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Multilingual Grapheme-to-phoneme Conversion

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Abstract

In this master's thesis I tackle the problem of multilingual grapheme-to-phoneme conversion. In a first part, I familiarized myself with the current data situation. Most G2P data is available as word lists that contain mappings from graphemes to phonemes. I created models for 22 different languages using an existing transformer model. As an additional challenge I tried to incorporate phonetic features by extending the input phonemes. The models I trained could keep up with state-of-the-art G2P models and outperformed existing models in different languages. Phonetic transcriptions are not as standardized as writing systems used for written language. As I tested the models on different datasets, this became clear. As for many of my languages there is no result available, my models can serve as a baseline for future research. All scripts and files produced along with this thesis are found in my GitHub repository: <https://github.com/theDebbister/masterThesis>.

Zusammenfassung

In dieser Masterarbeit beschäftige ich mich mit dem Problem der mehrsprachigen Graphem-Phonem-Konvertierung. In einem ersten Teil habe ich mich mit der aktuellen Datenlage vertraut gemacht. Die meisten G2P-Daten sind als Wortlisten verfügbar, die Zuordnungen von Graphemen zu Phonemen enthalten. Ich erstellte Modelle für 22 verschiedene Sprachen, indem ich ein bestehendes Transformer-Modell verwendete. Als zusätzliche Herausforderung habe ich versucht, phonetische Features einzubeziehen, indem ich die Eingabephoneme erweitert habe. Die von mir trainierten Modelle konnten mit den modernsten G2P-Modellen mithalten und übertrafen die bestehenden Modelle in verschiedenen Sprachen. Phonetische Transkriptionen sind nicht so standardisiert wie Schriftsysteme, die für geschriebene Sprache verwendet werden. Dies wurde deutlich, als ich die Modelle an verschiedenen Datensätzen testete. Da für viele meiner Sprachen noch keine Ergebnisse vorliegen, können meine Modelle als Grundlage für künftige Forschungen dienen. Alle Scripts und Dokumente, die ich im Zusammenhang mit dieser Arbeit erarbeitete, sind auf meinem GitHub zu finden: <https://github.com/theDebbister/masterThesis>.

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List of Acronyms

IPA	International Phonetic Association
IPA	International Phonetic Alphabet
WER	word error rate
CER	character error rate
PER	phone error rate
G2P	grapheme-to-phoneme
FST	finite-state transducer
EM	Expectation-Maximization
seq2seq	sequence-to-sequence
JIPA	Journal of the International Phonetic Association
SIGMORPHON	Special Interest Group on Computational Morphology and Phonology
NLP	natural language processing
OCR	Optical Character Recognition
PoS	Part-of-Speech
UZH	University of Zurich
WALS	World Atlas of Language Structures
ASR	automatic speech recognition
TTS	text-to-speech
HTR	Handwritten Text Recognition
NWS	<i>The North Wind and the Sun</i>
RNN	reccurent neural network

1 Introduction

With the advent of technologies that can process huge amounts of data, many linguistic tasks that were originally very tiresome and expensive to do, can now be accomplished much faster. Well known examples for this branch called natural language processing (NLP) are machine translation or search engines. A lot of available tools and consequently research done in this area is concerned with written text. For many scenarios like machine translation, large corpora of written text in many languages are collected that are used to train computational models that solve those tasks. Following, there is an ever-growing set of models that are trained on written text. Often the goal is to reach or outperform human solutions to those various tasks. The corpora that are gathered for tasks like machine translation serve as an example for how the outcome of the task in question should look like. For example, a corpus for machine translation contains a text in one language with its corresponding translation into an other language. It is therefore necessary that this training data represents the language and the task well enough for a machine to reach the required performance. From a linguistic point of view, the question comes up if focusing on written language only can ever represent human language adequately. Most of communication and daily language use happens through speaking. This is a first potential limitation of many (written) language technologies. It is not easy to find out what characteristics of a language can be observed in written representations. Although many NLP tasks use written language corpora exclusively, there are technologies like automatic speech recognition (ASR) or text-to-speech (TTS) that use corpora of spoken language as training data. Spoken language in those cases is mostly represented as phonetic transcriptions. As phonetic transcriptions are strings of characters they are much easier to process for a machine than raw audio files. In order to represent spoken language as phonetic transcriptions, we need to know a lot about how spoken and written language relate. Research on ASR or TTS contributes to a better understanding of the relation between spoken and written language as both of the tasks involve mapping of spoken to written language. But still, this does not answer the question of how well written language represents language in general. The representative power of written text is much less studied. This is where this current thesis connects to cutting-edge research.

I am going to present my attempt of studying a multilingual phonetic corpus and comparing it to its written-text version.

1.1 Research questions & goals

I will try to answer the following question as the result of my thesis: **Is it essential for the study of multilingual corpora to perform analyses on phonetic text (i.e. speech representations) rather than only written text?** This question addresses many different topics from the field of linguistics but also more technical aspects as I have mentioned in the paragraph above. None of these topics is trivial and the question cannot be answered easily. While in this thesis I cannot possibly discuss everything from the use of phonetic transcriptions up to the nature of human language use, the aim is to make a step into the direction of quantifying the representative power of written text. The goal is now to collect a corpus of phonetic transcriptions in various languages based on an already existing written corpus. Once I have the phonetic corpus, it is then possible to perform analyses on the phonetic corpus which can then be compared to the same analyses on the written corpus. In order to add a phonetic corpus to an already existing one, various steps need to be performed which are outlined below:

1. **Data collection:** The given dataset contains no phonetic transcriptions of those 100 languages. The first step is to find already existing data.
2. **Computational model to create phonetic transcriptions:** As existing data will not be available in sufficient amounts to perform meaningful analysis, the next step is to actually create phonetic transcriptions of as many languages as possible of the corpus. This will require to create and train a computational model.
3. **Test and Analysis:** Once the transcriptions have been obtained, the newly created phonetic corpus can be analysed and tests can be performed on the phonetic data.

The first two steps will already take up quite a lot of time which is hard to estimate in advance. It is not clear how much of the third step can be accomplished. I still think it is important to keep the bigger picture in mind and be aware of the overall importance of the topic. By performing the steps above I will have to tackle different challenges. Some of them are already known now, but there will be many more that will only become apparent while working on the tasks.

1. What types of phonetic data is available and how can it be used?
2. Which computational models can be used to create phonetic transcriptions automatically?
3. How can we use phonetic features to create phonetic transcriptions?

All of these challenges above are interesting and important, but the third challenge is of particular importance as it is a new approach to creating phonetic transcriptions.

1.2 Thesis structure

The thesis is subdivided into six chapters of which the first chapter is this introduction. Chapter 2 covers the linguistic basics that are necessary to understand the topic. It presents the linguistic foundation of phonetics and phonology and an introduction to corpus linguistics or rather corpus phonetics. Chapter 3 sets the boundaries of the technical background. I will present an overview of the possibilities for automated creation of phonetic transcriptions. In chapter 4, I document my first practical experiments which are concerned with data collection. It contains a descriptions of the data I will later use to create models for phonetic transcriptions. In chapter 5 I will present one of my experiments where I compare the word length of phonetic and written text. Chapter 6 presents my own experiments to create phonetic transcriptions of the corpus. I will present all models that I created and their results. Chapter 7 summarizes my findings and presents ideas for future research.

2 Representations of written and spoken language

From a linguistic point of view there are two high-level concepts that are important for this thesis. The first and most significant one is that of the **relation of written and spoken language**. Both spoken and written language are a way of representing a language in general. As I have already mentioned, computational linguistics deals mostly with written rather than spoken language, but what does traditional linguistics say about the relation of written and spoken language? Both written and spoken language are *valid* representations of the same language. As we will see later in this chapter, mapping a spoken language to its written representation is far from easy and never perfect. Elaborated writing systems like we are used to today came only much later compared to language in general [CrashCourse, 2021b; Hock and Joseph, 2019]. The question that gave rise to this thesis is if writing systems can capture language as such well enough in order to analyze a language properly. Or if we need spoken representations as well.

Whenever we study language we look at samples of that language. It is simply impossible to study an *entire* language as we would need all texts that were ever produced in that language. Consequently, we need to ask ourselves how much and what material of a language is enough to study it properly. In order to do that we can make use of techniques and methods from corpus linguistics. In linguistics and also in computational linguistics we typically talk about a corpus when we talk about a huge dataset that is used to represent a language. It does not matter if the corpus contains written or spoken samples of a language although it is mostly written [McEnery and Hardie, 2011]. Corpus linguistics allows for both qualitative and quantitative analysis of text. Due to recent technological advancement it has become possible to store large digital collections of speech recordings and their aligned transcriptions. These possibilities gave rise to a wider acknowledgement of corpus phonetics. Corpus phonetics deals with an abundance of linguistic variation. In addition to language, style or vocabulary variation, there are differences in dialect and idiolect, physiological state of the speakers and their attitude [Lieberman,

2019; Chodroff, 2019]. In section 4.3.2 I will present an interesting study on corpus phonetics [Baird et al., 2021]. The authors of the study aim to answer how much phonetic data is needed to cover all sounds of a language.

The second key concept is that of **multilingual analysis**. The corpus underlying this thesis is multilingual. I will introduce this corpus later in section 4.2. While a multilingual corpus in itself is not uncommon, the specific goal of the team maintaining the corpus is to compare the languages and study their similarities. Comparing languages and studying their similarities and differences is part of a well-established branch of traditional linguistics called comparative linguistics. For example, Gutierrez-Vasques et al. [2021] performed a study on a multilingual corpus comparing different subword tokenizations. They compared 47 languages to find out how languages can be tokenized such that it leads to similar distributions among that languages. This analysis was performed on written language and is an example for what could be done on a phonetic corpus as well.

While there is a lot that could be said about any of these topics, I am not going into more detail about these. The point I would like to make is that the task of creating phonetic transcriptions of a multilingual corpus with the long-term objective of comparing written and spoken language is deeply rooted in linguistics and not at all trivial. In the following sections I will mostly write about phonetics and phonology in general and writing systems.

2.1 Phonetics and phonology

Given that phonetics and phonology is a sub-area of traditional linguistics and often only touched on superficially in computational linguistics, I will summarise the most important assumptions and terms that are necessary to understand what phonetic transcriptions are. A very important terminological distinction is between **phonetics** and **phonology**. While phonetics refers to the study of actual sounds, phonology refers to the study of sound *systems*. In phonetics, it is not so much important what the different sounds mean, but how they are produced and perceived and what different sounds a human being can produce and perceive at all. When it comes to human communication using spoken language, many of sounds that can be produced are not actually meaningful. A human being is perfectly capable of producing many different sounds but some just do not mean anything to other people. The study of sound systems, phonology, is used to describe the set of meaningful sounds that make up a language. The sounds within a sound system of a language can be used to construct meaningful utterances. For this thesis, phonology is more important.

Example 2.1 shows how the word ‘request’ can be pronounced in English. The pronunciation is written in the International Phonetic Alphabet (IPA) which I will explain in more depth in section 2.3. For now it is important to note that the same word can be pronounced differently.

(2.1) Different pronunciations of the English word ‘request’. The spaces indicate how the letters can be mapped to the phonetic symbols on the right side:

(a) r e q u e s t [ɹ ɪ k w ɛ s t]

(b) r e q u e s t [ɹ̥ ɪ k w ɛ s t]

(c) r e q u e s t [r ɪ k w ɛ s t]

Example 2.1 a) shows how the word ‘request’ is pronounced in British English. Even someone that does not know the IPA can see that the letter ‘r’ can be mapped to the pronunciation symbol [ɹ]. This pronunciation of the Latin letter ‘r’ is a part of the British English phoneme inventory. Another English speaker might pronounce the letter ‘r’ a bit different as in example 2.1 b). We can see that the pronunciation symbol, [ɹ̥], for the /r/ differs from the one in the first example. There is a little circle added beneath it which denotes a slightly different pronunciation. However, an English speaker would still understand the word ‘request’ as the different pronunciation does not change the meaning of the word. In phonetic terms we would say that in the British English sound system there exists an abstract sound /r/ which is called the **phoneme** /r/. Note that we write a phoneme between slashes. Also, in our case there exist two **phones** [ɹ] and [ɹ̥] which are specific sounds that can be used to pronounce the phoneme /r/. No matter which phone I use, it does not change the meaning of the word. Note that we write a phone between square brackets. In British English, [ɹ̥] is an **allophone** of the phoneme /r/. Allophones are specific phones that are concrete realizations of the same phoneme. Realization in this case just means that it is a sound in an actual utterance. They can be used interchangeably without changing the meaning of the utterance. Very importantly, allophones are language specific. In other languages I cannot use the same phones to pronounce a phoneme. In example 2.1 c), the ‘r’ is mapped to the phone [r] which is part of the Spanish sound system. Although a British English speaker would still understand the word, [r] is not an allophone of /r/ in British English as typically British English speakers do not say the ‘r’ like in Spanish. The phone [r] does not exist in the British English sound system. Each language has a **phoneme inventory** of abstract sounds that exist in the language’s sound system. Also, there is a set of phones which are concrete realizations of abstract phonemes. Further, for each phoneme in a language, there exists a set of allophones all of which can be used

to replace another allophone in the same set without changing the meaning of an utterance.

When it comes to phonetic transcriptions, the difference between phoneme and phone is very important. When we speak about a phonemic transcription, also called **broad** transcription, we do not care about the exact sounds. If a person uses an allophone to pronounce the word, we still write down the respective phoneme. If we want a phonetic transcription, also called **narrow** transcription, we care about every detail of the pronunciation. If a person uses an allophone, we note down that allophone and not the respective phoneme.

It is important to note at this point that the terms phonetic and phonemic, and phone and phoneme are sometimes used interchangeably. Their linguistic definition as given above is relatively clear while the definition on the computational side is often less strict. Strictly speaking, phonemic transcriptions are not allowed to contain allophones but should write the respective phoneme of that language. This will not always be the case when it comes to data used in language technology [Lee et al., 2020]. Example 2.2 shows a German broad transcription of the word ‘Anrede’. Example 2.3 shows a broad transcription of the German word ‘Anredefall’ which should be pronounced the same except for the last additional part. However, as we can see in the example, the first part looks different. In the German sound system, the [r] is actually an allophone of [ʀ]. Correctly speaking, the second example is wrong as in broad transcriptions, we are not allowed to use allophones. The correct broad transcription of ‘Anredefall’ would be as shown in example 2.4.

(2.2) Anrede: [a n ʀ eː d ə]

(2.3) Anredefall: [a n r eː d ə f a l]

(2.4) Anredefall: [a n ʀ eː d ə f a l]

The above example is from the actual datasets I will be using later (see section 4.3).

In the following, I will outline different terms and concepts that appear a lot in research and literature concerned with phonetics and phonetic transcriptions.

