# Datenanalyse im Unternehmen

# Christopher Keibel 30341906 South Westphalia University of Applied Sciences

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### Abstract

This work is being done in the context of module *Datenanylse im Unternehmen* at the *South Westphalia University of Applied Sciences*. The goal of the project is to design a *retrieval augmented generation* pipeline in which the models for retrieving documents and those for generating answers can be easily swapped. The pipeline evaluates the individual components and their interaction as a closed system. In addition, an optional keyword retriever is provided for the retrieval of documents in order to implement and evaluate a hybrid search. At the beginning of the work, the individual components are explained in a theoretical section. This is followed by the methodology part, in which the implementation and evaluation are described.

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# 1 Introduction

Since the arrival of ChatGPT, generative AI algorithms have made their way into the mainstream media and the everyday lives of many people. Every day, numerous people use the chatbot ChatGPT developed by OpenAI to make everyday tasks easier. Behind its technological success is an algorithm called Transformer [21], an encoder-decoder network that can efficiently process long text passages. To be precise, ChatGPT utilizes only the decoder[14] component of the original Transformer.

In [14, 15], OpenAI shows that language models can learn different text-based tasks through generative pre-training (causal language modelling). The generative pre-training is done on an unlabelled dataset and the goal of the language model during language modelling is to predict the next token given a sequence of tokens as input. In [9], OpenAI shows that scaling these language models has a significant impact on the capabilities and performance. The capabilities of these models can be improved by further a so-called instruction fine-tuning [23] for specific tasks and by using reinforcement learning from human feedback [12] to more closely match the output to human intent.

Although models trained in this way may have impressive capabilities, they still have a serious problem. The knowledge acquired during the training of these models is stored in the model parameters[7]. Therefore, it is not possible to draw conclusions about the accuracy of the generated outputs. Here we often speak of model *hallucination*, as it may generate outputs with a high confidence that may not be factually correct.

Another issue is the knowledge cutoff problem. This refers to the limitation of models to learn only the information present in their training data. Despite being trained on a vast corpus of data, their knowledge is restricted to the date on which the data was last collected. To acquire new knowledge, these models have to be re-trained at some expense.

In [7], a framework called retrieval augmented generation (RAG) is proposed, which is intended to solve these problems. By adding a non-parameterized memory in the of a vector database, the knowledge of the language model is augmented.

The workflow of this framework is defined by three main components. In addition to the generative model, another *language model* which is called *retriever*, is used to search the additional knowledge that is stored in the vector database on the basis of a query. The found information and the query are then passed to the third component, the generative *language model*, which then answers the query based on the extracted information.

Unlike online web search engines such as *Google* or *Bing*, this setup also allows users to perform searches on their own data by indexing it using a vector database.

This work discusses and evaluates the various components and the workflow of the *retrieval augmented* generation framework. The capabilities of this technology are illustrated and possible drawbacks and problems are outlined. Section 2 provides a theoretical basis for the technology, which is a prerequisite for understanding and implementing the project described in section 3.

# 2 Retrieval Augmented Generation (RAG)

As we have learned in section 1, the retrieval augmented generation framework augments the context of our prompts with relevant information to address the knowledge limitations of generative models.

 $<sup>^{1}</sup>$ Reuters: ChatGPT sets record for fastest-growing user base (visited 04.03.2024)

In this chapter, we will take a detailed look at the components of the RAG framework and how they interact with each other to overcome the limitations of generative models.

# 2.1 Retrieving relevant information

A critical part of making this framework work well, is finding relevant information to a user query. If the information passed to the generative language model is not relevant to the query, the generative model will not be able to return a qualified answer. The search for relevant information therefore seems to be one of the most important components of the framework for it to be successful. In a RAG setup, the system receives a user query in the form of a string and must find relevant information from a  $data\ store^2$  based on this string. Since the query is delivered in string form, it is not easily possible to perform a classic SQL database query to find the desired information. To be able to easily find and store data in the form of a string, we make use of  $natural\ language\ processing\ (NLP)$  algorithms to convert given texts into vectors.

Transforming texts and documents<sup>3</sup> into vectors and thus into a Euclidean vector space with same dimension helps to perform meaningful retrieval operations [18]. A Euclidean vector space has the property that the similarity between vectors within it can be calculated. We make use of this property in information retrieval to map documents and user queries in the same Euclidean space and calculate the similarity between them. A similarity function like cosine similarity indicates how close a query vector is to a document vector in the vector space. The closer together these vectors are, the more likely we assume that they have a similar meaning. It is therefore important to find a good method that maps similar documents into a similar vector representation, so that they are close to each other in the vector space.

#### 2.1.1 Vector representations of text

There are different ways to represent texts as vectors, which can be divided into two fundamental categories, which will be used in section 3 to perform *information retrieval*. The first category are so called *sparse vectors*, which are commonly used for keyword searches. The second category are *dense vectors*. Dense vectors aim to encode the *semantic meaning* of a document without being limited to exact keywords.

# Sparse vector representation

To represent documents as sparse vectors, extensions of a bag-of-words model are often used. In a bag-of-words model, unique words are considered independently of each other, and their respective occurrences are counted. [1, pp. 13ff]. When splitting documents into smaller units such as words or sub words, we also speak of tokens<sup>4</sup>. The dimension of the resulting vectors is equal to the number of our unique words in our entire corpus, which is the size of the vocabulary used by the bag-of-words model. Each component of a vector is clearly assigned to a word from the vocabulary. The exact values

 $<sup>^2</sup>$ A data store is a collection of information that is searched by the RAG framework. The information can be stored in a database or even in a simple list.

<sup>&</sup>lt;sup>3</sup>I will use the words document and text interchangeably in this work, both mean the same thing in the context of this work.

