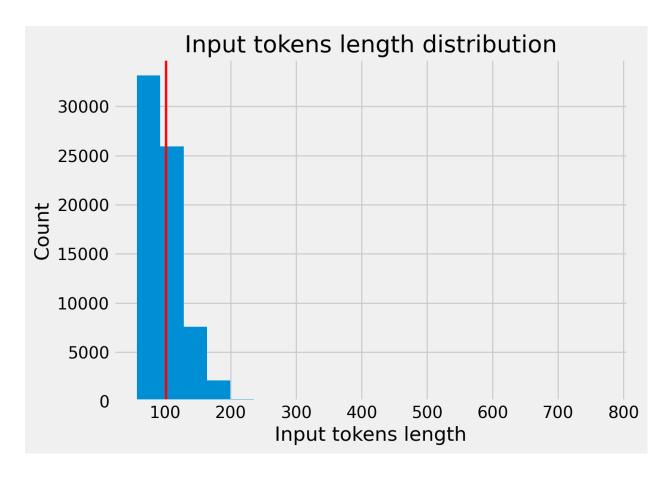
### Response:
{corrected_sentence}
```

Hello there!

Below is an example of the prompt applied to an example sentence.

Below is an instruction that describes a task, paired with an input that
Instruction:
Correct this sentence
Input:
Hellow there!
Response:

We observed a few things in the data that allowed us to make a few decisions about the max input sequence to the model. Here, the figure displays the input token length distribution with the prompt applied to text in the dataset. The mean is depicted as a vertical red line at around 101 tokens:



Given the distribution above, we decided to standardize token_size of 200 based on distribution of input token lengths for sentences (model still takes up to token_size 4096 assuming enough GPU VRAM).

However, during during inference, a token_size greater than 200 may not perform as well so a sliding window of 200 across each sentence may be necessary rather than predicting on the max token size of the particular LLM we are using.

Model Architecture

We fine-tune the following base model - **Mistral-7B-v0.1** (base model). For some context, the Mistral-7B-v0.1 Large Language Model (LLM) is a pretrained generative text model with 7 billion parameters. Mistral-7B-v0.1 outperforms Llama 2 13B on all benchmarks we tested.

Below, we describe some architectural details: It uses a similar architecture to Llama, but with improvements including: sliding window attention, rolling buffer cache, and pre-fill and chunking.

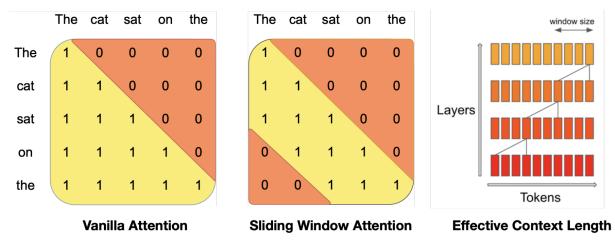


Figure 1: Sliding Window Attention. The number of operations in vanilla attention is quadratic in the sequence length, and the memory increases linearly with the number of tokens. At inference time, this incurs higher latency and smaller throughput due to reduced cache availability. To alleviate this issue, we use sliding window attention: each token can attend to at most W tokens from the previous layer (here, W=3). Note that tokens outside the sliding window still influence next word prediction. At each attention layer, information can move forward by W tokens. Hence, after k attention layers, information can move forward by up to $k \times W$ tokens.

	Timestep i				Timestep $i+1$				Timestep $i+2$				
This is an example of	This	is	an			This	is	an	example	of	is	an	example
Mistral is a good	Mistral	is				Mistral	is	а		Mistral	is	а	good
The cat sat on the mat	The	cat	sat	on		the	cat	sat	on	the	mat	sat	on

Figure 2: Rolling buffer cache. The cache has a fixed size of W=4. Keys and values for position i are stored in position i mod W of the cache. When the position i is larger than W, past values in the cache are overwritten. The hidden state corresponding to the latest generated tokens are colored in orange.

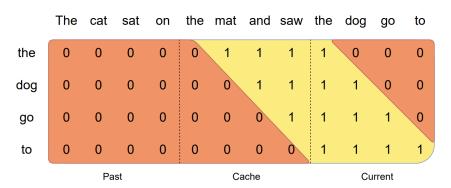


Figure 3: Pre-fill and chunking. During pre-fill of the cache, long sequences are chunked to limit memory usage. We process a sequence in three chunks, "The cat sat on", "the mat and saw", "the dog go to". The figure shows what happens for the third chunk ("the dog go to"): it attends itself using a causal mask (rightmost block), attends the cache using a sliding window (center block), and does not attend to past tokens as they are outside of the sliding window (left block).