Vowels and consonants In order to categorize the sounds in the sound systems of different languages, sounds are typically split into vowels and consonants. The terms to describe both vowels and consonants are inspired by the physical position of the tongue and the mouth when the sound is produced. To describe vowels, we use the position of the tongue in the mouth and if the lips are rounded or not. Using those two categories enables us to distinguish every possible vowel. The position of

the tongue in the mouth is described along two axes: The tongue can be moved from the back of the mouth cavity to the front. This is the back-front axis. The second axis is that the tongue can be moved up and down in the mouth. In phonetics, up and down is referred to as close and open. A vowel can then, for example, be described as close-back rounded. This means that in order to produce the sound, a person needs to move the tongue up and back in the mouth and round the lips. Figure 2 shows the vowel chart how it is usually represented in the IPA. More on this special alphabet follows in section 2.3.

Consonants are defined by the place and the manner of their physical production. The place, again, refers to the position of the tongue in the mouth and the overall form of the vocal tract. While for vowels we used the open-close and back-front schema to describe the position of the tongue, the consonant tongue positions are categorized differently. Table 1 shows all consonant place categories.

bilabial	labiodental	dental	alveolar
postalveolar	retroflex	palatal	velar
uvular	pharyngeal	glottal	

Table 1: This table displays all tongue positions and shapes of the vocal tract that are used to produce consonants.

Unlike for vowels, for consonants the entire vocal tract, the lips and the tongue are used to block the air and make it flow in a specific way. For example, a dental consonant means that the tip of the tongue is pressed against the upper front teeth. For palatal consonants, the body of the tongue is pressed against the hard palate in the back of the mouth cavity. The manner of consonant production describes very precisely how the air is lead through the mouth to produce a sound [CrashCourse, 2021a]. Examples for the manner of a consonant are plosive or trill. For a trill the tongue needs to move in a vibrating way which consequently makes the air vibrate. Note that the place is not changed by this vibration. The tip of the tongue can still be pressed against the teeth or be in the back of the mouth while vibrating. For a plosive sound we first completely block the air such that no air can leave the mouth and then the air is pushed out of the mouth in a fast manner, a bit like an explosion, therefore the name. The complete consonant chart is depicted in figure 2 as well and table 2 shows all consonant manner categories.

Once I have introduced the IPA in section 2.3, I will give examples of how the IPA helps us to write sounds with respect to the above introduced categories of vowels and consonants and how it differs from orthographic representations.

plosive	nasal	trill	tap or flap
fricative	lateral fricative	approximant	lateral approximant

Table 2: This table displays manner categories of consonants.

Syllables Letters can be grouped into larger units called *syllables*. Syllables can be an entire word or a part of a word. A syllable can be subdivided into different parts called *onset*, *nucleus* and *coda*. For every syllable in every language it is true that the nucleus cannot be empty. The onset and the coda can be empty. Example 2.5 shows English words with syllable boundaries.

(2.5) English syllables: the hyphen denotes a syllable boundary.

(a) the-sis

(b) o-ver

In English, the onset is typically a sequence of consonants. The nucleus is a sequence of vowels or just one vowel. The coda is another sequence of consonants. The second syllable in example 2.5 a) follows this structure exactly: ‘s’ is a consonant followed by a vowel ‘i’ followed by another consonant ‘s’. The first syllable in example 2.5 b) on the other hand, consists only of the nucleus, in this case the vowel ‘o’. Other than this three-part structure, syllables are organized very differently in different languages [Kracht, 2007]. For computational analysis it is important that syllables are a way of segmenting written or spoken language. They are larger than individual letters or phonemes but often still smaller than individual morphemes or words.

Diphthongs A diphthong is a sequence of vowels that is considered as just one phoneme if it is within one syllable. If a syllable ends with a vowel and the next one starts with a vowel, this vowel sequence is not called a diphthong. An example is the German word ‘Chaos’. The two vowels in the middle are not a diphthong as there is a syllable boundary right in their middle: ‘Cha-os’. A word like ‘aus’ contains a diphthong as it exists of only one syllable [Kracht, 2007].

Suprasegmentals Apart from individual sounds, there are features of spoken language like stress or intonation. Those are referred to as suprasegmentals. They are often related to syllables. For example, we can put stress on a different syllable or raise the pitch. Semantically, some suprasegmentals in some languages distinguish

meanings, some do not. A special case are tones. Tones are a special way of intonation. In some languages like Chinese or many African languages tones are used to distinguish meaning while in most European languages, the concept of tones does not exist [Kracht, 2007].

2.2 Mappings of written and spoken language

Unlike spoken language that was a part of human interaction all the time, writing systems only developed over time. There are different writing systems that developed in different places at different times. The structure of the spoken language, the cultural context or the tools that were at hand to write are a few of many factors that influenced the emergence of a specific writing system. In general, we can think of writing systems as mappings from sounds to written symbols. The systems used to represent sounds in different languages do not uniquely map a symbol to one specific phoneme. Most of the time, there is a standard pronunciation of each symbol in every language. However, as we have seen above, there is the notion of allophones which means that every sound can be replaced by some other sounds which are also understood in this language. These explanations make clear that the mapping of written symbols to spoken sounds in various languages is complex.

When taking a step back, we can see that a single symbol can represent either a single phoneme, a syllable or a word or even something in between. Different writing systems developed in parallel which means that today, we have an abundance of different strategies to put sounds into writing. Not all writing systems, which are typically called scripts, can be treated the same and this most certainly has implications on models to create phonetic transcriptions. Each major script that is used today will be presented below [CrashCourse, 2021b]. In the following I will use the terms letter or sign when referring to an individual symbol that is part of a writing system. In section 3.5, I will introduce a more fine-grained terminology when referring to symbols in writing systems and their computational representation.

ALPHABET When a letter maps roughly to one phoneme, we call the writing system an alphabet. In German, for example, the writing system is the Latin alphabet. The Latin alphabet is used for many different languages in western Europe and those languages that were influenced by colonisation. There are other alphabets like the Cyrillic or the Greek alphabet. If a language uses an alphabet this does not mean that each letter maps to exactly one phoneme. In fact, one letter can have many different realizations as example 2.6 shows.

(2.6) The examples show the different realizations of the English letter sequence ‘ough’ [CrashCourse, 2021a]. The first part is the letter sequence and the second part the phonetic transcription. The parts marked in red can be mapped onto each other.

- (a) **tough** [tʌf]
- (b) **cough** [kʌf]
- (c) **though** [ðəʊ]
- (d) **through** [θru:]
- (e) **bough** [baʊ]
- (f) **brought** [brɔ:t]

The above examples show that it is not always clear how to pronounce a certain letter sequence. There is no simple one-to-one mapping from one letter or a sequence of letters to one phone or a sequence of phones within the English language. Let alone within all languages that use the Latin alphabet. In addition, alphabets typically have diacritic marks that can be used to extend the main letters. As it is the case with the German ‘ä’ which is a slightly changed version of the letter ‘a’ and pronounced differently. Just as with single letters, also diacritic marks cannot simply be mapped to a phone.

ABJAD A special variant of an alphabet is abjad. Abjad represents only consonants and no vowels. This means that vowels need to be added while reading. Again, this means that there is a lot of ambiguity as it is not always clear which vowel should be added if there is no context. Semitic languages like Hebrew or Arabic make use of abjad.

(2.7) The example shows Hebrew words that are first mapped to the Latin alphabet, then to the Latin alphabet including vowels and in the end to the English translation.

- (a) **בצלם** bzlm bzelem name of an association
- (b) **בצלם** bzlm bzalam ‘their onion’

Example 2.7 shows that in both cases each letter maps to the same consonant but it can be completed with different vowels. The words presented above do not have the same meaning depending on the vowels added although their letter sequence looks exactly the same.

SYLLABARY In syllabaries, a letter, or rather a sign in this case, represents a syllable instead of a single sound. An example is the Korean script. Syllabaries typically do not have any internal ambiguities in their pronunciation as one sign maps to exactly one phoneme.

LOGOGRAPHIC SYSTEMS Logographic systems represent entire words or morphemes as signs. Chinese is an example for a logographic system. We cannot break down Chinese signs into single morphemes or letters. One sign, which is called a logogram, is often pronounced in the same way regardless of the context of the sign. This means that the pronunciation of one sign is less ambiguous than, for example, the pronunciation of one letter in an alphabet which can be mapped to different phonemes in different contexts. Logographic systems often have thousands of signs while alphabets typically have less than one hundred letters. For each logogram, one must learn how to pronounce it.

As the examples above show, it is very difficult to have a clear mapping from sounds to written symbols. For most languages it is not possible to derive the exact pronunciation from the written symbols. Many of the pronunciation rules of a language are based on convention. Speakers of a language just *know* how to pronounce a word. Still, there can arise heated debates about the correct pronunciation of certain words within a language. An example are Swiss German dialects. In Switzerland, dialects are considered very important by many people. Often, every town has their own dialect which people are proud of. This leads to an abundance of different pronunciations for one word. People are perfectly capable of understanding other dialects but just use their own pronunciation when it comes to speaking. As the personal way of speaking is often considered ‘normal’, there are a lot of playful but also more serious discussions about correct pronunciations.

Apart from different conventions, spoken and written languages change differently over time. Spoken languages are typically more flexible and ready to change while their written representation often stays the same [Moran and Cysouw, 2018]. This can lead to official governmental interventions like the German orthography reform of 1996 that intended to adapt the German spelling to represent the German pronunciation more adequately. Also, major inventions like printing machines gave rise to standardization of writing systems as reading and writing became more common [CrashCourse, 2021b].

2.3 The International Phonetic Alphabet (IPA)

The IPA is a special alphabet where each letter is intended to represent exactly one phoneme [CrashCourse, 2021b; Kracht, 2007]. As usual, reality is more complex than what we wish it to be. Even with the IPA there are inconsistencies. Figure 2 shows the full IPA chart including all characters that the International Phonetic Association (IPA) decided to use. Although the IPA seems very complete there are still sounds that cannot be represented using the IPA. This becomes clear when, for example, looking at the vowel chart (see figure 2). The tongue does not ‘click into place’ for the vowels on the chart. Vowel characterisation happens on a continuum. This means that it is always possible to characterize a vowel as in between two vowels on the chart.

When sounds are mapped to a written version we refer to this process as creating a transcription. When creating such transcriptions using the IPA there are different levels of detail. Not all transcriptions represent the sounds in equal detail. Generally, there is the distinction of broad and narrow transcription. These two go back to the linguistic distinction of phone and phoneme as I have explained in section 2.1. Broad transcriptions are less complex and usually easier to create and understand as they contain fewer details. Narrow transcriptions present every speaker individual or dialectal sounds as exactly as possible and are consequently more difficult to create. Narrow and broad transcriptions can diverge greatly. It is important to treat broad and narrow transcriptions as two different kinds of transcriptions.

(2.8) pɪ'kʊ kə'zɜf

(2.9) pɪ'k^hʊ k^hə'zɜf

Example 2.9 is a narrow (phonetic) transcription of the beginning of the Mapudungun version of the short story *The North Wind and the Sun*. The same text is transcribed broadly (phonemic) in example 2.8. As becomes clear in this example, the narrow transcriptions is longer as it contains more different characters. In this case it is only the superscript *h* that is different. The problem, with especially the narrow transcriptions, is that the transcriber still needs to define what narrow means in a specific case. One could argue that there are as many narrow transcriptions of a language as there are speakers of that language. This becomes tricky when given a task to automatically transcribe text. The training data might employ one definition of narrow, while there are texts in the test set that might follow another definition. This is mostly important when talking about data preprocessing and cleaning.

2.4 Phonetic features

As every language has a sound system and a set of phonemes that are part of that system, it is absolutely necessary that we can describe a phoneme very precisely. In computational linguistic analysis you would typically refer to a description of a phoneme as a set of features of that phoneme. For example, I could say that I use one feature to describe every phoneme. This one feature is if the phoneme is a consonant or not. If it is not a consonant, it follows that it is a vowel.

(2.10) The examples show how we can describe different phonemes using features.

‘1’ stands for: ‘is-a-consonant’. ‘0’ stands for: ‘is-not-a-consonant’ (which means that it is a vowel).

$$(a) \quad /ə/ \rightarrow 0$$

$$(b) \quad /r/ \rightarrow 1$$

$$(c) \quad /a/ \rightarrow 0$$

With only one feature I cannot really distinguish different phonemes as becomes clear when looking at example 2.10. In this case, I can in fact only distinguish vowels and consonants. Therefore we need to add more features to describe phonemes better. In linguistics it is very common to use the vowel and consonant schemes described above to describe each phoneme. But it is also possible to use a feature like ‘syllabic’ that tells us whether a sound can be used as a syllable on its own or not. Or we could use a feature like suprasegmental that tells use if a phoneme is a suprasegemntal. In section 4.4 I will introduce a specific set of features that allows to describe many phonemes uniquely.

3 Technical Background

This chapter presents the technical background that is needed for this thesis. It explores everything around automated models that can be used to create phonetic transcriptions. I will first set the basis and dive into evaluation metrics and general architectures and frameworks that are commonly used and then present current state-of-the-art models. Table 4 shows the state-of-the-art models for grapheme-to-phoneme (G2P) modelling. Phonetic transcriptions are often used in ASR and **stt!** (**stt!**) technologies. There are systems that can produce speech directly from orthography and question the necessity of phonetic transcriptions. When only little data is available, the training data might not be enough to train an orthography-to-speech system, making the intermediate step of creating phonetic transcriptions necessary. Another reason for creating phonetic transcriptions is that its usage is not limited to speech applications [Mortensen et al., 2018]. They might also be used to compare languages on speech basis. In order to do that, there needs to be a lot of knowledge about how language works.

3.1 Automated phonetic transcription

Creating phonetic transcriptions is essentially a sequence-to-sequence (seq2seq) task. Like other NLP tasks its goal is to transform a sequence of characters into another sequence of characters. In particular, it is very similar to machine translation [Rao et al., 2015]. In the present case, the input sequence is a sequence of letters or signs. These can look very differently depending on the script (see section 2.2). The output sequence is a sequence of phonemes¹ written in IPA. This process is typically referred to as grapheme-to-phoneme (G2P). There are a few problems and characteristics of the G2P task that are important:

- The input and output sequences are not always of the same length. It is

¹Please refer to section ?? in order to understand the terminological implications of phoneme. As it is common in research, I will stick to the term *phoneme* although strictly speaking it is not always correct. Phoneme in this case just refers to any symbol that is used to represent a sound.

difficult to align input and output which means to map one or more grapheme to one or more phoneme. Not all systems rely on aligning input and output but often it is needed to analyse the results (for example to create confusion matrices).

- Due to the open-vocabulary situation and the impossibility to cover all possible words, all systems must be able to deal with rare and unseen words [Rao et al., 2015; Bisani and Ney, 2008].
- Most of the research done in this area is limited to the English language. This is not uncommon in NLP research. The availability of English data resources and the unavailability and struggles to find data in other languages heavily influences this research. Ashby et al. [2021] report that the SIGMORPHON (Special Interest Group on Computational Morphology and Phonology) G2P shared tasks in 2020 and 2021 are the first attempt to tackle multilingual G2P.