<sup>&</sup>lt;sup>4</sup>In the context of *bag-of-words* models, I continue to write about words or terms instead of *tokens*, as these work on a word/ term basis.

of the vector components are created when a document is *vectorized*. For each *word*, the occurrence frequency in the document is counted and stored in the associated vector component as shown in fig. 1.

	Term 1	Term 2	 Term N
Document 1	3	0	 2
Document 2	0	5	 1
Document N	2	1	 0

Figure 1: Bag-of-words model - The column headers contain the unique words from our vocabulary. The table has a separate row for each document. Each column stores the frequency with which a term occurs in each document.

Using the resulting row vectors (document vectors) from fig. 1, we can construct our *vector space* for the search. User queries need to be *vectorized* in the same way (word counts per term over the entire vocabulary) so that we can then use *cosine similarity* to find matching documents in the *vector space*.

As can be seen in fig. 1, a small number of documents can result in *sparse vectors* (vectors with many zero components). *Sparse vectors* take up an unnecessary amount of memory. To implement a *bag-of-words* model efficiently, it is often done in the form of an *inverted index*. An *inverted index* stores a tuple consisting of a *document id* and the occurrence count for each term. If a term does not occur in a document, it is not stored for the current term.

However, simply counting the words in a document for *information retrieval* can be problematic. The vectors obtained in this way do not take into account the *importance* or *rarity* of a *word* in the entire document. Texts often consist of many words that do not contribute much to the information content of the document; these words are referred to as *stopwords*. A first extension of the *bag-of-words* model is *tf-idf*, short for *Term Frequency Inverse Document Frequency* [3, pp. 7ff]. The goal of *tf-idf* is to calculate some relevance for document with respect to a user query. *Tf-idf* adds a weighting to each word of the *vector representation* to capture the meaning of a word in a document relative to the frequency of the word in the entire corpus (all documents).

$$tfidf_{t,d} = tf_{t,d} \cdot idf_{t,D}$$
 where:  

$$t = \text{Term (word,)}$$

$$d = \text{document,}$$

$$D = \text{corpus (all documents.)}$$
(1)

Term Frequency counts the number of times a word appears in a document, with higher occurrences being more significant. The term frequency is then normalised to obtain the relative frequency, considering the length of the document.

$$tf_{t,d} = \frac{\sum_{t:t \in d} 1}{\sum_{i=1}^{|d|} 1}$$
 where: 
$$t: t \in d = \text{term (word) in document,}$$
 
$$|d| = \text{length of the document (all terms).}$$
 (2)

Inverse Document Frequency counts how often the word occurs in the entire corpus.

$$idf_{t,D} = log \frac{N}{\sum_{D:t \in D} 1}$$
 where: (3) 
$$N = \text{number of documents},$$
 
$$D: t \in D = \text{term (word) in document}.$$

If a word occurs frequently in the entire corpus, it is possibly a stopword and must be weighted lower. Tf-idf considers the length of a document, as longer documents contain more words, resulting in certain words appearing more frequently than in shorter documents. If a long document and a short document have the same Tf-idf score, the long document is weighted lower.

One problem with tf-idf is that  $term\ frequency\ (tf)$  has a very strong influence on the resulting  $relevance\ scores$ . If a word from a user query occurs very often in two different documents from the corpus, we do not know whether the document in which it occurs more often is actually more relevant to the query in the end.  $BM25\ (best\ match\ 25)$  is an improved ranking model that takes this into account by calculating the log of tf. The  $\log tf$  in BM25 rewards term frequency at low values like tf-idf, and as the tf increases, this reward flattens out.

$$BM25(Q,D) = \sum_{i=1}^{n} IDF(q_i) \cdot \frac{TF(q_i,D) \cdot (k+1)}{TF(q_{i,D}) + k \cdot (1-b+b \cdot \frac{|D|}{avgdl})}$$
 where: 
$$Q = \text{user query},$$
 
$$D = \text{document},$$
 
$$n = \text{number of terms in query},$$
 
$$q = \text{actual term},$$
 
$$|D| = \text{document length},$$
 
$$avdgl = \text{average document length},$$
 
$$k \text{ and } b = \text{are hyperparameters}.$$
 
$$(4)$$

#### Dense vector representation

In sparse vectors as described above, we look for an exact term match and describe documents by terms and how important they are to the document. However, these sparse vectors do not encode any semantic characteristics. This can quickly lead to problems if keywords are not included in a query to find the relevant document we are looking for. Sparse vector search relies heavily on a exact term match to find relevant documents.

To represent semantic characteristics without relying too heavily on keywords, we utilise machine learning models to learn and encode dense vectors. In order to learn these vectors through machine learning models, we make use of the distributional semantics theory [4]. According to the distributional semantics theory, words that appear in similar contexts tend to have a similar meaning.

The Word2vec models help to explain how the distributional semantics theory was used to obtain dense vectors encoded with semantic meaning [10]. Word2vec utilises two simple artificial neural network architectures: Continuous bag of words (CBOW) and the skip-gram model. CBOW takes a sentence with a masked word as input and predicts the masked word based on the context. The skip-gram model, on the other hand, predicts the context words given a word. Both models learn to represent words in relation to their context. However, the word vectors learned in this way have a limitation. Each word has only one representation. In natural language, however, the same words can have different meanings, which also depend on the context and word order. Transformer models, as used in this work, take advantage of the position of a word (positional encoding) in a sentence to better understand the meaning of the word in a given context and to allow different representations of the same word [21]. This section explains how the encoder part of the transformers learns its embeddings<sup>5</sup>, which will be used for information retrieval later.