We used the base model rather than the instruction-tuned model for chat since it makes more sense for our use-case to instruction-tune our model from the base model backbone (out-of-the-box only generates text, i.e., it does not follow instructions well).

Training Details

Given the model is fairly large (though it is not on the small end for LLM), extensive hyperparameter tuning not possible due to compute restrictions (7B parameter model)

We reviewed accepted literature and guides online to make use of more common settings for hyperparameters:

```
per_device_train_batch_size=2  # Batch size,
gradient_accumulation_steps=1  # No gradient accumulation steps,
num_train_epochs=1  # Fine-tune on entire dataset
learning_rate=2.5e-5  # Want a small lr for finetuning
bf16=True  # Use mixed precision training with bfloat16,
optim="paged_adamw_8bit"  # Use 8-bit AdamW
logging_steps=25  # When to start reporting loss
logging_dir=LOGGING_DIR  # Directory for storing logs
save_strategy="steps"  # Save the model checkpoint every logging step
save_steps=25  # Save checkpoints every 25 steps
evaluation_strategy="steps"  # Evaluate the model every logging step
eval_steps=25  # Evaluate and save checkpoints every 25 steps
```

We fine-tune on a subset of the CoEdIT dataset (around 40000 samples). The backbone of the model was not changed during fine-tuning since the architecture is already very robust for generation tasks.

We used VM instance with 8 vCPUs, 40GB RAM, 24GB VRAM (NVIDIA RTX A5000) to fine-tune the model. We use techniques to significantly improve the efficiency of training with little to no performance sacrifice. The basic premise of the fine-tuning was the following - we utilize a parameter-efficient fine-tuning method (PEFT) called Quantized Low-Rank Adaptation (QLoRA).

In the past, fine-tuning large LLMs was a resource-intensive task, requiring substantial computational power and high-end GPU resources, thus limiting accessibility. Traditional methods involved fine-tuning the model on extensive datasets, followed by a step called 4-bit quantization for the model to function on consumer-grade GPUs after fine-tuning. This approach reduced resource usage but at the cost of the model's full capabilities, leading to compromised results.

QLoRA emerges as a significant advancement within the realm of parameter-efficient finetuning (PEFT). PEFT aims to modify only a small fraction of a model's parameters, making finetuning more efficient and less resource-intensive. QLoRA aligns with this goal by allowing for the efficient fine-tuning of large language models using a single GPU. It maintains the high-level performance of a full 16-bit model even when reduced to 4-bit quantization. For perspective, while traditional methods required something like 780 gigabytes of VRAM to fine-tune a 65 billion parameter model, QLoRA achieves this with just a single 48 gigabyte VRAM GPU. This approach not only fits within the PEFT paradigm but also makes such advanced modeling far more attainable and practical.

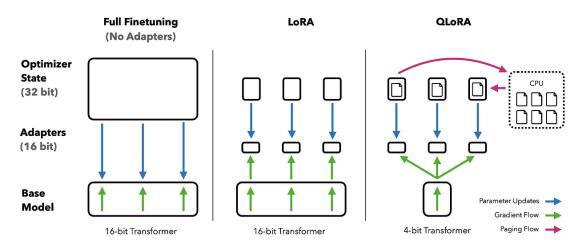


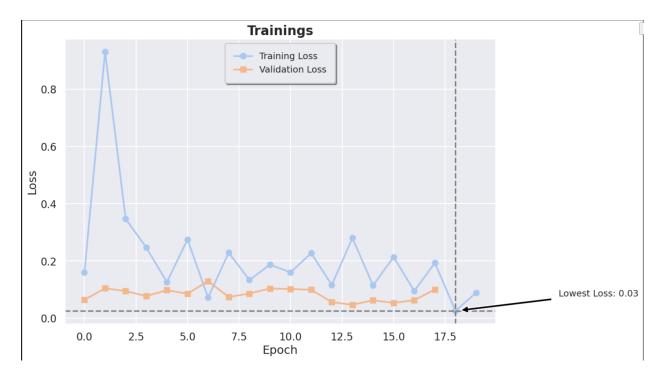
Figure 1: Different finetuning methods and their memory requirements. QLORA improves over LoRA by quantizing the transformer model to 4-bit precision and using paged optimizers to handle memory spikes.