In the following, I explain the most important model types. Often, those different types are used in combination as the different models have different advantages which can be used very well in combination.

3.1.1 Look-up dictionary

The simplest version of a G2P model is a look-up table where a grapheme sequence is stored together with its phonetic transcription. Such a dictionary is expensive to create and needs a lot of storage and has a very limited coverage. Although such a system is no longer useful on its own it can still be used in addition to other, for example, statistical models [Bisani and Ney, 2008].

3.1.2 Rule-based models

The first systems to create phonetic transcriptions of text were rule-based systems. Although rule-based systems are outperformed by more recent neural models [Ashby et al., 2021; Gorman et al., 2020], I will introduce them as they were an important step towards G2P modelling. Additionally, rule-based models can be used together with other models to reach a better performance. Rule-based transcriptions models are built using linguistic pronunciation rules. For example, if in a language, a certain word initial grapheme is always pronounced the same way, we can store a rule, that states that if that grapheme is encountered word initial it should be transcribe using a specific phoneme. This rule must of course be stored in a machine readable way

but the idea is the same. In order to be able to create such a system, one needs to collect pronunciation rules first. While there are only few languages where such rules are ready and available for the general public there are many languages where those rules need to be created first. In order to create the rules in the first place, a lot of linguistic expertise is needed. Apart from this initial effort to create the rules, a problem with rule-based approaches is the maintenance of the systems. To maintain the system, experts need to keep track of language change which is time consuming and expensive. Most languages are irregular in their pronunciation and those irregularities need to be tracked as well. Another drawback of such systems is that they might fail when presented with unseen or rare words [Bisani and Ney, 2008]. Many earlier systems published considered only one language and were not multilingual (see e.g. Toma and Munteanu [2009]).

Epitran Epitran is an example of a relatively new rule-based system. It makes use of the fact that there are languages that are more or less regularly pronounced and presents a rule-based system for G2P conversion for mostly low-resource languages. The system has the ability to provide a solution for every possible word and is consistent within its transcriptions. Epitran for all languages except English and traditional Chinese works with a map file that allows to map graphemes to phonemes. Additional pre- or post-processing can be applied that follow context specific rules [Mortensen et al., 2018].

3.1.3 N-gram Models / Statistical models

N-gram models, statistical models or joint-sequence models were used before neural models took over the field. These are sometimes referred to as traditional models. One reason why they were outperformed by neural models is that it is necessary to construct alignments between graphemes and phonemes. This is needed because one grapheme can be realized as multiple phonemes and vice versa. It is not possible to simply have a one-to-one alignment. Joint-sequence models were often used with different versions of the Expectation-Maximization (EM) algorithm [Lo and Nicolai, 2021]. The main intuition about those models is that they, in some way or other, try to statistically model the relationship between graphemes and phonemes. Typically, those models consist of two parts: first an alignment model. Second, a model that captures the relationship between graphemes and phonemes using clearly defined statistical methods. A very common model is the joint-sequence model.

Joint-sequence model The term *joint-sequence* hints already at the underlying architecture of those models: the idea is to process *joint* sequences of input and output symbols. In order to do that they must be aligned. We can then concatenate those alignments, which means to concatenate the grapheme and the phoneme which it is mapped to, and receive what is called a graphone. Using the concept of finite-state transducer (FST), we can build a model. FSTs are similar to finite-state automata. But instead of just telling whether a certain sequence belongs to a certain language (which is pattern matching), they can output none or many symbols at every step as well. This means that in the process of iterating over the sequence, they produce another sequence. Knowing that, it is easily understandable that this works well for our given G2P task. The idea is now, that we model the joint-probability of the input graphemes and output phonemes or rather our graphones. Doing this we can use the EM algorithm to find a mapping of graphemes to phonemes that is most probable. As those graphones consist of n-grams of both the input and output sequence, they themselves can be considered n-grams which is why those models are sometimes referred to as n-gram models [Bisani and Ney, 2008; Lo and Nicolai, 2021].

3.1.4 Neural models

Neural G2P models have been reported to outperform most other models [Lee et al., 2020]. Many researchers experiment with different variants of LSTM models [Lee et al., 2020; Hammond, 2021; Gautam et al., 2021; Rao et al., 2015]. But there are also other models that have been used. The most important of those will be introduced in the following.

Sequence-to-sequence seq2seq is not a model type as such but rather an architecture. seq2seq models include an encoder and a decoder or more than one of each depending on the exact implementation. Both encoder and decoder can be the same model type or different ones but all of them are recurrent neural networks (RNNs) or some variant of it. The output of the (last) encoder RNN is used as input for the decoder RNN. What makes such a model architecture powerful is that they can map an input sequence to an output sequence of a different length and different type. Generally, there is a difference between models that assume conditional independence between each output step (e.g. Hidden Markov Models) and there are models that do not make this assumption but condition the current output on the entire sequence before. seq2seq models are of the latter type. Depending on what model type is used as encoder or decoder, considering the entire preceding

input can get tricky if the input sequence is really long. This is true especially for simple RNN models. Apart from this potential limitation, seq2seq models have to wait until the full input sequence is processed before they can start decoding. This does not work well for input that gets continuously longer [Kostadinov, 02.05.2019; Sutskever et al., 2014].

RNN The important intuition about RNNs is that they process each unit of an input sequence one unit at a time. In our case, we can think of one unit as one grapheme in the input sequence. Instead of only outputting something at each time step, a hidden representation is passed on to be processed in the next time step. This sequential processing is crucial as it means that each unit gets information about the units preceding it. This makes sure that information about the preceding context is included. The output of such an RNN is therefore a manipulated version of the input sequence of the same length. A special kind of RNN is the LSTM. LSTM means long short-term memory. As their name suggests they include what we could call a memory. Instead of just adding some representation of the preceding units at each time step, LSTMs can more flexibly decide what information is added and what information should be forgotten [Olah, 29.01.2022; Kostadinov, 12.02.2017]. Many earlier neural LSTM models use a connectionist temporal classification layer to include alignment information [Lo and Nicolai, 2021]. This is a special methodology that can deal with n-to-m alignments which is the case in G2P modelling.

Transformer Transformers are seq2seq models as well. A central concept and what makes transformers powerful models is their self-attention mechanism. Also, they can process the entire input sequence in one go such that each unit is processed by a separate part of the model instead of just using the same cell for every unit. This allows to have dependencies between the different input units which means that we can model any dependencies within a sentence or a grapheme sequence. This is the reason why we call it *self*-attention as the attention is focused on other parts of the input sequence, but still on the sequence itself [Alammar, 03.01.2022].

Neural Transducer Neural transducers, as presented by Jaitly et al. [2016], extend previously used seq2seq models. They can treat more arriving input without having to redo the entire calculation for the entire updated sequence. At each time step, the neural transducer can output zero to many output symbols.

3.2 State-of-the-art G2P models

As I have to decide what model I will use to train on my language set, I will now have a look at the state-of-the-art models that can be used for G2P conversion. The Special Interest Group on Computational Morphology and Phonology (SIGMORPHON) [Sigmorphon, 2021] regularly organizes shared tasks concerned with morphology and phonology. For the years 2020 and 2021 they organized a G2P conversion task [Ashby et al., 2021; Gorman et al., 2020]. The tasks represent a first attempt at creating benchmarks for multilingual G2P conversion. Although there is other research on G2P, many recent publications have been made within the SIGMORPHON shared tasks. In the next sections, I will summarize the results and insights from G2P research with a focus on those shared tasks. As they covered so many different languages, I will use their results to compare my results to. Tables 3 and 4 list the models and their results from those two tasks.

3.2.1 Model architectures

An essential part of G2P modelling is the actual model. In section 3.1, I explained the most important theoretical basics. For this part here, I familiarized myself a bit more with strategies that work well in practice and what some concrete problems are. The methodologies mentioned below are to the most part task-agnostic. This means that they often improve results on most NLP tasks and are not specifically developed for the G2P task. Still, I think it is insightful to be aware of the great variety of approaches that nowadays technology offers.

Ensembles Many models that are used and achieve peak performance for G2P modelling are ensemble models. An ensemble is essentially just a pool of different models that are trained on the data with different settings or they are completely different models. The way such a model can be used for inference is that all of the models process the input and present their predicted results. Out of all possibilities, one prediction will be chosen that is then the final model output. In order to get the final output, an ensemble needs some kind of decision algorithm to output the best result. A disadvantage of ensembles is that the models need a lot of storage. Also, it is to some extent a bit of a *brute-force* approach as it could lead to preferring quantity over quality.

Learning edit actions Instead of learning the output a phoneme sequence, a model can also learn how to edit the input sequence in order to get the output sequence. Such a model would then output an edit sequence which can be applied to the input sequence in order to obtain the final phoneme output. The model therefore learns to create sequences of edit actions. The edit actions are typically ‘insert’, ‘substitute’ and ‘delete’. ‘Substitute’ and ‘insert’ need a vocabulary to perform their action. The problem with this approach is that there are many possible sequences of edit actions that produce the same result. For example, it is always correct to delete every unit in the input sequence and then insert every unit from the output sequence. But this does not tell us anything about how graphemes and phonemes relate. To this end, we would also want to use substitution actions to see whether one grapheme is always substituted by the same phoneme. Imitation learning is proposed as a solution for this problem. Easily put, imitation learning is a variant of reinforcement learning. The idea is that the model learns to imitate the behaviour of an expert (for example, a human expert that provides correct samples of the task in question) [Ai, 19.9.2019].

Multi-task learning What has worked well for G2P models is to use multi-task learning. This means that the model is not only trained on one task but on multiple tasks that are related. In the present case, a model was trained on phoneme-to-grapheme conversion as well [Gorman et al., 2020].

Neural models Not surprisingly, models that achieve peak performance are almost exclusively neural models [Gorman et al., 2020]. Due to their ability to process increasingly longer sequences and the above discussed techniques like attention, they are ideal for almost all NLP tasks. What type of neural model is chosen also depends on the amount of data available. Transformers are suggested to work better for larger datasets, while they are outperformed by LSTMs on medium-size datasets (a few thousand training pairs) [Gorman et al., 2020].

3.2.2 Data manipulation

A model is only as good as the data that is used to train it. While this is a very basic paradigm, in reality assuring data quality is not always easy. In this section I list a few strategies that are used to preprocess and prepare G2P data and how to deal with too little available data.

Data quality Authors mostly include a section about their preprocessing and what should be done to ensure high quality datasets. The list given below is an incomplete list of potential problems and measures taken in different settings for G2P data:

- **Exclusion of words with less than two Unicode characters or less than two phone segments** [Ashby et al., 2021]
- **Separation by script** [Ashby et al., 2021]: It is very straightforward why this is done. There is no obvious connection between the different scripts of a language and its pronunciation. It makes sense to treat different scripts as different languages.
- **Exclude foreign words with foreign pronunciations** [Ashby et al., 2021]: Foreign words in a language with their original pronunciation can add phonemes that are not in that language’s phoneme inventory. If they were to be included it would make sense to include a pronunciation adapted to the actual language.
- **Words with multiple pronunciations in word lists**: Ashby et al. [2021] excluded those words, however, it might also be possible to add Part-of-Speech (PoS) tags or other linguistic information to distinguish at least some of these words.
- **Consistent broad transcriptions** [Ashby et al., 2021]: With broad transcriptions it is important to be consistent and not use allophones. Ashby et al. [2021] did this specifically for Bulgarian. They identified allophones of a language and replaced them by their respective phoneme.
- **Linguistic variation and processes** [Ashby et al., 2021]: Some transcriptions include examples for monophthongization or deletion which are ongoing linguistic processes but should not be part of a dataset representing a standard variation. Monophthongization just means that diphthongs are replaced by monophthongs which are just single vowels. Ashby et al. [2021] dealt with monophthongization by choosing the longer of two transcriptions as this logically exclude the monophthonged version. This does of course only work if there are more than one pronunciations available. The idea behind this type of preprocessing is that G2P modelling should focus on current standard variations.
- **Tie bars**: Ashby et al. [2021] notice that some languages (English and Bulgarian) have inconsistent use of tie bars. This can be correct by replacing all inconsistencies by the tie-bar-version. It is also possible to exclude the tie bars

at all.

- **Errors in the transcriptions:** Gautam et al. [2021] noticed many errors in the WikiPron English data. They identified errors by looking at the least frequent phones and then check the word-pronunciation pairs where those phones occurred in. As the number of phones in a language is often known this can be used to check the phones in the datasets and identify uncommon ones.

Especially the task of finding errors in the transcriptions is quite tricky. It requires a lot of knowledge about the phonology and phonetics of a specific language. But this is exactly where databases like PHOIBLE can provide important information.

Low-resource setting Apart from a few well-studied examples, for most languages there is only little data available. It is therefore highly interesting and important to find solutions of how to deal with lack of data. Hammond [2021] presented a system focusing on data augmentation methods. The primary goal of their approach was to test how successful a minimalist data augmentation model would be, knowing it would most probably not outperform any of the other models. They identified two approaches that might improve low-resource models. The first one is to use as much as possible of the development set for training. The second is to train all languages together differentiating the languages only by a tag added to the word representations. The model they used was purposefully a very simple model that does not use a lot of resources. They used a seq2seq neural net with a LSTM decoder and encoder. Both LSTMs have two levels. Both of these strategies were successful for some languages but did not improve the results for others.

Yu et al. [2020] propose a data augmentation model for low-resource settings. The methodology applied in their approach is ensemble learning combined with a self-learning strategy. They use their ensemble to make predictions on unlabelled data. This newly created data is then added to the training data and the models are trained for another epoch on the extended data. This strategy worked well and produced good results.

Results in a low-resource setting are still bad when only using 800 samples for training. More research needs to be done in data augmentation techniques and improving the systems to cope with only little available data.

Reduce vocabulary size Some syllabary languages like Korean allow the decomposition into smaller units that make up the signs. Many other languages that

do not use the Latin alphabet allow to be written with Latin letters. If a reduction of the vocabulary size is possible in one of these ways, it almost always improves performance as smaller vocabulary sizes are easier to handle for models [Gorman et al., 2020].

3.2.3 Error and result analysis

In this section I will list different types of analysis that have been performed by different authors on a trained model to improve future research.

Broad and narrow transcriptions In the SIGMORPHON tasks, there are great differences in the performance of models for different languages. One possible explanation is that the datasets were a mix between broad and narrow transcriptions. As narrow transcriptions are a lot more detailed, it can be argued that this is more difficult for any system [Ashby et al., 2021]. This assumption still needs to be analysed more closely. This differing performance for various languages calls for the questions what makes a language hard to pronounce. Especially as for Georgian all models from the SIGMORPHON task reached a word error rate (WER) of 0.0. For Georgian the provided training set only contained 10,000 samples. Interestingly enough, the WER for English which was trained on 42,000 samples reached one of the highest WERs.