In order for the *encoder* to process text in string form, it is necessary to convert it into numerical values. This is achieved by tokenizing<sup>6</sup> the documents at the start. Tokenization is carried out by a tokenizer that learns how to split texts. The rules that the tokenizer learns to split the text determine the vocabulary. Texts are often divided into sub-words. In the original BERT paper, WordPiece was used for tokenization, which generated a vocabulary of 30,000 tokens [6]. Each token is assigned a unique, nurmeric id. After tokenization, the resulting ids are transformed into vectors using an embedding layer [20, p. 65]. This layer acts as a lookup table, assigning a vector to each token id. The embedding layer is essentially a matrix with dimensions equal to the vocabulary size and target size of our vectors. Next, positional encoding is added, a mechanism to better understand the word order of the input. There are different ways to add positional encoding to the embedded vectors. The original transformer used sine and cosine functions of different frequencies. To add positional encodings to the embedded vectors, we create for each input word another vector of the same dimension as the embedded vectors. The vector components are defined by a formula (5) that takes into account the alternation of sine and cosine curves. Each component is represented by a separate sinusoid, with even-indexed components represented by a sine function and odd-indexed components represented by a cosine function. The position of the word in the input determines the part of the sinusoid from which the values are taken.

 $<sup>^5{</sup>m The~term}~embeddings$  or word~embeddings is frequently used when referring to data that has been encoded in lower-dimensional vector~space.

<sup>&</sup>lt;sup>6</sup> Tokenization is the process of splitting text into smaller units called tokens.

$$PE_{(pos,2i)} = sin(pos/10000^{2i/d})$$

$$PE_{(pos,2i+1)} = cos(pos/10000^{2i/d})$$
(5)

where d = dimension of embedded vector.

The vectors generated represent the *position* of the words in the input relative to the rest of the input. These vectors are then added to the *embedded vectors*.

In [21], the generated vectors are fed into a multi-head attention block comprising several self-attention mechanisms. The self-attention mechanism involves creating three representations of each embedded vector by passing them through a linear layer. These triplets are referred to as Key (K), Query (Q), and Value (V). The attention weights are then calculated using the scaled dot-product formula based on this new representation. The scaled dot-product formula takes the dot-product of the query and key representations, producing attention scores, which are then scaled by the dimension of the vectors to avoid large values during training [20, pp. 62ff]. Finally, the scaled attention scores are normalised using a softmax function. The attention weights obtained are then multiplied by the corresponding value representation of the triplet.

$$Attention(Q, K, V) = softmax(\frac{QK^{T}}{\sqrt{d}})V$$
 where d = dimension of vector. (6)

This process defines a single attention head. The advantage of multiple attention heads is, to be able to focus on various semantic characteristics of the input. The mechanism is comparable to that of convolutional neural networks, where the filters learn to detect different types of edges in images.

When multiple attention heads are used, their outputs are concatenated so that we again obtain a single representation. To prevent this from becoming too large, the output dimension of the attention heads can be a factor of the dimension of the input vectors. BERT, for example, uses 12 attention heads with a vector dimension of 768. The partial output dimensions of the attention heads then have the dimension 768/12 = 64 [21].

After our *embeddings* have been processed by the *multi-head attention* Block, we receive so-called *contextualised embeddings* for each *token* of the input sequence. The *contextualized embeddings* are then processed independently by a normal *feed forward network*.

The output is an *embedded vector* for each *token* in the input sequence. As the input sequences can vary in length, the output will have a corresponding number of vectors. In order to be able to perform the search in *vector space*, however, we want to represent our input (documents and queries) with just a single vector. To obtain a single vector representation, a pooling operation, such as *mean pooling*, can be performed over the vectors. This results in single values for each dimension in the final vector representation.

BERT is a model that is suitable for representing natural language. BERT is often extended with task-specific heads, e.g. for classification or sentiment analysis. Another use case is to compare a pair of inputs for similarity, as required in information retrieval. To achieve this, both inputs must be fed into the model simultaneously, allowing the attention heads to work on them in parallel (cross-attention). However, this process has proven to be computationally intensive [17]. In [17], an

architectural modification called *SBERT* (Sentence *BERT*) is proposed that reduces this computational overhead while maintaining the same accuracy. *SBERT* uses a siamese network structure, in which weights are shared to generate input embeddings separately. These embedded inputs can then be compared using cosine similarity.

In addition, SBERT was fine-tuned with a multiple negative ranking loss. It ensures that the vector representations are optimized for semantic similarity. With the multiple negative ranking loss, we have a triplets consisting of three input sentences. One is called the anchor. In addition, we have a similar sentence (positive sentence) to the anchor and an opposite sentence (negative sentence) to the anchor. The objective of the multiple negative ranking loss is to produce vector representations that bring the anchor and the positive sentence closer together in the vector space while pushing the negative sentence further away.

## 2.1.2 Retrieving relevant document vectors

In section 2.1.1 we learnt how to convert vectors into meaningful representations in order to search for them in a *vector space*. In this section, we will look at how to retrieve relevant document vectors that are close to a query vector, to achieve satisfactory results.

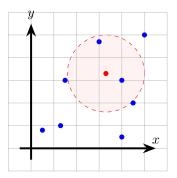


Figure 2: Here we can see a simplified 2-dimensional search space. The blue dots represent our documents, the red dot represents a user query. The goal of a search in the *vector space* is to find documents that are close to a user query.

In fig. 2 we see a simplified representation of *vector search* in a 2-dimensional *vector space*. The blue dots represent *vectorized* documents and the red dot represents a user query. The goal of a *vector search* is to retrieve nearby documents, shown here as a red search radius.

#### Distance measures

There are several different metrics to compute a score for vector similarity. Chroma, the vector store used in this project, offers three different metrics: cosine similarity, the inner product of two vectors and the euclidean norm ( $\ell^2$ -norm). In order to use and evaluate these correctly, it is important to understand how they work.