Evaluation

Feedback Scores (Bert-cased)

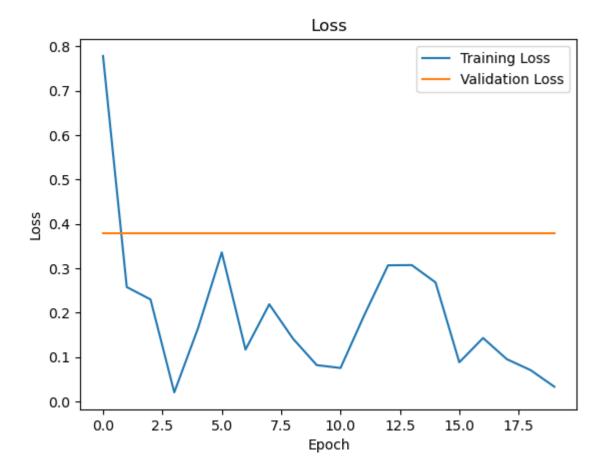
Train and Validation Performance

The train and validation loss of the model can be seen below:



From this we can see that the model is performing quite well on the out-of-sample data, in-fact, it is performing better on the samples than on the training data itself.

Additionally, we tested two different bert models - cased and uncased. Testing on both models, it was clear that the cased model was better:



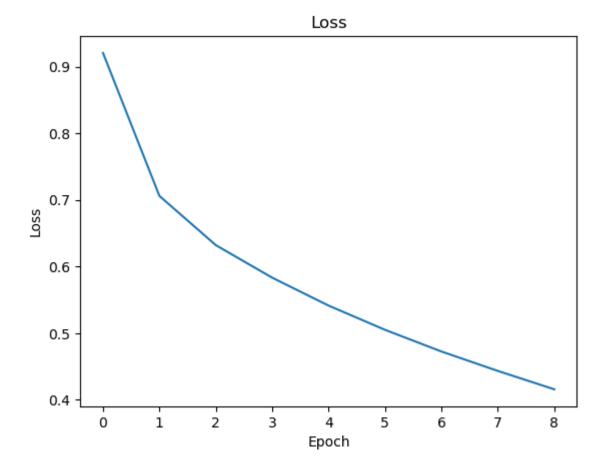
From the above, we concluded that the casing of a word, and by extension, how casing is applied within the context of a sentence alters its correctness on various metrics, and by using a uncased model and grouping these words together we were losing this signal.

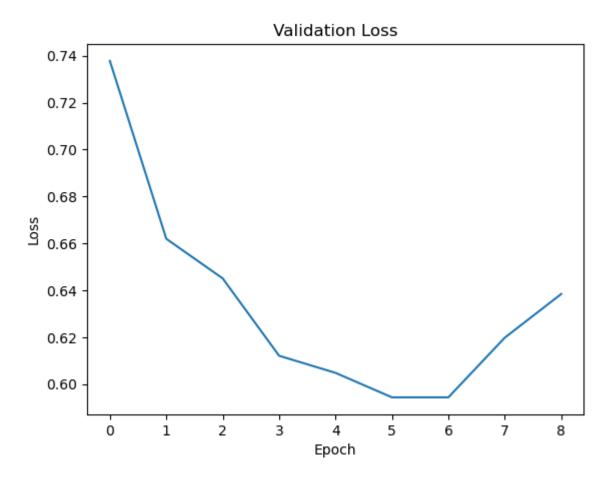
Hence, we finally decided to stick with the cased model with the hyper-parameters discussed above.