Linguistic error analysis Lo and Nicolai [2021] chose to perform an error analysis and try to minimize the frequent errors of a model in a multilingual low-resource setting. The analysis showed that often the model gets vowels and diacritics wrong. They extended the model in a way such that wrong vowel and diacritics predictions are punished more than other errors. Compared with the unchanged model, this extended model reached a better performance for some languages. The predictions with their model shows an improvement in vowel prediction. A further analysis showed that many errors still happen with vowels. Vowels get often confused with similar vowels. Their conclusion is that many of these errors make sense in a linguistic sense. They also tested augmenting the input data with syllable boundaries which did not improve the results.

Another type of linguistic analysis that can be performed is to analyse the data and check for uncommon pronunciations or language internal ambiguities. If a model produces a lot of wrong output because of ambiguities or uncommon data, then this is not necessarily the model's fault but just a language inherent inconsistency. As

languages are generally ambiguous, this type of analysis is very insightful to find out about *real* errors of the model. These are errors that could be derived from the data, but the model did not get there. Ashby et al. [2021] did such an analysis for the SIGMORPHON 2021 task which showed that many errors are due to language internal ambiguities.

Include linguistic information What has been suggested by Gorman et al. [2020], is to make use of phonetic resources or rule-based systems to improve the quality of current models. The advantage of such an approach is that it is specifically tailored to the problem at hand and not at all task-agnostic. Makarov and Clematide [2020] confirm this suggestion as they performed an error analysis which showed that including linguistic information such as PoS-tags might be useful.

As is always the case with such research there are many different aspects that can be tuned in order to improve model results. For my thesis I will have a closer look at how we can use phonetic features to improve models.

ISO396-3	BS20						DeepSPIN20		IMS20		BS21	CL21	UBC21		DP21	
	LSTM		transformer		pair n-gram		WER	PER	WER	PER			CL	UBC-1		UBC-2
	WER	PER	WER	PER	WER	PER										
ady ^B	28.00	6.53	28.44	6.49	32.00	7.56	24.67 ³		25.00	5.79	22.00	22.00 ²³	25.00	22.00		
bul ^A	31.11	5.94	34.00	7.89	41.33	9.05	-		22.22	4.85	18.30	18.80 ⁶				
cym (wel_sw) ^B											10.00	10.00 ¹	13.00	12.00		
ell (gre) ^B	18.89	3.30	18.89	3.06	21.78	4.05	-		18.67	2.97	21.00	20.00 ¹³	22.00	22.00		
eng(_us)											41.94				37.43	
fra (fre) ^A	6.22	1.32	6.89	1.72	13.56	3.12	5.11 ³		6.89	1.60	8.50	7.50 ⁴⁵⁶				
hbs ^A											32.10	35.3 ⁷				
hin	6.67	1.47	9.56	2.40	12.67	4.05	-		5.11	1.20						
hun ^A	5.33	1.18	5.33	1.28	6.67	1.51	-		5.11	1.12	1.80	1.00 ⁶⁷				
hye (arm_e) ^A	14.67	3.49	14.22	3.29	18.00	3.90	-		12.67	2.94	7.00	6.40 ⁷				
ice ^B	10.00	2.36	10.22	2.21	17.56	3.62	-		9.33	2.04	12.00	10.00 ¹³	13.00	11.00		
ita ^B											19.00	31.00 ³	20.00	22.00		
jpn(_hira) ^A	7.56	1.79	7.33	1.86	9.56	2.07	4.89 ⁴		5.33	1.26	5.20	5.00 ⁷				
kat (geo) ^A	26.44	5.14	28.00	5.43	37.78	6.48	-		24.89	4.57	0.00	0.00 ⁴⁵⁶⁷				
khm ^B											34.00	32.00 ¹³	31.00	28.00		
kor ^A	46.89	16.78	43.78	17.50	52.22	15.88	24.00 ¹³		26.22	4.38	16.30	16.20 ⁴				
lav ^B											55.00	49.00 ²³	58.00	49.00		
lit	19.11	3.55	20.67	3.65	23.11	4.43	-		20.00	3.63						
mlt(_ltn) ^B											19.00	12.00 ¹	19.00	18.00		
nld (dut) ^A	16.44	2.94	15.78	2.89	23.78	3.97	-		13.56	2.36	14.70	14.70 ⁷				
rum ^B	10.67	2.53	12.00	3.62	11.56	3.55	9.78 ³		10.22	2.23	10.00	12.00 ³	14.00	10.00		
slv ^B											49.00	50.00 ¹	56.00	47.00		
vie ^A	4.67	1.52	7.56	2.27	8.44	1.79	0.89 ²		1.56	0.48	2.50	2.00 ⁵⁷				

Table 3: The table lists the SOTA models from the SIGMORPHON tasks in 2020 and 2021. Superscript: model numbers. Numbers in bold: best WER. Languages in bold are in the 100LC corpus. ^A: 10,000 training samples in 2021. ^B: 800 training samples in 2021. Model explanations: table 4.

Model name	Authors	Model Architecture
CL21	SIG21: Clematide and Makarov [2021]	These are seven models in total. LSTM-based neural transducer with pointer network-like monotonic hard attention trained with imitation learning. All models 1-7 are majority-vote ensembles with different number of models (5-30) and different inputs (characters or segments).
UBC21	SIG21: Lo and Nicolai [2021]	UBC-2 outperforms the baseline. They analysed the errors of the baseline and extend it by adding penalties for wrong vowels and wrong diacritics. Errors on vowels actually decreased. UBC-2 achieved the best macro average for low-resource languages in 2021. UBC-1 included syllable prediction which did not improve the results.
DP21	SIG21: Gautam et al. [2021]	Dialpad-1: Majority-vote ensemble consisting of three different public models (weighted FST, joint-sequence model trained with EM and a neural seq2seq), two seq2seq variants (LSTM and transformer) and two baseline variations.
BS21	Ashby et al. [2021]	The baseline in 2021 is a neural transducer trained with imitation learning similar to a model submitted in 2020 [Makarov and Clematide, 2020]. As they are so similar, I did not include the model twice, but only the newer one.
DeepSPIN20	SIG20: Peters and Martins [2020]	DeepSPIN-2,-3,-4: Transformer- or LSTM-based seq2seq models with sparse attention. Add language embedding to encoder or decoder states instead of language token.
IMS20	SIG20: Yu et al. [2020]	IMS: Self training ensemble of one n-gram-based FST and three seq2seq (vanilla with attention, hard monotonic attention with pointer, hybrid of hard monotonic attention and tagging model).
BS20	Gorman et al. [2020]	The baseline for the task in 2020 consisted of three different model types. An LSTM, an transformer and a pair-n-gram model that is based on a weighted FST. All of them were outperformed by other models except for Lithuanian. The LSTM performed best among these three.

Table 4: This table presents the state-of-the-art G2P models which will be important to compare my own results to. The results for these models can be found in table 3

3.3 Evaluation metrics

In order to judge whether a computational system performs wells or not we need an appropriate metric. Mostly, you need a reference transcription and a system output that can be compared to the reference and then the difference can be quantified and be given a score. The evaluation of phonetic transcriptions depends on whether the system output and the reference are sentences or single words, i.e. character sequences. The most common metric to evaluate the former is the WER. The lower

the score, the better the model. If the WER is 0, this means that the texts are exactly the same. The idea behind the score is that we can capture the cost that it takes to transform the system output into the reference phoneme sequence. The following formula is used to calculate the WER:

$$WER = \frac{S + I + D}{N} \quad (3.1)$$

In equation 3.1 S stands for substitution, I for insertion, D for deletion and N denotes the total number of words in the reference sequence. Those numbers can be calculated by using an algorithm to get the edit distance. It is quite common to multiply the resulting number with 100 [Gorman et al., 2020]. Note that the WER can be more than 100%. This happens if, for example, there are a lot of additional insertions or deletions in the system text. If the system output and reference are character sequences, the score is called character error rate (CER). It is calculated in the exact same way as the WER, but instead of words everything is calculated on character basis. In the phonetic transcriptions setting, the CER is typically replaced by the phone error rate (PER) to match the correct terminology. The calculations are not changed. In a multilingual setting, it is sometimes necessary to have a score for the entire system covering more than one language. In such cases it is custom to use a macro-averaged WER or CER. This means that the sum of the scores for each language is divided by the number of languages [Leung, 24.6.2021].

In the case of G2P conversion, the WER actually just reflects the rate of wrongly predicted words, as one sequence consists of only one word. It is therefore just 1 (or 100) minus the accuracy. The accuracy is the rate of the correctly predicted words (which means we divide all correctly predicted words by the total number of words). It is important to note, that the WER and the PER suggest slightly different things. If each word has exactly one wrong character this means that the WER would be 100 as all words are predicted wrongly. Now, if the words are really long, the PER score would be very good compared to the really bad WER. If the words are rather short, the PER would get worse as well but the WER would just stay the same. So, when analysing the results it is important to pay close attention to how those two metrics relate.

3.4 CMUSphinx

For my personal experiments, I decided to use the CMUSphinx seq2seq G2P model. This model has been used in the SIGMORPHON 2021 task as part of the Dialpad

ensemble [Gautam et al., 2021]. It was not used on many languages but promised a good performance which is why I decided to use this model for this present thesis. The CMUSphinx model is a transformer-based acas2s model implemented with tensorflow. Unfortunately, there is not a lot of information available online about the model. There exists a pre-trained version of the model for acg2p, however, they use a transcription format other than IPA which means it cannot be used for our dataset [GitHub, 03.02.2022].

3.5 Unicode and the International Phonetic Alphabet

When it comes to representing characters in a machine-readable format things get very tricky, very quickly. In order to understand this fundamental problem it is necessary to understand the basic concept behind Unicode and how characters are represented behind the scenes. Moran and Cysouw [2018] present a neat overview in their book. As discussed in chapter 2, there are many different kinds of what we typically call letters, graphemes, characters or signs². Just as a human writer must be able to uniquely identify each different grapheme, so must a computer. The most widely spread standard to represent scripts is called Unicode. Graphemes are mapped to unique numbers that can be rendered differently depending on the font and the context. There are different stages of representation until a graphemes can be represented on screen:

CODE POINT A unique numerical, non-negative value usually expressed as a hexadecimal number (U+0000). Allows one-to-one mapping between letters and codes. Each code point has a set of properties attributed to it. Properties like the script, uppercase or not, etc.

CHARACTER An abstract representation of the shape of the grapheme. Can in theory not be represented visually, as this includes a font. A Unicode character is *not* the same as what we would call a grapheme in different writing systems.

GLYPH The rendered and therefore visual representation of one or more Unicode characters that can be identified by its code point(s). A glyph is rendered in a specific font in a specific context. No matter how different it looks to the user, for Unicode all different representations of one code point are exactly the same. Sometimes one character is represented as two glyphs. It is important

²Please note that I will from now on use grapheme to denote the smallest meaningful element of any writing system. Grapheme does not imply any specific writing system nor does it take the Unicode background into consideration. If I wish to distinguish the Unicode specifications I will use the correct Unicode term as described in this section.

to note here, that the exact visual representation of a glyph is not at all defined by its code point. This means that the exact same glyph can represent more than one code points. This happens sometimes in the IPA. An example is the post-alveolar click:

- ! : this glyph represents an exclamation mark with unicode code point U+0021.
- ! : this glyph represents an post-alveolar click with unicode code point U+01C3.

It is striking that these two look exactly the same. Things like this become important when, for example, I want to count the different characters in a text.

Unicode code points are often organized in blocks. A block can, for example, contain all letters of the Latin script. Those blocks are helpful although not always consistent. The IPA is represented in a basic block but many IPA symbols are actually found in other blocks. Confusion often arises from the fact, that one human-perceived grapheme is sometimes represented as more than one code point.

GRAPHEME CLUSTERS A grapheme cluster is one visual letter that is represented in Unicode as more than one code point. This is the case for diacritic marks. A problem with grapheme clusters is that some diacritic marks, which are characters that cannot really exist without any base character, are underspecified. This means that when we want to split into clusters, we do not know if that character belongs to the left or the right base character. This is the case for many characters in the IPA. Unicode does not specify these but leaves it to the user to create tailored grapheme clusters. An example is the superscript aspiration ^h. It is typically written after the respective aspirated character, but there is no reason why I could no be written before the respective character (except for convention).

PRECOMPOSED CHARACTERS Note that sometimes, grapheme clusters can be pre-composed and the combination of those two or more characters is assigned a new number. These clusters can be problematic if in a specific context, the graphemes should not be clustered but read separately. An example is the German ‘ä’.

Additional complexity is added through the possibility of Unicode to create Unicode locales. These allow users to specify language- or writing-system-specific cases. An additional challenge is that of picking the right font. Our standard font format

can only contain about half of all the Unicode code point. It is therefore simply not possible to display the entire set of Unicode characters with one font. Many problems encountered with displaying writing systems are somehow connected to the font rather than Unicode itself [Moran and Cysouw, 2018]. Moran and Cysouw [2018] list a few more ‘pitfalls’ that one might encounter when dealing with Unicode.

For the present thesis, this topic is relevant for multiple reasons:

1. The IPA contains many special characters and many diacritics.
2. The language data is available in many different scripts.

It is crucial that all data files, be it phonetic or ‘normal’ scripts, are formatted and read correctly. Or rather that the encoding and processing is made transparent as often there is not one correct way of how to treat IPA characters.

Segments library Moran and Cysouw [2018] present a python library called `segments`³ that can be used to process phonetic text. It includes a method to segment IPA strings into grapheme clusters that make sense in a IPA context. This means that, for example, diacritics are put on the base character but also that characters connected by a tie bar are displayed as one unit. I will make use of this library later when processing the data.

3.5.1 Unicode normalization forms

The above explanations make clear that there are considerable differences in what a human reader perceives and in what happens in the background. Unicode therefore provides normalization forms that can help to process written data. Unicode publishes extensive explanations along with their standards which also includes those normalization forms. I will therefore not explain everything in full detail as this is done so already online [Inc., 27.08.2021]. What is important is that each normalization form results in very different behaviour if a text is processed. There are two important aspects to normalization. One is that we can have decomposed or composed characters. The second is that we have a compatibility form and a non-compatibility form. In a decomposed string, we split the characters into their individual components. This means that characters with diacritics are split up into two or more characters. This means that a characters that had originally assigned one code point can in a decomposed form have more than one code points. In a

³<https://pypi.org/project/segments/>

composed normalization usually precomposed forms are kept. This means that some parts can be similar or the same to the decomposed version but if for a character there exists a precomposed version, this one is usually used. If a normalization is according to compatibility decomposition, this means that any formatting is removed such that we receive the underlying character in its original form. Superscript characters are then shown normal. How exactly these normal forms work is not always equally important, but what is absolutely crucial to make sure that when characters are compared or counted, the same normalization form is used. The names for these normalization form are as follows:

NFD : Canonical Decomposition

NFC : Canonical Decomposition, followed by Canonical composition

NFKD : Compatibility Decomposition

NFKC : Compatibility Decomposition, followed by Canonical Composition

4 Experiments: Data Collection and Preprocessing

The first practical part of this thesis is concerned with data collection. Although phonetics is an important sub-area in linguistics, phonetic transcriptions are hard to find. If there are any transcriptions available, there are various hindrances that prevent it from being used as is. In the following, I outline the different data types which are available and the different strategies that are used to convert the data into one well-formatted corpus. There is quite a famous set of phonetic texts which is a collection of short stories called *The North Wind and the Sun*. Those stories were the starting point for my search for data. The reason for this is that the short stories are available in many different languages. As the corpus this thesis is based on is a large multilingual corpus, the availability of data in many languages was one of the key criteria to look choose the datasets.