To calculate the **cosine** similarity between two given vectors  $\vec{a}$  and  $\vec{b}$ , we calculate the angle  $\theta$  between  $\vec{a}$  and  $\vec{b}$  as we can see in fig. 3. The cosine of  $\theta$  is then calculated to obtain an similarity score for the two vectors  $\vec{a}$  and  $\vec{b}$ .

$$\cos \theta = \frac{\vec{a} \cdot \vec{b}}{||\vec{a}|| \cdot ||\vec{b}||} \tag{7}$$

Cosine has the property that the cosine of equal vectors is 1 and 0 if they are orthogonal [19]. Therefore, if we want to obtain matching documents for our query, we look for *cosine similarity scores* that are close to 1.

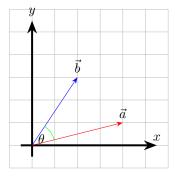


Figure 3: In this figure we have the two vectors  $\vec{a}$  (red) and  $\vec{b}$  (blue). Between them we have the angle  $\theta$  (green). We calculate the cosine of  $\theta$  to get our *similarity score* of  $\vec{a}$  and  $\vec{b}$ .

To illustrate this in an example, let's look at the three sentences "Hello World", "Hello Mars" and "Transformers are awesome" and calculate the similarity score between them using cosine similarity. To obtain vectors for these three sentences, we embed them with an encoder transformer model. We then use the cosine similarity formula to calculate the similarity score between them. As we can see in fig. 4, the cosine similarity for identical sentences is 1, for similar sentences it is less than 1 and the more dissimilar they are, the closer the score is to 0.

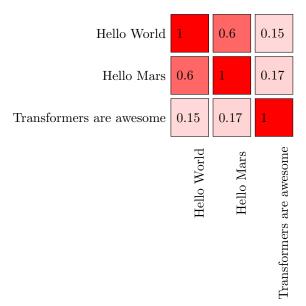


Figure 4: In this heatmap (confusion matrix), we can see that identical sentences receive a score of 1. If the sentences are slightly different, such as "Hello World" and "Hello Mars", the score decreases. The less similar the sentences are, the closer the score gets to 0. So we know that cosine similarity scores can be between 1 and 0.

The *inner product* of two vectors  $\vec{a}$  and  $\vec{b}$  is a generalised form of the *dot product*. It is calculated by multiplying the components of vectors  $\vec{a}$  and  $\vec{b}$  and then summing the result.

$$ip = \sum_{i=1}^{n} a_i \cdot b_i$$
, with  $a_i \in \vec{a}$  and  $b_i \in \vec{b}$  (8)

fig. 5 demonstrates how scores are derived from the inner product geometrically. In fig. 5, the similarity with vector  $\vec{a}$  is calculated using vectors  $\vec{b}$  and  $\vec{c}$ . The vectors  $\vec{b}$  and  $\vec{c}$  are projected onto vector  $\vec{a}$ , and the length of the resulting projection determines the similarity score of the inner product [5, pp. 10ff]. It is evident that vectors that are farther apart produce smaller projections and, therefore, have a lower score. For vectors of equal length, the inner product results in identical values as the cosine similarity.

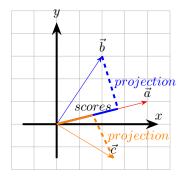


Figure 5: This figure shows geometrically how the *inner product similarity scores* are calculated. The similarities between vectors  $\vec{a}$  and  $\vec{b}$  and vectors  $\vec{a}$  and  $\vec{c}$  are calculated. Vectors  $\vec{b}$  and  $\vec{c}$  are projected onto  $\vec{a}$ . The length of the respective projection on the vector  $\vec{a}$  then determines the *inner product similarity score*.

The **Euclidean distance**, or Euclidean norm ( $\ell^2$ -norm), can be used to measure the distance between vectors in a vector space. To calculate the Euclidean distance between the vectors  $\vec{a}$  and  $\vec{b}$ , the length of a vector  $\vec{c}$  is calculated. Vector  $\vec{c}$  starts from vector  $\vec{a}$  and points to the end of vector  $\vec{b}$  as shown in fig. 6. This gives the following formula:

$$d = \sqrt{\sum_{i=1}^{n} a_i - b_i}, \text{ with } a_i \in \vec{a} \text{ and } b_i \in \vec{b}.$$
 (9)

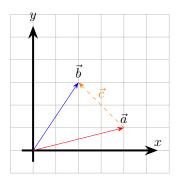


Figure 6: The *Euclidean distance* measures the distance between two vectors  $\vec{a}$  and  $\vec{b}$ . The distance can be represented by a vector  $\vec{c}$  whose length must be calculated to determine the distance between  $\vec{a}$  and  $\vec{b}$ .

Unlike cosine similarity or the inner product, smaller values for Euclidean distance mean a better similarity score because it measures the actual distance between two vectors.

# 2.2 Generating text

After our retriever (model that embeds documents and queries) has retrieved relevant documents with the help of a similarity function, it is now time to generate text based on these documents that answers the user query. This is done by another language model. The model receives the documents and the user query in a structured (prompt) form in order to answer the question. The language model uses the decoder architecture of the original transformer [14]<sup>7</sup> and, as indicated in section 1, the goal of the language model is to predict the next most probable token given the input sequence.

$$p(x) = \prod_{i=1}^{n} p(s_n | s_1, ..., s_{n-1})$$
(10)

This language model can also be seen as a document completion engine, it receives a sequence of tokens as input and generates new tokens until it predicts an end-of-sequence token (EOS). This is an interative task because until the model predicts the eos-token, all previous generated tokens are appended to the input sequence and the next token in the sequence is predicted. Through special task-specific fine-tuning, the model learns a probability distribution over the token vocabulary that matches more likely our task.

# 3 Methodology

This section of the report describes the practical part of the project. As presented in section 2, several components work together to implement a complete RAG system. Only the most important parts of the implemented code are described.