Essay Dissection (Longformer)

Train and Validation Performance

For this model, we compute the cross entropy loss. We compute the validation with the following formula: number of correct predictions / total number of tokens. Our model had an **accuracy of about 70%** on the testing data. Some of the performance curves can be seen below:

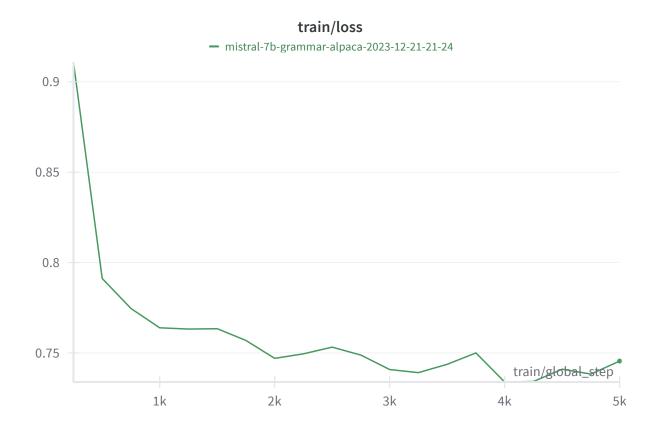


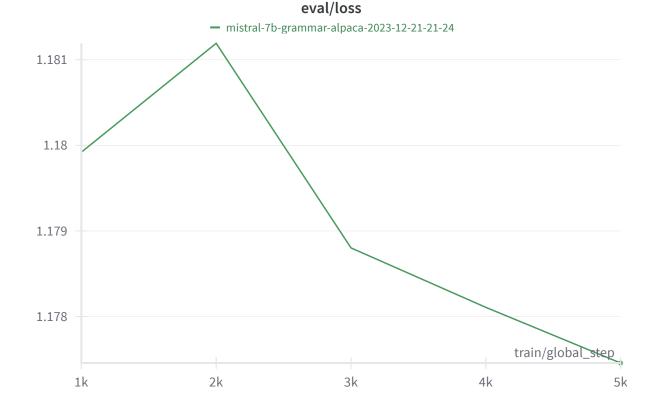


Grammar Correction (Mistral 7B)

Train and Validation Performance

The train and validation loss of the model can be seen below:





Note for the following before-after tests the final output is post-processed by cutting off the next sequence of conversations (prevent model from continuing indefinitely or until token limit)

Benchmark Performance vs Fine-tune Performance

We compare the performance of the model before fine-tuning on the dataset (benchmark) and the fine-tuned model. In the figures below are the content after the ### Response tag are generated:

Before Fine-tuning

```
### Instruction:
Remove grammar mistakes
```

Input:

NLP, it stand for Natural Language Processing, is a field in computer sc

Response:

NLP, it stands for Natural Language Processing, is a field in computer s

After Fine-tuning

Instruction:
Remove grammar mistakes

Input:

NLP, it stand for Natural Language Processing, is a field in computer sc

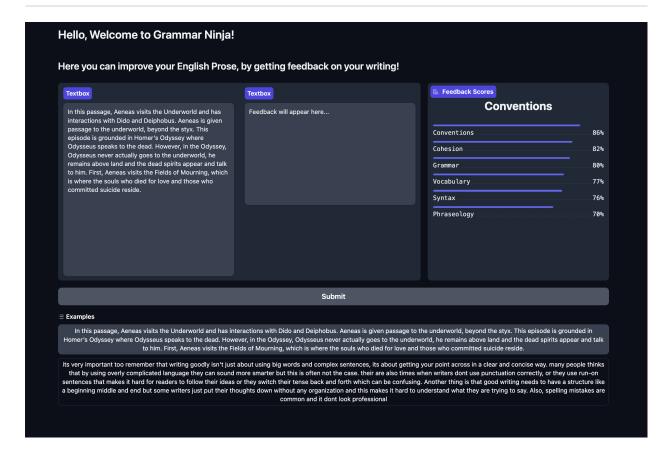
Response:

NLP, it stands for Natural Language Processing, is a field in computer s

As can be seen, fine-tuning improves grammar correction of LLM compared to out-of-the-box.

Presentation of Results

Feedback Scores (Bert-cased)



For the feedback score model, we have the above frontend to display the results as percentages. We planned to combine all three models into this UI, but due to time constraints were not able to complete this.

Inference Scripts

We created inference scripts that can be used to test our trained models. They can be accessed on our repo by going to the tools folder and visiting one of the three models. More details are provided on the README of our GitHub repo for setup.

Here is an example text that we use the test our trained models:

```
cat ../../examples/nlp.txt
```

NLP, it stand for Natural Language Processing, is a field in computer so In field of NLP, machine learn algorithms is used for make computers can

Below our the outputs of our model for the example sentence.