One of the first things I have noticed while collecting my datasets is the abundance of different codes and names to refer to the languages. I have decided to use the ISO 396-3 convention, not to be confused with the ISO 396-2 convention which uses sometimes completely different codes and does not cover all of my languages. ISO 396-3 distinguishes some more dialects and variants. This also means that throughout my research I paid close attention to what variant was actually referred to. For some cases, it seems that there exists a dataset for a language which is in the 100LC, but actually it is not the same variant. To make sure I pick the same code and language, I used the Glottolog as reference¹.

4.1 Transcription sources & formats

Phonetic transcriptions of various languages are available from different sources in different formats. From what I have found out in my research phonetic transcriptions are available as either full texts or a word lists.

¹<https://glottolog.org/glottolog>

Full Text For the task of G2P conversion, phonetic transcriptions in the form of fully transcribed texts would be ideal. As became clear, it is hardly possible to find those. There is plenty of material describing how different languages can be transcribed but those rarely contain fully transcribed text. If they do, it is mostly limited to one or a few sentences. Additionally, some texts include short descriptions where certain pronunciations rules are explained which are not included in the transcriptions (especially stress). Apart from *The North Wind and the Sun* (NWS) short texts described in section, there are no full texts available in multiple languages that could be used.

Pronunciation Dictionaries Another data type that is found quite often are word lists. Those are sometimes referred to as pronunciation dictionaries. However, these often mean that there are words mapped to an audio representation which is not what is meant in this present case. Pronunciation dictionary in this present case refers to the mapping of an orthographic word to its pronunciation using phonetic symbols. Although such lists are very handy, especially as they can easily be used to train a transcription model, transcriptions of individual words and of entire texts are not exactly the same. There are two major problems:

- Pronunciation depends on the context of the word in question. Word forms are ambiguous and sometimes their pronunciation differs given on their specific context.
- Phonetic boundaries are not always equivalent with word boundaries. Spoken language sometimes merges certain words which leads to one phonetic unit.

Data in the form of word lists can be found quite a lot when compare to the availability of full text. An example is the Nidaba² website that presents a few word lists in different languages. What becomes clear when looking at this data is that there are two important limitations: First, there are only very few languages available that are in my corpus. Second, there are different transcription conventions used.

Transcription conventions No matter how the transcriptions are formatted, a crucial factor is the transcription convention. The IPA is a well-known convention but there is a lot of data that is available using a different convention. Although it might be possible to convert one convention into another convention this is very tedious and often those conventions are made specifically for one language or a group of related languages.

²<http://nidaba.co.uk/Contents/OriginalWordList>

Iso 639-3	Language name	Type WikiPron	# Words WikiPron	Type NWS
cmn	Chinese	broad	133,686	unk
deu	German	broad	34,145	broad
deu	German	narrow	10,984	narrow
ell	Greek (Modern)	broad	10,547	unk
eng	English US	broad	57,230	broad
eng	English US	narrow	1,633	narrow
eng	English UK	broad	60,422	-
eng	English UK	narrow	1,284	-
eus	Basque (Goizueta)	broad	1,742	broad
fin	Finnish	broad	69,015	-
fin	Finnish	narrow	69,008	-
fra	French	broad	56,911	unk
hin	Hindi	narrow	9,563	-
hin	Hindi	broad	10,812	unk
ind	Indonesian	broad	1,555	unk
ind	Indonesian	narrow	2,637	-
jpn	Japanese (Hiragana)	narrow	19,689	-
kat	Georgian	broad	15,123	broad
kor	Korean	narrow	14,141	unk
mya	Burmese	broad	4,631	broad
rus	Russian	narrow	402,586	unk
spa	Spanish (Castilian)	broad	60,677	broad
spa	Spanish (Castilian)	narrow	52,190	narrow
spa	Spanish (Latin America)	broad	48,649	-
spa	Spanish (Latin America)	narrow	41,845	-
tgl	Tagalog	broad	3,321	-
tgl	Tagalog	narrow	1,915	-
tha	Thai	broad	15,050	unk
tur	Turkish	broad	1,789	unk
tur	Turkish	narrow	1,812	-
vie	Vietnamese	narrow	15,240	unk
zul	Zulu	broad	1,677	-

Table 5: In this table I list all languages that I am using for my experiments.

4.2 The 100 language corpus

The basis of the data used in this thesis is a corpus provided by the SPUR lab at the University of Zurich (UZH). The text group of the Language and Space lab at the University of Zurich maintains a project that provides a multilingual corpus consisting of 100 language text samples [SPUR project] which is referred to as 100LC (see section 4.2). Those 100 languages are meant to be representative for all the world’s languages. It is therefore meant to give insight on relations, similarities, differences or properties of individual languages or language families. Specifically, their goal is to use quantitative methods like statistical modelling, machine learning and information theory to study language variation and compare languages. The corpus contains 100 languages which are proposed by Comrie et al. [2013]. This is an online book that contains different chapters each of which shows a different linguistic feature including a map which shows the distribution of that feature over the world’s languages. While the number of languages presented on the individual maps depends on the amount of research done in a specific area, the sum of all maps gives quite an impressive overview on the structure of nearly half of the world’s languages. Out of the 2676 languages a sample of 100 languages was chosen. This sample does not contain too many languages from one area, neither does it contain too many languages from one family. Those are the 100 languages that are in the corpus. Not considering the aforementioned criteria of maximizing genealogical and areal diversity can lead to misleading results in multilingual analysis. Figure 1 shows the distribution of the corpus on a world map. The different icons show the genus of the languages which is a classification of languages defined by the World Atlas of Language Structures (WALS) team that maintains the language description collection. The interactive map can be viewed online [100-language-sample]. Table ?? in the appendix A shows all languages that are in the 100 language corpus. None of the text samples are provided by WALS. The entire corpus is provided by the SPUR team that collected the corpus over the last few years and is continuously working on and with it.

4.3 Data used for this thesis

Based on the outcome of my research on the availability of phonetic data, I decided to use two datasets for my experiments in this thesis. I will introduce both of them below.

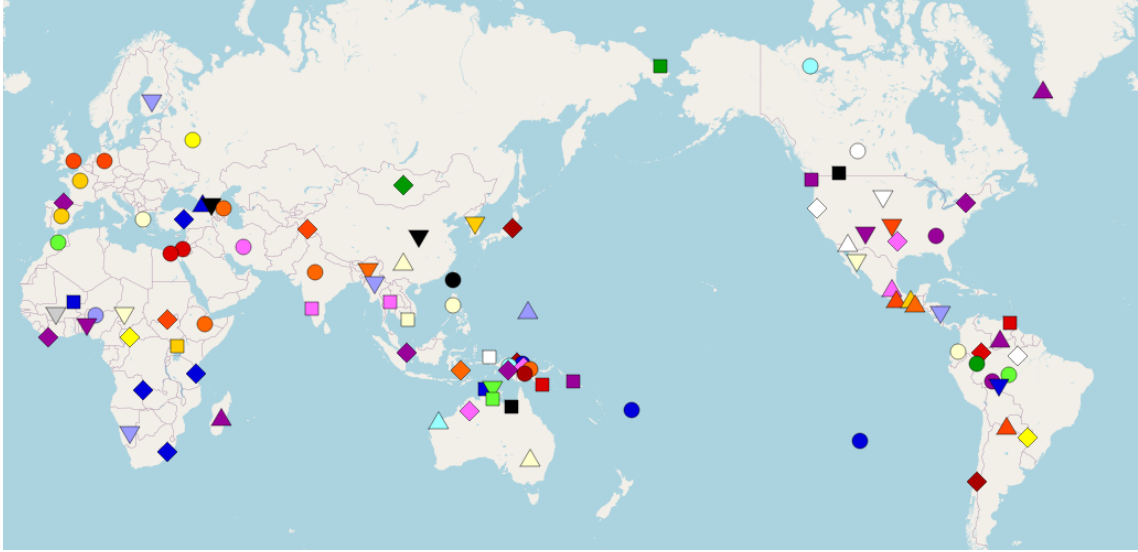


Figure 1: WALS - Map that shows the 100 Languages

4.3.1 WikiPron

A very recent project that publishes pronunciation lists is WikiPron. The WikiPron project [Lee et al., 2020] is an open-source Python mining tool to retrieve pronunciation data from Wiktionary. Their database contains 1.7 million word/pronunciation pairs in 165 languages. Both, the database and the tool, are freely available online. Apart from the mining tool and the database, WikiPron can be used for grapheme-to-phoneme modelling. The WikiPron data for one language is always structured the same. It is a tsv-file that has graphemes as a first column and corresponding phonemes as the second column. While the graphemes are just listed as-is, the phonemes are split using the segments library which I shortly presented in section 3.5. The phoneme segments are separated by white spaces. The WikiPron data has already been used in shared tasks which means that results on this data can easily be compared to other results. For a shared task in 2021 organized by SIGMORPHON, the WikiPron data was improved and additional scripts were added based on feedback and findings from a similar task in 2020. One major improvement was concerned with languages written in different scripts. WikiPron supports now the detection of different scripts and languages can be sorted according to those scripts. For some languages, there is a filtered version of either the broad or the narrow transcription available. Whenever possible, I used this one.

4.3.2 The North Wind and the Sun

Quite a well-known phonetic corpus is a collection of short stories. The Journal of the International Phonetic Association (JIPA) continuously published different phonetic transcriptions of a short story called *The North Wind and the Sun*. The story is a fable said to be written by Aesop and has been translated into many languages. Additionally, for many languages there exists a phonetic transcription. As I have mentioned before, many of these texts are available already transcribed and free to use [GitHub, 04.02.2022]. Table ?? shows the languages for which the short story is available and which are also in the 100 language corpus. Baird et al. [2021] performed an analysis on these texts to find out how many tokens we need to cover a phonetic inventory of a language adequately. What they have found out in their study is that those short stories are by far not long enough to give a good picture of the phonetics of a language. This is of course problematic if those texts are used to demonstrate how a language works phonetically. That said, I will use these stories with care and only as an additional dataset to have more variety.

Iso 639-3	Type	Variation	Language
arn	broad and narrow		Mapudungun
cmn		Pekinese	Mandarin Chinese
deu	broad and narrow	North German	German
ell			Modern Greek
eng	broad and narrow	Americsn	English
eus	broad and narrow	Goizueta	Basque
hau	narrow		Hausa
heb			Modern Hebrew
hin	narrow		Hindi
ind			Indonesian
kat	broad and narrow		Georgian
kor			Korean
mya			Burmese
pes			Western Farsi
spa	broad and narrow	Castilian	Spanish
tha			Thai
tur	broad	Istanbul	Turkish

Table 6: The table shows a list of all the short stories *The North Wind and the Sun* that are available as phonetic text and whose languages are in the 100 language corpus.

Below, I include the English version of the short story.

The North Wind and the Sun were disputing which was the stronger, when a traveler came along wrapped in a warm cloak. They agreed that the one who first succeeded in making the traveler take his cloak off should be considered stronger than the other. Then the North Wind blew as hard as he could, but the more he blew the more closely did the traveler fold his cloak around him; and at last the North Wind gave up the attempt. Then the Sun shined out warmly, and immediately the traveler took off his cloak. And so the North Wind was obliged to confess that the Sun was the stronger of the two.

4.3.2.1 Transcription of NWS stories

A collection of the NWS stories is available in a handbook of the JIPA which is only available as a pdf scan of the original book [Press, 2010]. Luckily, most of those texts have been transcribed and made available by Simon Greenhill [GitHub, 04.02.2022]. At least the phonetic part of it, the original version was still not available already transcribed. Before I knew about the availability of these texts, I did some research on Optical Character Recognition (OCR) for IPA texts and manual transcription. While OCR is technically possible it turns out to be very difficult for IPA characters. There are tools that include IPA character recognition like the ABBYY FineReader which can be acquired for a fee. The CL institute at the UZH owns a version of the ABBYY tool but this version does not include the IPA module. I ran the ABBYY version on a JIPA pdf containing said phonetic transcriptions but the result could not be used. Mostly diacritics and special phonetic symbols were not correctly transcribed. These are exactly those characters that make it difficult to transcribe IPA manually, so there is not point in using this tool. There are also open source tools. One of which is called tesseract. tesseract does not include the IPA alphabet. It is possible to train the model to include the IPA alphabet but this would need appropriate training data³.

As it was not possible to use OCR to get the texts, a next approach is to transcribe them. I experimented with a software called Transkribus⁴ to manually transcribe the pdf scans. The software allows to make use of neural Handwritten Text Recognition (HTR) models. There exists no pre-trained model for transcribing IPA characters, but I trained my own while transcribing some of the documents. On the website they mention that, ideally, training needs 5,000 - 10,000 words already transcribed. Although my available data is not nearly enough to train a reliable model (the

³<https://github.com/tesseract-ocr/tesstrain>

⁴<https://readcoop.eu/de/transkribus/>

short stories have around 40 - 100 tokens), it was a great help to transcribe. As the scans were not handwritten but machine typed text, the model still reached a surprisingly good quality. As an example: For the Hebrew transcription, the model reached a WER of 34.52 and a CER of 6.11. The two main mistakes were made for two characters that were not even in the training data. The quality of the scans differed quite a lot which had an influence on the performance of the model as well. After transcribing another document I trained the model again and transcribed a few more documents. The transcriptions got continuously better such that in the end I did not take me nearly as much time as in the beginning. Most of the errors resulted from characters that had not been in the previously described documents.

It was interesting to see that it is relatively easy to transcribe IPA text when using the right software. The original texts were easier to transcribe as they do not contain as many diacritics or other special characters. Still, for some texts like Chinese or Hindi, I needed help from native speakers as it is nearly impossible to find the matching sign if the language is unknown.

4.4 PHOIBLE

A way to analyse or use phonetic corpora is to use phonetic features to represent each phoneme. These features are a list of properties that are overlapping with the phonetic description of each phoneme that I have mentioned before when talking about vowels and consonants. It is a minimal list that can be used to describe unique phones. There exists an online database called PHOIBLE [Moran and McCloy, 2019] that contains over 3,000 phonemes for more than 2,000 languages. PHOIBLE includes a feature system that can describe each phoneme uniquely. In total, there are 37 features that are used to describe the phonemes. Each of the features can take on three values: either ‘+’ (applies to this phoneme), ‘-’ (does not apply) or ‘0’ (not applicable). In addition to the phonetic features each phoneme has other features like for example what allophones are used for it in a specific language. The entire PHOIBLE inventory is structured around languages. It lists all phonemes including features for all languages that are in there. This means that one phoneme can be listed for more than one language, but the features will always be the same. All the PHOIBLE phonetic features are listed in table 7. Some of them might sound familiar from the linguistic background chapter. The exact description of each and every feature is not important at this point. The main point is that these features allow to uniquely describe each phoneme.