# 3.1 Components

Python classes have been implemented for the retrievers (encoder and bm25) and for the causal language model (decoder). For components working on the basis of a neural algorithm, a \_\_call\_\_ method was implemented to allow simple inference calls. In addition, a separate vector store has been implemented from scratch to make hybrid search easier to handle.

# 3.1.1 Encoder and BM25/Retriever

#### Encoder

The EncoderModel class is responsible for *embedding* and *searching* in the *dense vector space*. It is initialized with a model\_id and an optionally device parameter which determines on what *device* (cpu or cuda:<Index>) the model should work on.

```
class EncoderModel:
    def __init__(self , model_id , device=None):
        self.model_id = model_id
        fallback_device = "cuda" if torch.cuda.is_available() else device
        self.device = device if device else fallback_device
```

<sup>&</sup>lt;sup>7</sup>This does not always have to be the case. T5 [16] from Google, for example, is a complete sequence-to-sequence transformer.

The \_\_call\_\_ function implements the inference call. The input document is tokenized and the model creates the token embeddings. Mean pooling is performed on the token embeddings to obtain the document embeddings. The document embeddings are then normalized.

```
def __call__(self, document):
    tokenized_document = self.tokenize(document)
    outputs = self.model_inference(tokenized_document)
    embeddings = self.mean_pooling(
        outputs, tokenized_document["attention_mask"]
    )
    torch.cuda.empty_cache()
    return F.normalize(embeddings, p=2, dim=1).cpu().numpy()
```

#### **BM25**

The BM25Model uses the BM25Okapi algorithm from the rank\_bm25 package to search and embed in the sparse vector space. The class is initialised without specifying any parameters. The important methods of this class include the add\_documents and search functions. The add\_documents function initializes the BM25Okapi algorithm. Before this, the function cleans the document using a WordNetLemmatiser from the nltk package and tokenizes the document into words using the word\_tokenize function from the nltk package.

```
def add_documents(self, documents):
    cleaned_docs = self.clean_documents(documents)
    self.bm25 = BM25Okapi(cleaned_docs)
```

The search function of the BM25Model class receives a query as input and cleans and tokenizes it in the same way as the add\_documents function. The bm25 scores are then calculated on the tokenized corpus and returned.

```
def search(self, query):
    cleaned_query = self.clean_string(query)
    scores = self.bm25.get_scores(cleaned_query)
    return scores
```

## 3.1.2 Decoder/ Autoregressive model

The DecoderModel class is responsible for the *causal language model* and therefore for *generating answers*. The class is initialised via the model\_id parameter and the optional generation\_config and device parameters.

```
class DecoderModel:
    def __init__(self , model_id , generation_config=None, device=None,
        **kwargs):
        assert isinstance(
        generation_config ,
        GenerationConfig) or generation_config is None
```

```
self.model_id = model_id
self.device = device if device else "auto"
... # Initialize model and tokenizer
self.tokenizer.chat_template = self.get_jinja_tempalte()
self.generation_config=generation_config if generation_config
else self.default_generation_config()
```

The tokenizer is applied a *jinja-template* in the *constructor*, which configures the *prompt structure* for the model to perform the task. After applying the *template*, *prompts* will be structured like this:

```
"""
<|system|>
Use the following pieces of context to answer the question at the end.
If you can not answer the question from the given context, just say so.
Don't make any things up.
<|context|>
{context}
<|question|>
{question}
<|assistant|>
"""
```

**NOTE:** In a first experiment, the model was provided with explicit instructions on how to respond if it was unable to generate an answer based on the given context. It was observed that the models struggled to replicate this exact phrasing and made grammatical errors. The prompt template was adjusted to allow the model more freedom in its generated answers.

During initialization, a default generation\_configuration is also created. The number of new to-kens to be generated is limited to 150. The parameter no\_repeat\_ngram\_size is used to ensure that no identical strings of three words are repeated. This is particularly important as large language models tend to repeat themselves.

```
def default_generation_config(self):
    gen_cfg = GenerationConfig.from_pretrained(self.model_id)
    gen_cfg.max_new_tokens = 150
    gen_cfg.pad_token_id = self.tokenizer.pad_token_id
    gen_cfg.begin_suppress_tokensrepetition_penalty = 5
    gen_cfg.no_repeat_ngram_size = 3
    return gen_cfg
```

The function responsible for generating answers is \_\_call\_\_ . It takes a question and a list of contexts as input and applies the prompt structure as described above. The received prompt is tokenized using the tokenizer and then fed into the language model to generate an answer. The generated answer is saved in the answer variable, which includes the prompt tokens and up to 150 new tokens (as configured). To decode the generated answer, only the newly generated tokens are passed to the tokenizer by outputs [0] [len(tokenized\_prompt.input\_ids [0]):]

```
def __call__(self, question, context):
```

```
prompt = self.construct_rag_prompt(question, context)
tokenized_prompt = self.tokenize(prompt)
outputs = self.model.generate(
    **tokenized_prompt,
    generation_config=self.generation_config
)
answer = self.tokenizer.decode(
    outputs[0][len(tokenized_prompt.input_ids[0]):]
)
return answer
```

#### 3.1.3 Vector store

The VectorStore class is initialized with a model\_id for the encoder. The retriever models are not located as components in the inference loop but are called via the VectorStore class. In addition to the model\_id, there are also optional parameters for hybrid, weight and distance\_metric. The parameters hybrid and weight relate to the hybrid search. The parameter hybrid is a boolean value that determines whether the VectorStore should enable hybrid search, weight determines how much each search score should be considered if hybrid is set to True. The distance\_metric determines which metric should be used for the search in the vector space; if none is specified, the cosine similarity is used by default. Possible values for distance\_metric are cosine, 12 and ip.

The documents are stored using a pandas.DataFrame with columns id, text and vector.