Feedback Scores (Bert-cased)

```
cd feedback_scores
python inference.py ../../examples/nlp.txt
```

```
[3.84, 3.84, 3.49, 3.71, 2.77, 3.56]
```

Essay Dissection (Longformer)

```
cd essay_dissection
python inference.py ../../examples/nlp.txt
```

```
{'Lead_1': ['NLP,', 'it', 'stand', 'for', 'Natural', 'Language', 'Processing,', 'is', 'a', 'field', 'in', 'computer', 'sci ence,', 'where', 'focus', 'on', 'how', 'computers', 'can', 'understanding', 'and', 'interact', 'with', 'human', 'language.', "It's", 'goal', 'is', 'to', 'make', 'computers', 'can', 'understand', 'and', 'respond', 'to', 'text', 'or', 'voice', 'd ata.', 'But,', "it's", 'hard', 'because', 'languages', 'is', 'very', 'complex', 'and', 'have', 'many', 'rules', 'that', 'o ften', 'not', 'follow', 'logic.', 'In', 'field', 'of', 'NLP,', 'machine'], 'Position_1': ['learn', 'algorithms, 'is', 'us ed', 'for', 'make'], 'Claim_1': ['computers', 'can', 'process', 'and', 'analyze', 'large', 'amounts', 'of', 'natural'], 'E vidence_1': ['language', 'data.', 'The', 'problems', 'is', 'that,', 'even', 'with', 'advanced', 'algorithms,', 'computers', 'often', "don't", 'understand', 'the', 'nuances,', 'like', 'sarcasm', 'or', 'idioms,', 'in', 'human', 'languages.', 'So, ', 'many', 'times,', 'they', 'makes', 'errors', 'when', 'they', 'tries', 'to', 'interpret', 'what', 'a', 'human', 'is', 's aying', 'or', 'writing.']}
```

Grammar Correction (Mistral 7B)

Note that this model takes a while to run if you are not using GPU
 since it involves multiple sliding windows if text is too long

```
cd grammar
python inference.py ../../examples/nlp.txt
```

NLP, it stands for Natural Language Processing, is a field in computer science where the focus is on how computers can understand and interact with human language.

Its goal is to make computers can understand and respond to text or voice data. I'm not sure if this is the correct way to write it, but I hope it's close. I'm not sure if this is the correct way to write it, but I hope it's close.

But it's hard because languages are very complex and have many rules that often don't follow logic. But it's hard because languages are very complex and have many rules that often don't follow logic. But it's hard because languages are very complex and have many rules that often don't follow logic.

In the field of NLP, machine learning algorithms are used to make computers can process and analyze large amounts of natural language data.

The problem is that, even with advanced algorithms, computers often don't understand the nuances, like sarcasm or idioms, in human languages. They can't understand the meaning of the words.

So, many times, they make errors when they try to interpret what a human is saying or writing. So, many times, they make errors when they try to interpret what a human is saying or writing. So, many times, they make errors when they try to interpret what a human is saying or writing.

Future Work

For the BERT model specifically, we had some thoughts on how improve the model selection and the training process. Ideally, we would like to work with larger datasets and perform Hill Climbing because it can take lots of models and pick the best small subset of models. Additionally, we could add more specificity to the loss function - custom different loss rate per target; 2 stage pooling. First pool either words, sentences, or paragraphs and then also using different max_lens for training and inference. Also, it seems like there could be merits to doing some form of ensemble models though architecting this might be different with pre-trained large language models.

In the essay dissection model, we discussed utilizing a dual-Longformer setup that could potentially enhance our model's accuracy as seen here. To counteract the skewed learning from imbalanced datasets, we also discussed implementing data rebalancing strategies such as SMOTE, which generates synthetic samples for minority classes, thus enhancing the training process.

In our grammar improvement LLM, we currently combine a set of prompts (clarity, coherence, grammatical error correction, and neutralize the sentence)

We decided to not fully utilize the entire CoEdIT dataset, since we thought that it wouldn't directly relate to improving a sentence. These additional capabilities can be trained by including them in future fine-tuning runs and manifested in a separate UI that allows a user to paraphrase and/or simplify their writing.

We also used a barebones CLI inference script. In the future, given more time, we would package the three inference scripts as a Gradio space to make our demo more interactive.

Provided Code

https://github.com/rvineet02/cs505_final_project

Contributions

In the milestone planning, we had defined clear roles to complete the project.

As described, we each decided on one aspect/model of the problem. In this regard, we were responsible to deliver the model from end-to-end. Vineet built the BERT model for the feedback scores, Alex built the Mistral-7B model for generating the corrected version of a given input sentence and Dhruv built the LongFormer model to tag the input text into classifications of writing. This is including the inference for the models as well—we each were responsible individually for the inference scripts for our models. We then, together, decided on the UI of the frontend and how we would like to design the pipeline for the full end-to-end frontend experience. Vineet built the Gradio frontend for the Feedback Score model. Alex refactored the repo and modularized the project into a custom python package <code>grammar_ninja</code>. He also created 3 CLI inference scripts for each of the models to demo each model's performance. Additionally, Alex and Vineet added finishing touches on the final report document.