For this present thesis, PHOIBLE does not only serve as a phonetic feature database

tone	tap	strident	epilaryngealSource
stress	trill	dorsal	spreadGlottis
syllabic	nasal	high	constrictedGlottis
short	lateral	low	fortis
long	labial	front	raisedLarynxEjective
consonantal	round	back	loweredLarynxImplosive
sonorant	labiodental	tense	click
continuant	coronal	retractedTongueRoot	
delayedRelease	anterior	advancedTongueRoot	
approximant	distributed	periodicGlottalSource	

Table 7: This table displays all the features that are use in PHOIBLE to describe one phoneme. Each feature is either ‘+’ (applies to this phoneme), ‘-’ (does not apply to this phoneme) or ‘0’ (not applicable).

but as a phonetic reference database in general. As it contains a very complete list of phonemes for each language it is a great help to clean phonetic data and detect mistakes or uncommon transcriptions.

4.5 Pronunciation dictionary coverage

As I am using two different datasets I want to find out how these two datasets relate. In order to do that , I am using the WikiPron lists to write the NWS stories. In a way, I used the WikiPron data as a look-up table to create phonetic transcriptions for the NWS stories. By comparing the text produced by the WikiPron data to the reference transcriptions form the JIPA articles, I could calculate the coverage, the WER and the PER score. The comparison and the analysis of the metrics gives the following insights. In addition, I manually checked some errors which gave me insights about the word lists in general:

- For some NWS transcriptions it is not clear whether their transcription is narrow or broad. On the other hand, sometimes there is no broad or narrow word list available for a specific language but only one of those. For the short stories where the type was unclear, I tried both word list types if those are available. The analysis shows that transcriptions written with broad lists give better results if the type of the reference is unclear. See, for example, Indonesian (ind) or Hindi (hin) in table 8. As this is the case, it makes sense

to treat the unknown transcriptions as broad.

- The IPA allows to transcribe intonation segments. In German, those correspond mostly to punctuation marks like end of sentence symbols or commas. But this must not be true for every case. It needs to be decided if those should be kept or potentially deleted.
- The pronunciation dictionaries sometimes included duplicates with different pronunciations. This is not surprising but still it needs to be handled well. A solution is to simply delete duplicate words.
- In order to do this very simple experiment, it is necessary to tokenize the texts. This works well for languages using the Latin script. For languages like Chinese or Korean this is more difficult to accomplish. However, this issue needs to be tackled to create G2P models anyway. I will therefore not explore this issue here.
- Interestingly enough, for some language the coverage is really low although there are quite a lot of words in the word list. This is the case for German (narrow) where the coverage is only at 22% and the word list contains more than 10,000 words. A manual analysis showed that some frequent words like ‘sich’ and ‘und’ are not in the word list. It becomes clear that a few thousand words are not enough to reach reasonable coverage. 10,000 - 15,000 words seem to be a lower bound for covering around 50% of this short text.
- For most of the languages the PER is lower than the WER. This is a good sign, as it suggests that if a word is in the list, it is at least partially spelled the same as in the reference text. Still, it is surprising that for broad Spanish, the WER is actually lower than the PER. Results like this show that even if the words are covered, their phonetic transcription might be spelled differently. This is related to the fact that the IPA is not really standardized. It always depends on the person who transcribe a text.

The results from this experiment are summarized in table 8. Generally it is good to see that most texts are at least partially covered by the pronunciation dictionary. It will later be interesting to see how the neural models perform on these short texts.

4.6 Language profiles

Once I collected my datasets, I wanted to find out what characters they include. I collected phoneme and grapheme profiles of the data and compared it to the

Iso 639-3	Coverage	WER	PER	Type ref	Type list	# Words list
cmn	85.15	93.07	59.26	unk	broad	133,686
deu	75.00	72.22	52.67	broad	broad	34,145
deu	22.22	97.22	84.85	narrow	narrow	10,984
ell	26.32	84.21	87.74	unk	broad	10,547
eng	92.04	83.19	37.27	broad	broad	57,230
eng	7.08	100.00	108.35	narrow	narrow	1,633
eus	5.75	96.55	97.95	broad	broad	1,742
ind	22.22	96.30	88.04	unk	broad	1,555
ind	1.85	100.00	101.69	unk	narrow	2,637
kat	44.29	90.00	73.85	broad	broad	15,123
mya	7.50	97.50	97.70	broad	broad	4,631
spa	64.95	46.39	48.64	broad	broad	60,677
spa	35.05	98.97	65.92	narrow	narrow	52,190
tha	83.85	88.20	40.62	unk	broad	15,050
tur	18.46	100.00	88.89	unk	broad	1,789
tur	6.15	100.00	92.53	unk	narrow	1,812
hin	31.75	100.00	84.00	unk	narrow	9,563
hin	57.94	93.65	69.81	unk	broad	10,812
kor	18.64	100.00	51.97	unk	narrow	14,141
fra	86.11	51.85	40.76	unk	broad	56,911
vie	96.58	79.49	48.66	unk	narrow	15,240
rus	94.79	95.83	56.30	unk	narrow	402,586

Table 8: The table shows the coverage, WER and PER when the pronunciation dictionaries are used to write *The North Wind and the Sun*.

PHOIBLE dataset. A profile lists all used characters or tokens (this depends on your chosen settings) for a text and lists its frequency. All profiles were collected for each language and each dataset separately. For each language and each data type I got these three lists:

- Grapheme list: contains all graphemes in that language. Characters that need a base character like diacritics are shown together with their base character.
- Phoneme list: contains all phonemes in that language. Again, diacritics and similar characters are shown with their base characters.
- Phoneme cluster list: Phonemes can be clustered into bigger sound groups. How to do this, is an ongoing discussion, but I used the segments library to

get the clusters (compare Moran and Cysouw [2018])

Having this overview for the characters for each language allowed me to compare the character vocabulary to the characters or character clusters available in the Phoible dataset. This comparison showed that quite a few characters are missing that are included in the WikiPron data and the short stories. My observations are listed below.:

- Characters that are not included in the official IPA chart: There are sometimes characters included that are no part of the IPA. There might be a reason why the authors of the transcriptions decided to use this special character to denote a particular sound, but this is not always known. A possibility is to try and map it to a character that is available in Phoible and that represents a similar sound (or even the same sound actually).
- Tie bars: The creators of Phoible decided to exclude tie bars because they add no real value to the transcriptions⁵.
- Stress marks: Stress marks are not represented in Phoible as they do not represent a sound. They are included in the NWS short stories. The same is true for other suprasegmental elements.
- Tones: Even within the IPA there exist different conventions of how to represent tones. Some are better suited for different languages. Apart from different ways of representing tones, it is not always sensible to have tones represented. Mostly, tones are not written as speakers of that language know how to pronounce the tones. So, the question is whether it is necessary to include the tones at all. When looking at the written representation it does not matter what the tones are as the basic phonemes do not change. This is of course different when the phonetic representation is mapped to a spoken representation.

I conducted this preliminary analysis at an early stage of my thesis. In chapter 6 I will clean the datasets to use them for training and testing. At this point, I will give a more detailed list of what characters are in my datasets but not included in PHOIBLE.

⁵<https://phoible.github.io/conventions/>

5 Experiments: Phonetic & Orthographic Word Length Correlation

Multilingual corpora, like the one I am using for the present thesis, are used more and more in NLP research. This leads to the question how well these corpora represent linguistic diversity. Having a linguistically diverse corpus is important as multilingual does not necessarily mean that the languages in the corpus are very different from each other. While this is often neglected in present NLP research there are approaches to measure linguistic diversity. No matter how we intend to create such a diversity score, we will eventually need to compare languages and tell whether they are similar or not. This is needed as a multilingual corpus can only be diverse if the languages it contains are not all very similar. But they need to be similar to languages that are not in the corpus as those need to be represented as well. In order to compare languages, we need to be able to describe them properly. There are many different characteristics like language family, script, grammatical features and other characteristics of languages that can be used to describe a language. The entire process of describing languages and comparing them would take up another entire thesis. But with the help of my corpus I can make a contribution to push work on linguistic diversity.

5.1 Mean word length correlation

A very simple characteristic to describe a language is to calculate the mean word length of a language.

Although this task sounds very simple there are in fact multiple challenges to tackle before it is possible to accomplish the task:

- **Tokenization:** In order to calculate word lengths, the texts need to be tokenized properly. This sounds extremely trivial, but in reality this can be a

huge problem.

- **Preprocessing:** Different scripts use different punctuation symbols. Should we exclude those? For example hyphens between two words. Do they count as part of the word? What to do about tones in the phonetic transcription?
- **Tokenization on character-basis:** For the phonetic texts it was of particular importance to decide how the individual tokens should be tokenized on character-basis to calculate the length.

As the texts I am using for this experiment are rather short, many of those challenges are already resolved because the phenomenon is not present in the corpus. Still, it is important to be aware of these challenges if further experiments are to be conducted.

Iso396-3	Avg len orthographic	Avg len phonemes	Type
aey	5.21	5.5	unk
arn	4.81	4.65	narrow
cmn	1.59	4.44	unk
deu	5	4.35	narrow
ell	4.62	4.23	unk
eng	4.19	3.46	narrow
eus	5.3	4.98	narrow
fra	4.55	3.18	broad
hau	3.8	4.07	narrow
heb	6.62	6.57	unk
hin	3.53	3.93	narrow
ind	5.92	5.25	unk
jpn	1.59	3.77	unk
kat	5.99	6.32	narrow
kor	2.85	6.56	unk
mya	10.22	8.15	unk
pes	3.99	5.03	unk
spa	4.62	4.36	narrow
tha	3.25	3.03	unk
tur	6.74	7.02	broad
vie	3.24	3.87	unk

Table 9: This table shows the mean word lengths for the NWS phonetic and orthographic.

To calculate the mean word lengths I exclusively used the narrow transcriptions if those were available. If not, I used the broad transcription or just the one I had

with unknown transcription type.

6 Experiments: Automatic Grapheme-to-Phoneme Conversion

In this chapter, I present the experiments I conducted to obtain models for phonetic transcription. As a result of my first practical part in chapter 4 I presented two datasets that I will now use to perform these experiments. The NWS corpus is much smaller which is why I will use it as a test dataset only. This means that I will train any model on the WikiPron dataset. As the NWS corpus is quite well known among linguists or at least among phoneticians, it will hopefully give interesting insights when testing the models on these short texts. Table 5 shows all languages that I am using in my experiments. Note that WikiPron continuously adds data to their repository. This means that there might be new languages that are in the 100 language corpus and that I did not use. Additionally, there were languages which had a WikiPron dictionary that was smaller than 1,000 pairs. This is simply not enough to train a decent model which is why I excluded these. If the transcription type of the NWS story is unknown, I only added it to the broad type of the WikiPron dictionary, such that there are no duplicates in the table.

The model that I am using for my G2P experiments is explained in section 3.4. Setting it up was not very easy as there were issues with the tensorflow version and some other dependencies. Also, they do have a pre-trained model, but this uses a completely different transcription convention than IPA. So we cannot use this pre-trained model. But it is of course well possible to train new models for the languages I am using. Generally, broad and narrow transcriptions are treated as separate languages and thus trained separately as well. The same is true for dialects if there is any information available. American and British English are trained in different models. In order to compare my results to already existing models, I will use the results of the SIGMORPHON tasks in 2019 and 2020. They present results for quite a few languages. Also, they use the same data type. They cleaned some of the data for some languages and made these datasets available. Whenever a filtered version is available I use that one as the basis for my experiments.

6.1 Reproducability

All scripts that I used to preprocess, train and evaluate my models are found on my GitHub. I installed the model on my machine as specified on the model's GitHub¹. In order to reproduce my results I include a table below that lists all links to my relevant scripts and settings including the settings for the virtual environment.

6.2 Training settings

As I have shortly mentioned before, some WikiPron word lists have less than one thousand word pairs. I marked them as low resource and only used the high resource languages to train my models. There are different settings which I will use to train the models and analyse their performance. The settings are chosen in a way such that they are built on each other. The first experiments are set up very simply and without much effort. Then I continuously add more complexity. Below I added a short description of each setting as a n overview. I will add more details on the individual steps in separate sections.

There are a few properties that are always the same for all models:

- The CMU model splits the data automatically if no specified otherwise. The automatic split is 85% training data, 5% development data and 10% test data. I did not change this split.
- The hyperparameters are left as default if not mentioned otherwise. More on those can be found on the model's GitHub².
- The training time for the short model with 10,000 steps is about 20 minutes. The long models took another six hours to complete the training.

Setting 1: Baseline Short This is the most basic setting. I will train a model for each language to get a baseline result. The model is trained with the least amount of effort. The model is trained for the minimum number of steps which is 10,000 steps in this case. First, I trained it on those languages where I have results from the SIGMORPHON 2021 challenge. The results are compared in table ??.

¹<https://github.com/cmuspinx/g2p-seq2seq>

²<https://github.com/cmuspinx/g2p-seq2seq>

Setting 2: Baseline Long This setting is similar to setting 1 except that the model is trained as long as possible for each language. All models have been trained for 200,000 steps and the default settings. This setting is supposed to show if training for a considerably longer time changes the results a lot or not.

Setting 3: Baseline clean short I will train another model that is the same like the Baseline Large, but I will use the cleaned WikiPron data. What I will clean is described below (section 6.3).

Setting 4: Baseline clean long The same as setting 3 except that the models were trained for as long as possible like in setting 2.

Setting 5: Feature input version 1 short The final experiments will be with phonetic features as input. I use the cleaned WikiPron data from setting 3 and add the features to it. In section 6.4, I explain how I encode the features. Before training this set of models, I did some experiments with other features than the final ones I used here. All of this is explained in section 6.4 as well.

Setting 6: Feature input version 1 long The same as setting 5, but again the models for trained for 200,000 steps like in setting 2.

6.3 Preprocessing

In section 2.3 I talked about the incompleteness and difficulties of transcribing using the IPA. How exactly a sound is mapped to an IPA symbol also depends on whoever transcribes a particular text. The WikiPron data has been put together by many different people. There are conventions on how to add transcriptions by Wiktionary, but there still might be inconsistencies. Other than that, it is always possible that something is correct, but neural models just cannot handle it well. That said I will carefully examine and clean the datasets. There is preprocessing that is done for both datasets, further down I will describe some preprocessing that had to be done for each individual dataset.

The first analysis is a character based comparison. This means that I used the python segments library³ to split up all the phonemes into unicode characters. Then

³<https://pypi.org/project/segments/>

I compared those characters to the PHOIBLE character set. This way I can identify characters that are a bit suspicious in the context of IPA as the PHOIBLE dataset is very extensive. While it is possible that a correctly used phoneme is not in PHOIBLE, it still gives a good overview of potential ambiguities in the transcription or even points out mistakes. This character based cleaning is done for both datasets. The more detailed list of what needs to be cleaned is found in table 10.