The function search combines the search functions for the two search paradigms which return the top\_n best scores for each search. The scores of both searches are *normalized* to ensure that they behave the same when combined.

The scores from both searches are combined using the following formula.

```
score_{hybrid} = (1 - weight) \cdot score_{sparse} + weight \cdot score_{dense}
```

The *sparse search* is only executed if the *hybrid search* is activated. The search function returns the result as a list of *dictionaries* sorted in descending order by best score. Each *dictionary* contains the respective *score*, as well as the *document text* and the *id*.

```
def search(self, query, top_n=10):
    vectorized_query = self.encoder(query)
    dense_results = self.dense_search(list(vectorized_query), top_n)
```

```
sparse_results = None
if self.bm25:
    sparse_results = self.sparse_search(query, top_n)
hybrid_results = self.merge_results(dense_results, sparse_results)
result = [
    {
        "score": score,
        "document": self.documents.text.values[key],
        "id": self.documents.id.values[key]
    } for key, score in hybrid_results.items()
]
return sorted(result, key=lambda x: x["score"], reverse=True)[:top_n]
```

It is **important** to note that the calculation for score using  $\ell^2$  has been adjusted. As mentioned in section 2.1.2,  $\ell^2$  measures the distance between vectors, so smaller values are preferable to larger ones. The scores have been *normalized* and then subtracted from 1 so that the  $\ell^2$  metric behaves like the other search metrics, where higher scores are better.

## 3.1.4 Inference loop

The inference loop enables components to interact through nesting. The two outer loops iterate through lists of model ids for decoder and encoder models to be tested. Another loop tests the hybrid and dense search for each model combination, followed by iterating through the entire dataset. The search is performed for each data point in the dataset using each distance metric. The highest-scoring search result (measured by  $ndcg^8$ ) from the various distance metrics is then inputted into the causal language model to generate an answer.

```
for causal_id in causal_models:

# Initialize causal lm
...

for retriever_id in retriever_models:

# Initialize VectorStore with retrievers
...

for hybrid in [True, False]:

# differentiate between hybrid and dense search
...

for row in dataset.iterrows():

# iterate through the dataset
...

for distance_metric in ["cosine", "ip", "12"]:

# differentiate between distance metrics
# retrieve document with distance
# generate answer from best found contexts
```

<sup>&</sup>lt;sup>8</sup>NDCG is described in section 3.2.3.

## 3.2 Results

#### 3.2.1 Dataset

**SQuADv2** Squadv2<sup>9</sup> is a dataset that extends SQuADv1, which comprises 100,000 data points containing an id, a title, a context, a question, and answers. SQuADv2 adds 50,000 examples without answers. SQuADv1 is appropriate for testing a system's ability to answer questions based on a given context. SQuADv2 requires the system to determine whether the question can be answered based on the provided context. It is important to note that the answers provided only consist of bullet points/text snippets. However, the task in this work is to generate full-text answers with a causal language model. This makes it difficult to evaluate the model based on the answers provided.

Machine Learning by Zhi-Hua Zhou (Book) The book Machine Learning by Zhi-Hua Zhou<sup>10</sup> [25] is used as a second dataset. The book was read using the Python package unstructured and split into semantic chunks using and langchain—experimental. The SemanticChunker <sup>11</sup> uses a embedding model to calculate similarity between consecutive sentences.

```
# read file
elements = partition(str(file.resolve()))
    # chunk by title
    text_elements = chunk_by_title(elements)
    chunks = []
    for element in text_elements:
        element.apply(
             partial (
                 clean,
                 bullets=True,
                 extra_whitespace=True,
                 dashes=True,
                 trailing_punctuation=True
        chunks.append(element.text)
text_splitter = SemanticChunker(model)
docs = text_splitter.create_documents([book])
```

The sections for the *table of contents* and *index* were sorted out, resulting in 2513 chunks for the RAG system to work on.

The RAG system was evaluated by answering 28 machine learning questions related to the books context. A list of these questions can be found in appendix A.

<sup>&</sup>lt;sup>9</sup>SQuADv2 on huggingface (visited 04.03.2024)

 $<sup>^{10}\</sup>mathrm{The}$  book can be found online in the FH SWF library. (visited 04.03.2024)

 $<sup>^{11}</sup>$ Langchain documentation and youtube presentation of the idea of semantic chunking. (visited 04.03.2024)

#### **3.2.2** Models

Each combination of the models listed below was tested. In addition, each combination was tested in a *hybrid setup* with a *score weight* of 0.5 for each.

#### Retriever

The selected *embedding models* were taken from MTEB (  $Massive\ Text\ Embedding\ Benchmark)^{12}$ . MTEB [11] is a benchmark for measuring text embeddings models in various tasks. All models used are open source.

### Causal models

The causal models used are all instruction fine-tuned models, identifiable by their suffix (it or instruct). The list of causal models include the open source models from Mistral.AI, the mistralai/Mistral—7B—Instruct—v0.2 and the mistralai/Mixtral—8x7B—Instruct—v0.1. The HuggingFaceH4/zephyr—7b—beta is a variant of mistralai/Mistral—7B—v0.1 that has been further fine-tuned by HuggingFace and therefore has the same architecture. In contrast to the other models, mistralai/Mixtral—8x7B—Instruct—v0.1 is a significantly larger model with 46.7 billion parameters<sup>13</sup>.

In addition to the model variants from *Mistral*, the new open source models from *Google* called *Gemma* were also tested. There are two variants, one model with approx. 2 billion parameters and another with 7 billion. A 7B Zephyr variant of Gemma fine-tuned by *HuggingFace* was also tested.

The causal language models were all loaded in torch. float16 format.

```
"HuggingFaceH4/zephyr-7b-beta"
```

NOTE: Model google/gemma-7b-it was excluded from the evaluation because it only generated the sequence <pad><pad><pad><eos> for all pairs. However, the small variant google/gemma-2b-it was able to generate good answers.