There are two more general things that I removed from the datasets. The first is **tones**. The reason for that is that it is not possible to infer tones from graphemes. Although they are used to distinguish meaning, they are not written down, but usually just known. This means that there are grapheme sequences that look exactly the same and their transcription is the same as well *except* for the tones. There are different ways to represent tones and I excluded all of them. The second additional preprocessing step I took is to remove all **punctuation symbols**. This accounts mostly for the NWS short stories. For a character-based modelling, punctuation does not add any valuable information as this only becomes relevant, and can only be inferred, on a word, respectively, sentence basis.

NWS corpus

Apart from cleaning the stories as described in table 6.3, I had to transform them into dictionary format in order to be able to use them as testing data. In order to do that it is necessary to tokenize both orthographic and phonetic texts and then align them. As the number of tokens (when split naively at blanks) is not always the same for orthographic and phonetic text, it was necessary to align some of them manually. While this is not a problem when talking about languages that I know, it is a bit tricky for languages completely unknown to me. Luckily there are tools online that provide a rough pronunciation of a word in a given language or even a phonetic transcription (although rarely in IPA)⁶.

Chinese and Thai required special tokenization. To tokenize Chinese and Thai I used *polyglot*⁷, a Python library.

Apart from the characters that I cleaned out in table 10, there were a few more characters that needed to be cleaned. The cleanings here were a classic example for the Unicode normalization issues that I presented in section 3.5. As the models

⁶For most languages I could use Google translator (<https://translate.google.com/?hl=de&sl=ko&tl=de&op=translate>), for languages like Hebrew, I could ask someone who knows the language to help me out.

⁷<https://polyglot.readthedocs.io/en/latest/>

Phon.	Unicode name	Repl.	Explanation
'	MODIFIER LETTER VERTICAL LINE	NULL	These are all IPA suprasegmentals except the long and half long marker and the extra short (: ˙ ˘). The reason why these were excluded is that they don't carry any meaning on the character level. The vertical lines, for example, mark intonation groups which only matter in a larger sentence or text context. There are a few rare occurrences of COMBINING VERTICAL LINE ABOVE which is probably meant to be MODIFIER LETTER VERTICAL LINE as they look similar. It is excluded as well.
,	MODIFIER LETTER LOW VERTICAL LINE	NULL	
	VERTICAL LINE	NULL	
	DOUBLE VERTICAL LINE	NULL	
.	FULL STOP	NULL	
˘	UNDERTIE	NULL	
ˆ	COMBINING DOUBLE INVERTED BREVE	NULL	Both tie bars below and above are excluded in PHOIBLE ⁴ which is why I am excluding it as well. Put plainly, those do not add any additional information that cannot be derived otherwise.
˜	COMBINING DOUBLE BREVE BELOW	NULL	
3˘	LATIN SMALL LETTER REVERSED OPEN E WITH HOOK	ə˘	Both base letters are very similar vowels. It is just more common to use the latter than the former.
ɡ	LATIN SMALL LETTER G	ɡ	The IPA 'ɡ' has a different code point and is a different character than the typical keyboard small Latin 'g'. This is just an IPA decision. For some fonts the two characters do not look different, for some they do.
˘	SWUNG DASH	NULL	All of characters make out less than 1% of their respective dataset, most of the time it is less than 0.1%. A close examination of the dataset and the Wiktionary transcription conventions for the respective language did not show any reason why to keep the phoneme. Note that the 'v' for the tilde is only there for correct representation.
,	COMMA	NULL	
˜	TILDE	NULL	
ə	MODIFIER LETTER SMALL SCHWA	NULL	
ˣ	MODIFIER LETTER SMALL X	?	The ˣ only occurred in the broad Finnish transcription and is used to denote possible gemination. In the narrow transcriptions there is a glottal stop instead. The occurrence of glottal stops and gemination follows the same rules. Therefore, for consistency, the gemination ˣ is mapped to a LATIN LETTER GLOTTAL STOP.
()	(SUPERSCRIPT) [LEFT RIGHT] PARENTHESIS	NULL	Parentheses are used to denote optionality for phonemes or tones. WikiPron actually discards those but keeps the content ⁵ . I will do the same for all parenthesis found.

Table 10: The table shows what phonemes were changed or excluded and what the reason is for this preprocessing. All characters that were excluded are replaced by a NULL value.

cannot handle characters that are not in their vocabulary, I needed to remove those to be able to evaluate the models on those stories. However, for some characters, they were in the vocabulary, but as precomposed characters while they were not precomposed in the stories. Those characters look exactly the same but the model (or any other text processing application) treats them as two completely different

characters.

WikiPron

As the CMU model does not expect the graphemes to contain any whitespace, I replaced any whitespace in the input with underscores. This was necessary only for Vietnamese. In a first run, I did not do this. When running the evaluation on the Vietnamese model, it did not work but showed an error message. As the model creates a vocabulary file, I had a look at it which revealed the following:

- The model does not actually split the input file at tab characters, but splits it at spaces and then uses the first item of the resulting list as input grapheme and the rest as a list containing the output phonemes. Consequently, if the input grapheme contains a white space, everything following it will be in the phoneme part. Part of the input will be treated as a phoneme character in the training which results in a huge vocabulary and wrong predictions.

The alternative to removing the white space would have been to split the grapheme at white space and align the phonemes. However, this is firstly much more work as it is not clear where to split the phonemes and it might be that the word on its own is pronounced differently.

In Japanese there is the superscript letter ^{β} . This actually means compression, which is a special type of rounding⁸. This is one of these things where the difficulties of transcribing texts becomes clear. This special case is not reflected in PHOIBLE nor in the official IPA but this does not mean it is wrong. As it is quite common in the Japanese data, I decided to keep it.

A last thing I did to clean the datasets was to exclude duplicate graphemes with different pronunciations. Although ambiguities and multiple possible pronunciations for one word are linguistically speaking very common, it is not possible for a neural model to distinguish such cases without any context. As G2P modelling happens on character basis and not on word basis there is no context available in our case that could account for at least some ambiguities. Also, in the WikiPron data that was preprocessed for the SIGMORPHON task, duplicates were excluded as well. It makes sense to clean all datasets in the same way.

⁸<https://en.wikipedia.org/wiki/Roundedness>

6.4 Feature encoding

In order to incorporate the phonetic features into the dataset, we decided to add what we refer to as ‘flags’ to the phonemes. Generally, the idea is to encode the phonetic features that PHOIBLE provides and add them to the WikiPron dataset. There are still a few phonemes that are not in the PHOIBLE set that are in my data. These will just not be encoded in the following experiments. An exception is the length marker. There are a few long vowels that are not like that in the PHOIBLE data. As there is the ‘long’ feature, it is easily possible to use the base character feature vector and mark the ‘long’ as ‘+’. This is the only thing I changed for the following experiments. All other features vectors are just used as is. I encoded the features in a jupyter notebook which is in my GitHub repository⁹.

I experimented with several things before I trained a first set of models. I will list the steps of my experiments in the following. For all feature experiments I used the cleaned WikiPron data.

Version 1

The first thing I tried was to add two flags to every phoneme. In order to do this, I encoded the features as two strings. This means that I split the features into two sets and then used a different set of random capital letters to encode all of these uniquely. After each phoneme I will add two flags that describe that phoneme. The combination of the two flags will be unique for each phoneme. The individual flags can overlap between the phonemes, if one half of the features is the same for that phoneme. The example below shows a possible encoding of a phonetic text:

(6.1) p AB CD k^h AA BC

(6.2) k^h AA BC ʊ BF CC

A problem with this approach is that the sequences get really long if all phonemes are represented as three strings. The CMU model cuts the sequence off at 30 characters. This means that only phoneme sequences that no longer than 10 characters will be represented in full. This is way too short to account for many words. It is possible to increase that limit. However, 30 characters is already quite long and there is no point in setting this limit too high as those models still experience difficulties in processing sequences that are too long. Not only the sequence length is influenced

⁹https://github.com/theDebbister/masterThesis/blob/main/data/MA_Encode_features.ipynb

but also the vocabulary size. Each flag will be added to the vocabulary.

It is therefore not surprising that the model did not perform well. In fact, the WER was about double the WER of the long model trained on the uncleaned data. Given this results, I decided not to train any more models and look for another way of how to encode the features.

Version 2

For the second attempt at encoding phonetic features I tried a different approach. Instead of adding two flags, I added only one flag after each phoneme. Also, this time the idea was not to add a unique flag to each phoneme but to encode only some features for all phonemes. This means that for some phonemes the same flag was added. As the phonemes were not replaced by the flags, but the phonemes were kept in the text, there is no information lost. The intuition behind it is to encode high-level patterns in phoneme sequences. For example, as vowels typically have some overlapping features, very similar vowels will be encoded using the same flag. In order to choose appropriate features for each language, I calculated the pearson correlation between the features. I only used those features whose correlation with all other features is not above a certain threshold. For this experiment I set the threshold to be 0.5. If the threshold was too high, for many languages, no correlating features were found. This is not surprising as features in general do not make a lot of sense if they are all telling the same thing. This is why I set the threshold relatively low which means I kicked out a lot of features. The reason for this is that all languages only use a relatively small subset of phonemes. This means that many features are not important for a language as those features are not used to distinguish meaning. In order to recognize high-level patterns, only the most distinctive features are important. Other than that I encoded the features in the same way as for version one. First results for this version reached similar scores as the baselines ones. I trained a full set of models for this version.

While encoding these features, I noticed that some phonemes are not listed as individual phonemes in PHOIBLE but are listed as allophones of one phoneme that is in there. This means that some broad transcriptions contain allophones which, ideally, should not be the case. Also, it means that I did not have features for these phonemes which means that those were not encoded.

In table 11, I list the features that were encoded for every language. Every possible combination of those features was assigned a different flag. For most languages, about half of the 37 features were encoded.

ISO 396-3	Type	Features
all langs	-	tone, stress, syllabic, short, consonantal, tap, trill, lateral, labial, epilaryngealSource, spreadGlottis, loweredLarynxImplosive
cmn	broad	long, continuant, nasal, constrictedGlottis, raisedLarynxEjective
deu	narrow	long, continuant, nasal, constrictedGlottis, raisedLarynxEjective
ell	broad	long, sonorant, continuant, delayedRelease, nasal, coronal, dorsal, constrictedGlottis, raisedLarynxEjective
eng uk	broad	long, continuant, nasal, constrictedGlottis, raisedLarynxEjective
eng uk	narrow	long, continuant, nasal, constrictedGlottis
eng us	broad	long, continuant, nasal, constrictedGlottis, raisedLarynxEjective
eng us	narrow	long, continuant, nasal, constrictedGlottis
eus	broad	long, continuant, delayedRelease, nasal, constrictedGlottis, raisedLarynxEjective
fin	broad	long, continuant, nasal, constrictedGlottis, raisedLarynxEjective
fin	narrow	long, continuant, delayedRelease, nasal, constrictedGlottis, raisedLarynxEjective
fra	broad	long, nasal, constrictedGlottis, raisedLarynxEjective
hin	broad	constrictedGlottis, raisedLarynxEjective
hin	narrow	nasal, constrictedGlottis, raisedLarynxEjective
ind	broad	long, continuant, nasal, constrictedGlottis, raisedLarynxEjective
ind	narrow	long, continuant, nasal, constrictedGlottis, raisedLarynxEjective
jpn	narrow	long, delayedRelease, coronal, dorsal, constrictedGlottis, raisedLarynxEjective
kat	broad	long, continuant, delayedRelease, nasal, coronal, constrictedGlottis
kor	narrow	continuant, nasal, constrictedGlottis, raisedLarynxEjective
mya	broad	long, delayedRelease, nasal, raisedLarynxEjective
rus	narrow	long, continuant, nasal, coronal, dorsal, constrictedGlottis, raisedLarynxEjective
spa ca	broad	long, sonorant, continuant, delayedRelease, nasal, dorsal, constrictedGlottis, raisedLarynxEjective
spa ca	narrow	long, sonorant, continuant, nasal, constrictedGlottis, raisedLarynxEjective
spa la	broad	long, sonorant, continuant, nasal, coronal, dorsal, constrictedGlottis, raisedLarynxEjective
spa la	narrow	long, sonorant, continuant, nasal, constrictedGlottis, raisedLarynxEjective
tgl	broad	long, continuant, delayedRelease, nasal, constrictedGlottis, raisedLarynxEjective
tgl	narrow	long, sonorant, continuant, nasal, dorsal, constrictedGlottis, raisedLarynxEjective
tha	broad	nasal, constrictedGlottis, raisedLarynxEjective
tur	broad	long, continuant, nasal, constrictedGlottis, raisedLarynxEjective
tur	narrow	long, nasal, constrictedGlottis, raisedLarynxEjective
vie	narrow	long, nasal, coronal, constrictedGlottis, raisedLarynxEjective
zul	broad	continuant, delayedRelease, nasal, coronal, constrictedGlottis, raisedLarynxEjective

Table 11: The table shows all features that were encoded for each language. Some of them are encoded for all languages, those are at the top of the table.

6.5 Results

For the evaluation of my models, I first used the test sets that were produced automatically by the model. In addition, I tested each model on the respective NWS story if available in that language. Below, I first discuss the results separately for the short (setting 1, setting 3 and setting 5) and the long (setting 2, setting 4 and setting 6) models. I will mostly compare the WER scores of the model, as this is common in research. Also, the PER scores are a bit harder to interpret, as the edit distance it involves is not very intuitively understandable as there are many different ways of how to get to the same result.

To evaluate the models that were trained on data containing feature flags, I cleaned out the features to calculate the scores on the predicted phonemes only. Leaving in the feature flags would make it difficult to compare it to other models' results.

6.5.1 Results short models

Table 12 shows the results for all models that I trained for 10,000 steps. It is interesting that the results for all three models are actually very similar. I was surprised by the fact that for a vast majority of the languages the version trained on the cleaned data did not improve the performance although the cleaning was supposed to make the data more consistent. For example, excluding duplicate words with different phoneme sequences should remove some ambiguities. However, it is also possible that it actually introduced more ambiguities as I randomly took one version of the duplicates which might be an uncommon one. An interesting result is the one for Zulu (zul). The cleaned model performed a lot better than the other two. As this result seems rather strange, I will have a close look at the data again. Manually comparing the test prediction for Zulu to the reference phonemes (there are only 167 words in the test set), reveals that the model almost exclusively got the tones wrong. Those were excluded in the cleaned version which might explain the wide gap. Finnish (fin) had a very similar result but the gap was not as large.

Table 13 shows a comparison between my short models and the SIGMORPHON models. As the comparison results from the shared task stay the same for both, my long and short models, and as the long models perform all better than these here, I will only compare the results further down for the long models.