<sup>&</sup>quot;sentence-transformers/all-MiniLM-L6-v2"

<sup>&</sup>quot;BAAI/bge-base-en-v1.5"

<sup>&</sup>quot;WhereIsAI/UAE-Large-V1"

<sup>&</sup>quot;BAAI/bge-m3"

<sup>&</sup>quot;mistralai/Mistral-7B-Instruct-v0.2"

<sup>&</sup>quot;HuggingFaceH4/zephyr-7b-gemma-v0.1"

<sup>&</sup>quot;mistralai/Mixtral-8x7B-Instruct-v0.1"

<sup>&</sup>quot;google/gemma-2b-it"

<sup>&</sup>quot;google/gemma-7b-it"

<sup>&</sup>lt;sup>12</sup>HuggingFace MTEB (visited 04.03.2024)

<sup>&</sup>lt;sup>13</sup>Mixtral of experts (visited 04.03.2024)

#### 3.2.3 Metrics

First of all, it must be said that evaluating an entire RAG system is difficult. This is because evaluating the responses of a causal language model alone is difficult, as the answers generated can vary but may have the same meaning. This means that there is no single correct answer. Another problem with evaluating the causal language model is that it relies on the performance of the retriever, otherwise incorrect contexts will be retrieved and the model will not be able to answer the user's queries.

We use different metrics to measure and evaluate the performance of the language models. For the retriever model, we measure its performance using the normalized discounted cumulative gain (NDCG). As described, since the evaluation of causal language models proves to be difficult, several metrics come into play. For the causal model, BLEU, ROUGE, different versions of Recall and Precision, F1 and Judging LLM-as-a-Judge are used for evaluation.

Normalized Discounted Cumulative Gain (NDCG) The *NDCG* is a metric for measuring rankings in which relevant documents are found and suggested [22]. In contrast to other ranking metrics, the *NDCG* enables us to give documents some kind of a continuous *relevance score* based on query, whereas other metrics only allow binary values for relevance.

The NDCG puts the resulting ranking in relation to the ideal ranking.

$$NDCG = \frac{DCG}{IDCG} \tag{11}$$

with

$$DCG = \sum_{i=1}^{k} \frac{rel_i}{\log_2(i+1)}$$

where:

i = ranking index,

rel = relevance score.

and *IDCG* is same as *DCG* (resulting ranking order) but with optimal ranking order.

**BLEU** (Bilingual Evaluation Understudy) BLEU is a metric that is frequently used in machine translation [13]. Similar to generating text, as a RAG system does, machine translation also generates free text, but in a more dependent context.

BLEU compares the quality of a translation by searching for identical n-grams to evaluate the correspondence between the  $machine\ translation$  and a  $human\ translation$ .

$$BLEU = BP \cdot \exp(\sum_{n=1}^{N} w_n \cdot \log p_n)$$
 where:
$$BP = \text{brevity of the translation compared to the reference,}$$

$$p = \text{precision value for n-grams is,}$$

$$N = \text{the maximum n-gram size.}$$
(12)

ROUGE (Recall-Oriented Understudy for Gisting Evaluation) The ROUGE metric is mainly used to evaluate machine-generated summaries [8]. In contrast to BLEU, which is often used for the evaluation of translations, ROUGE focuses on assessing the quality of text summaries.

It takes into account aspects such as the overlap of words, sentences and phrases between the summary and the references. *ROUGE* has different variants such as *ROUGE-N* (for word match) and *ROUGE-L* (for longest common subsequence).

**Precision** Precision measures the ratio of correctly predicted positive instances to the total number of predicted positive instances. It is a measure of the model's accuracy in predicting positive instances.

$$Precision = \frac{\text{True positives}}{\text{True positives} + \text{False positives}}$$
 (13)

**Recall** Recall measures the ratio of correctly predicted positive instances to the total number of actual positive instances. It indicates how many of the actual positive instances the model has correctly identified.

$$Recall = \frac{\text{True positives}}{\text{True positives} + \text{False negatives}}$$
 (14)

**F1** The *F1 score* is a metric that balances *precision* and *recall*. The *F1 score* is calculated as a harmonic mean between precision and recall and takes into account both false positive and false negative predictions.

$$F1 = 2 \cdot \frac{\text{precision} \cdot \text{recall}}{\text{precision} + \text{recall}}$$
 (15)

**Judging LLM-as-a-Judge** [24] describes a modern approach that uses a *large language model* to evaluate generated answers based on context, similar to human evaluation. It is crucial to use a language model with greater capacity for evaluation, which is why *GPT-3.5-turbo*<sup>14</sup> was used in this work. Only *answers* generated in combination with the best *retriever model* were evaluated.

 $<sup>^{14}\</sup>mathrm{GPT\text{-}}3.5\text{-turbo}$  documentation (visited 04.03.2024)

## 3.2.4 Result - SQuADv2

#### Retriever

Figure 7 shows the NDCG scores for the SQuADv2 dataset. Each combination was tested on 500 examples. The examples were divided into 400 examples with questions that could be answered using a given context and 100 examples whose questions could not be answered using the context provided.

It is noteworthy that the *sentence-transformers/all-MiniLM-L6-v2*, the smallest of the models used, achieved the best scores. The *hybrid search* consistently improved the results for all combinations. It is peculiar to observe that the different distance metrics per model produced identical results. All models perform best when *cosine similarity* is used as the distance metric.

model	cosine	cosine hybrid	ip	ip hybrid	$\ell^2$	$\ell^2$ hybrid
sentence-transformers/all-MiniLM-L6-v2	0.823	0.874	0.823	0.874	0.823	0.874
WherelsAl/UAE-Large-V1	0.857	0.870	0.857	0.870	0.857	0.870
BAAl/bge-m3	0.796	0.859	0.796	0.859	0.796	0.859
BAAI/bge-base-en-v1.5	0.856	0.870	0.856	0.870	0.856	0.870

Figure 7: NDCG results rounded to three decimal places.

#### Decoder

In fig. 8, the results for the metrics ROUGE, Precision, Recall and F1 of the causal language models are presented. As previously mentioned, evaluating free text can be challenging, especially since the answers provided by SQuADv2 are only individual words and not complete sentences. Therefore, the results do not provide much insight into the correctness and quality of the answers. It is evident that, for the most part, the small variant of the gemma models performed the best.

**NOTE:** It is worth noting that BLEU is not included in this evaluation as all models scored towards 0.

model	r1 precision	r1 recall	r1 f1	rL precision	rL recall	rL f1
mistralai/Mistral-7B-Instruct-v0.2	0.068	0.510	0.107	0.0631	0.477	0.092
mistralai/Mixtral-8x7B-Instruct-v0.1	0.0765	0.522	0.121	0.070	0.484	0.111
HuggingFaceH4/zephyr-7b-beta	0.041	0.559	0.071	0.037	0.527	0.065
google/gemma-2b-it	0.090	0.394	0.133	0.080	0.364	0.119
HuggingFaceH4/zephyr-7b-gemma-v0.1	0.028	0.475	0.050	0.026	0.454	0.046

Figure 8: Here, r1 and rL represent ROUGE-1 and ROUGE-L, respectively. (Results were rounded to three decimal places.)

Figure 9 shows the results for Judging LLM-as-a-Judge. This metric provides a better indication of the quality and accuracy of the answers generated. Unlike in fig. 8, HuggingFaceH4-zephyr-7b-beta performs the best in this scenario. It is worth noting that the largest model, mistralai-Mixtral-8x7B-Instruct-v0.1, only ranks third and is not significantly better than the smallest model google-gemma-2b-it.

model	correct	wrong	%
mistralai-Mistral-7B-Instruct-v0.2	400	100	80.0
mistralai-Mixtral-8x7B-Instruct-v0.1	396	104	79.2
HuggingFaceH4-zephyr-7b-beta	427	73	85.4
google-gemma-2b-it	387	113	77.4
HuggingFaceH4-zephyr-7b-gemma-v0.1	337	163	67.4

Figure 9: Results from Judging LLM-as-a-Judge on SQuADv2.

### 3.2.5 Result - Machine Learning Book

The complete raw pairs from context, question and answer can be found in LLM—as—a—judge—ML—BOOK.xlsx . A combination of the best retriever model (sentence-transformers/all-MiniLM-L6-v) and all causal language models, except for google/gemma—7b—it and mistralai/Mixtral—8x7B—Instruct—v0.1 (due to its size) were used to process the dataset. Overall, it can be said that all causal language models were able to answer most of the questions satisfactorily. Like section 3.2.4, the Mistral.Ai models architecture perform best.

model	correct	wrong	%
mistralai/Mistral-7B-Instruct-v0.2	25	2	92.59
HuggingFaceH4/zephyr-7b-beta	25	2	92.59
google/gemma-2b-it	21	6	77.77
HuggingFaceH4/zephyr-7b-gemma-v0.1	23	4	85.18

Figure 10: Results from Judging LLM-as-a-Judge on Machine Learning Book dataset.

No evaluations with metrics *ROUGE*, *BLEU*, *Precision*, *Recall* and *F1* are available for this experiment as the dataset does not contain any reference answers.

#### 3.2.6 Conclusion

RAG appears to be a promising alternative to address the limitations of large causal language models. However, as is often the case in machine learning, it is important to ensure the cleanliness of the data, as demonstrated in section 3.2.4.

The retrievers have performed exceptionally well, particularly in the hybrid variant. It would be interesting to compare their performance with that of commercial embeddings. To improve results, it may be possible to fine-tune the retriever models on the given data.

Training the *generative model* can also lead to better results, but it should be noted that the whole system is dependent on the good results of the *retriever*.

Another component that was not considered here is a neural reranker that uses cross attention to calculate a similarity between the question and the found contexts and reranks the results.

One problem with the evaluation of *generative language models* is the difficulty in assessing the correctness and semantic similarity of the generated text. Using another *large causal language model* as a *judge* seems to work well, but caution is advised, as *large language models* can generate unpredictable results and are therefore not always dependable as an evaluation system.

Another approach that follows this concept is RAGAS (Automated Evaluation of Retrieval Augmented Generation) [2]. [2] also uses a large language model to evaluate results. The evaluation provides several values that determine the quality of the generated answers.

# A Appendix List of questions

- 1. What is overfitting?
- 2. What is underfitting?
- 3. How do we test a models generalization error?
- 4. Why should training data points not be in the test set?
- 5. What is cross validation?
- 6. What are commonly used metrics to measure the performance of a model?
- 7. What is linear regression?
- 8. What are problems of linear models?
- 9. What is a decision tree?
- 10. What is the McCulloch–Pitts model?
- 11. What is a neural network?
- 12. How is a neural network optimized?
- 13. How does backpropagation work?
- 14. What is deep learning?
- 15. What is ensemble learning?
- 16. What is the goal of ensemble methods?
- 17. What is supervised learning?
- 18. What is unsupervised learning?
- 19. What is the difference between supervised and unsupervised learning?
- 20. What is the goal of clustering?
- 21. How is k-Nearest Neighbor trained?
- 22. Which algorithm can be used to reduce dimensions?
- 23. What is semi-supervised learning?
- 24. What is reinforcement learning?
- 25. What is an Markov Decision Process used for in reinforcement learning
- 26. How interacts an reinforcment learning agent with its environment?
- 27. What is the goal of an agent in reinforcement learning?
- 28. What is the Exploration-Exploitation dilemma?

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