Iso 639-3	Type WikiPron	WER BS	PER BS	WER BS-clean	PER BS-clean	WER F2	PER F2
cmn	broad	19.6	4.5	18.1	5.0	20.3	6.0
deu	broad	40.3	5.4	38.1	5.3	40.2	6.6
deu	narrow	52.1	7.5	59.9	9.4	59.9	10.4
ell	broad	9.8	0.9	10.4	1.0	11.5	1.1
eng us	broad	54.4	10.6	53.0	10.8	57.7	12.1
eng us	narrow	84.6	32.0	84.2	31.8	86.6	32.2
eng uk	broad	48.6	9.5	50.0	10.1	51.8	11.1
eng uk	narrow	88.8	32.2	93.8	34.5	90.5	31.2
eus	broad	19.4	2.8	22.8	2.8	23.6	4.0
fin	broad	5.8	0.4	11.3	1.3	11.5	3.1
fin	narrow	14.3	1.0	7.1	0.7	17.1	4.3
fra	broad	7.2	1.0	8.8	1.3	9.1	1.6
hin	narrow	8.4	1.4	10.1	1.6	8.3	1.3
hin	broad	5.6	1.2	7.3	1.2	6.9	1.2
ind	broad	35.3	5.3	36.3	4.8	38.6	5.7
ind	narrow	43.5	5.5	44.1	6.0	44.3	6.1
jpn	narrow	6.6	0.9	6.7	0.9	6.8	1.1
kat	broad	0.3	0.2	1.1	0.2	1.7	0.9
kor	narrow	28.7	4.6	27.0	4.4	26.7	4.9
mya	broad	34.2	7.4	34.1	7.1	87.5	18.8
rus	narrow	14.8	1.7	16.4	1.9	22.0	3.6
spa ca	broad	2.3	0.3	2.7	0.4	5.3	1.5
spa ca	narrow	3.6	0.5	9.1	1.5	8.4	1.6
spa la	broad	2.6	0.3	2.8	0.5	4.4	1.0
spa la	narrow	3.3	0.4	3.2	0.3	5.4	0.8
tgl	broad	33.3	5.9	33.4	5.1	34.8	5.6
tgl	narrow	42.9	6.6	51.1	7.6	50.3	7.2
tha	broad	14.1	2.9	13.0	2.8	15.3	3.7
tur	broad	52.8	8.0	53.4	7.9	54.3	8.6
tur	narrow	57.9	8.4	53.9	7.9	52.0	8.5
vie	narrow	3.0	1.3	3.0	1.5	4.3	2.0
zul	broad	61.7	11.5	10.4	1.1	95.1	13.7

Table 12: This table shows the results for setting 1 (BS), setting 3 (BS-clean) and setting 5 (F2). These are the short models.

ISO396-3	BS WER	SIG WER	Transcription type
eng (us)	53.00	37.43	broad
fra	7.20	5.11	broad
ell	9.80	18.67	broad
kat	0.30	0.00	broad
hin	5.60	5.11	broad
jpn	6.60	4.89	narrow
kor	26.70	16.20	narrow
vie	3.00	0.89	narrow

Table 13: The table shows the WER results for the best of my short models compared with the SIGMORPHON 2020 and 2021 results. For each language the best score is reported no matter what year or what model from the SIGMORPHON task.

6.5.2 Results long models

Table 14 shows the results for all models that I trained for 200,000 steps. The results and the differences between the models are very similar to the short models. The observation for Zulu (zul) is still the same. In fact, the model on the uncleaned data performed even worse than its short equivalent.

The largest improvement from the short to the long models was for Russian (rus). Russian had by far the largest dataset with more than 400,000 samples. It makes sense that for this large dataset, the model takes more time to extract all the relevant information.

Table 15 shows a comparison of the best long model compared with the best SIGMORPHON shared task result. For three of those eight languages, my model reached a better or the same score. The SIGMORPHON Korean model performed quite a lot better than mine which can be explained by the fact, that this model was not trained on the Korean logograms but those were converted to hangul characters which is a Latin alphabet representation of Korean. As explained in section 3.2, this can lead to better results as it reduces the vocabulary size. My result for English is a lot worse as well. Generally, English (as well as German) seems to be a language that is relatively difficult to pronounce as the results are often a lot worse than those of other languages. In the present case, the different might result from the fact that the SIGMORPHON model was an ensemble and it was only trained for English and therefore optimized to work for the English language. My model performs better on

Iso 639-3	Type WikiPron	WER BS	PER BS	WER BS-clean	PER BS-clean	WER F2	PER F2
cmn	broad	17.6	3.9	17.0	4.0	18.1	4.4
deu	broad	37.1	4.8	38.1	5.0	38.6	5.9
deu	narrow	52.2	7.1	56.9	8.4	55.9	9.4
ell	broad	7.1	0.6	9.0	0.7	9.1	0.8
eng us	broad	50.7	9.5	51.2	10.2	51.3	10.4
eng us	narrow	84.6	31.4	84.2	33.1	84.9	32.3
eng uk	broad	45.5	8.6	47.5	9.2	47.6	9.7
eng uk	narrow	90.3	30.2	94.8	35.3	93.7	34.2
eus	broad	21.2	2.7	19.6	2.3	21.7	3.2
fin	broad	2.8	0.2	3.3	0.3	8.9	3.1
fin	narrow	3.2	0.3	3.9	0.4	9.0	3.5
fra	broad	5.3	0.7	5.8	0.8	5.9	0.8
hin	narrow	7.7	1.2	8.4	1.4	8.4	1.3
hin	broad	4.4	0.7	6.4	1.0	6.3	1.0
ind	broad	37.9	5.3	34.9	5.3	39.3	6.1
ind	narrow	43.1	5.4	43.0	5.6	43.1	5.6
jpn	narrow	6.5	0.6	6.8	0.6	6.6	0.8
kat	broad	0.0	0.0	0.0	0.0	1.0	0.8
kor	narrow	23.4	4.1	25.3	4.4	25.8	4.6
mya	broad	35.1	6.5	36.0	6.9	88.0	17.4
rus	narrow	1.9	0.2	2.4	0.3	5.0	1.5
spa ca	broad	1.1	0.1	1.3	0.1	2.2	0.7
spa ca	narrow	2.3	0.3	2.2	0.3	2.8	0.6
spa la	broad	1.4	0.1	1.5	0.2	1.9	0.6
spa la	narrow	2.6	0.3	2.7	0.3	2.7	0.4
tgl	broad	28.4	4.6	31.2	5.2	33.9	5.0
tgl	narrow	45.5	6.4	47.3	7.0	48.6	6.9
tha	broad	12.5	2.6	11.1	2.5	12.1	3.1
tur	broad	50.6	7.8	52.3	7.5	49.1	7.2
tur	narrow	55.1	7.6	55.6	8.3	54.8	8.0
vie	narrow	1.5	0.8	1.6	0.8	2.6	1.5
zul	broad	65.9	10.7	9.8	0.9	91.4	12.0

Table 14: This table shows the results for setting 2 (BS), setting 4 (BS-clean) and setting 6 (F2). These are the long models

Modern Greek. This is an expected result, as the model from the SIGMORPHON task was trained on only 800 samples. I had more than 10,000 samples. This aligns with cutting edge research, as 800 samples have been found not to be enough to train a decent model. These results confirm this finding as my model was *considerably* better.

ISO396-3	BS WER	SIG WER	Transcription type
eng (us)	50.70	37.43	broad
fra	5.30	5.11	broad
ell	7.10	18.67	broad
kat	0.00	0.00	broad
hin	4.40	5.11	broad
jpn	6.50	4.89	narrow
kor	23.40	16.20	narrow
vie	1.50	0.89	narrow

Table 15: The table shows the WER results for the best of my long models compared with the SIGMORPHON 2020 and 2021 results. For each language the best score is reported no matter what year or what model from the SIGMORPHON task.

6.5.3 Results NWS tests

Table 16 presents the results for the tests on the NWS stories. The models do not perform very well on these stories in general. Concerning the model types, there is no real pattern in how they perform. The clean models seems to perform a bit better than the other two types. Most of them reach the best performance at least once. Also, as the texts are really short, the performance is often very similar.

Still, when compared to the coverage experiment in section 4.5, many G2P models perform quite a lot better. Although this is not surprising, it is good that we can confirm that those models are able to extract a lot of information. What is surprising is that there are a few languages like broad Hindi (hin) or Vietnamese (vie) where the WER of the G2P models is higher or very similar to the coverage experiment. Neither of these models performed particularly bad when tested on the WikiPron test set (WER hin: 4.4, WER vie: 1.5), in fact the results were very good. In addition, both training sets contain more than 10,000 samples. The most plausible explanation is probably that the transcription conventions are just very different.

ISO 396-3	Type	Long		Short		Long clean		Short clean		F2 long		F2 short	
		WER	PER	WER	PER	WER	PER	WER	PER	WER	PER	WER	PER
cmn	broad	95.0	72.5	100	73.8	94.1	45.2	86.1	40.5	93.1	39.2	84.2	36.4
deu	broad	74.1	26.0	76.9	28.0	72.2	24.5	63.0	23.4	72.0	20.8	65.4	19.8
deu	narrow	50.9	16.2	54.6	16.4	63.9	19.6	63.9	20	57.0	16.6	59.8	17.9
ell	broad	78.9	32.8	75.4	31.1	75.4	29.2	75.4	29.5	77.0	17.3	77.0	18.3
eng us	broad	66.4	22.3	87.6	32.5	84.1	27	87.6	30.4	83.9	20.9	92.0	24.7
eng us	narrow	94.7	47.5	97.3	44.5	97.3	51.8	97.3	47.9	97.3	45.1	97.3	44.1
eus	broad	36.8	4.2	48.3	5.9	36.8	4.8	34.5	4.7	34.9	5.1	34.9	4.8
fra	broad	37.0	15.6	45.4	17.7	34.3	14.3	48.1	19.3	34.6	11.6	48.6	15.5
hin	narrow	91.1	62.6	91.1	62.6	90.3	58.5	90.3	58.5	90.2	29.9	90.2	30.4
hin	broad	96.0	56.1	96.0	55.8	95.2	57.8	95.2	57.8	95.1	47.0	95.1	47.0
ind	broad	81.5	30.8	81.5	27.9	81.5	29.7	81.5	30.1	82.2	17.8	82.2	19.4
ind	narrow	85.2	33.0	82.4	32.2	82.4	31.3	82.4	31.0	86.0	23.8	86.0	23.6
kat	broad	64.2	14.6	64.2	14.6	58.2	11.5	58.2	11.5	59.1	7.5	59.1	7.8
kor	narrow	100.0	52.5	100.0	52.7	100.0	52.5	100.0	51.4	100.0	100.0	100.0	100.0
mya	broad	97.5	34.4	97.5	36.5	95.0	21.3	95.0	23.2	97.4	29.4	97.4	35.8
rus	narrow	91.6	37.2	91.6	37.3	89.5	36.3	91.6	37.6	91.5	35.7	91.5	37.1
spa ca	broad	27.8	5.1	28.9	5.4	27.8	5.1	27.8	5.1	32.3	5.4	32.3	5.4
spa ca	narrow	56.7	17.3	53.6	16.5	58.8	16.2	54.6	16.2	65.6	17.3	54.2	16.1
tha	broad	99.4	62.3	99.4	62.2	56.9	19.0	56.9	19.1	57.2	16.3	56.6	16.4
tur	broad	53.8	7.8	63.1	10.1	46.2	6.7	55.4	8.7	56.2	9.1	64.1	9.2
tur	narrow	76.9	15.6	66.2	11.6	76.9	11.9	75.4	13.7	75.0	11.9	68.8	12.1
vie	narrow	100.0	99.8	100.0	99.8	79.5	44.2	79.5	44.2	80.2	36.1	80.2	36.1

Table 16: In this table I present the results for all models when they predicted the NWS stories.

6.5.4 Results overall

All three model types show a similar performance no matter how long they were trained. The most notable observation is that the difference between the performance of the long and the short models is rather small. For some language, the best short model performed even better than the long one or the results were the same. This was the case for narrow German (deu), narrow US English (eng us), narrow UK English (eng uk), broad Goizueta Basque (eus), broad Burmese (mya), narrow Tagalog (tgl) and narrow Turkish (tur). I find it surprising that most of these are the narrow variants as those are more detailed. I would expect a model trained

for longer to catch more details. However, none of the scores were really low, the lowest being 19.4 for Basque. Additionally, the differences are never larger than 2.8 percentage points.

The feature models did not show any relevant improvement compared to the models trained without features. They did not worsen the results but performed very similarly. Overall they showed for almost all languages quite a large improvement when trained longer. This could mean that they would get better still when trained longer, but the relation between the three model types did not change with longer training time, so I suppose it is rather unlikely that they outperform the other model types.

7 Conclusion

In the beginning I listed a few questions that I would like to answer while working on this thesis. While I cannot answer all of them in full, there are a few interesting insights.

1. What types of phonetic data is available and how can it be used?
2. Which computational models can be used to create phonetic transcriptions?
3. How can we use phonetic features to create phonetic transcriptions?
4. Is there any significant difference in comparing spoken or written languages?
5. Does written text represent language well enough to justify text-based research only?

Concerning question one, the overwhelming majority of available data is word lists. But in general, there is quite a lot of data that I could use for my experiments. The WikiPron data keeps growing and even if it is not preprocessed a lot the models are able to produce good results. As G2P conversion is a well-known seq2seq task, there exist a lot of models and architectures that can be used. The experiments with the feature models showed that it is possible to manipulate the input and the model do still perform well. The last two questions are still open. However, one thing I did notice is that phonetic text (given it is written in IPA) is more complex and has more characters than most alphabets. The case for Zulu also showed that the tones (which are represented as diacritics in Zulu) add to that complexity. The comparison of the NWS transcriptions and the WikiPron test sets showed that phonetic transcriptions are not as standardized as typical writing systems.

All in all, there were some good results produced for a large collection of many languages. Some outperformed existing state-of-the-art models, for some there was no comparison available.

7.1 Future work

Non surprisingly, apart from many exciting things I *could* do, there are many others that would have gone beyond the scope of this thesis. I would like to list a few entry points on where further research could start.

- **Data preprocessing:** Ashby et al. [2021] cleaned broad transcriptions for Bulgarian and replaced allophones by their standard phoneme. This could further improve model quality by having consistent broad transcriptions.
- **Phonetic features:** Although my models did not outperform the other models on many language, they did not perform a lot worse. In this thesis, I tried to add the features by manipulating the input to the model. It might also be possible to chose a different model architecture such that the features can be passed more explicitly to the model. I am sure that it is worth digging deeper into phonetic features and how we can use them.
- **Low resource languages:** There are a lot of languages where only very little data is available. I did not deal with those in this thesis but it is technically possible to use transfer learning or other techniques to make use of those languages. Low resource languages still are a challenge.

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A Tables

Iso 639-3	Name-WALS	Family-WALS
abk	Abkhaz	Northwest Caucasian
amp	Alamblak	Sepik
aeu	Amele	Trans-New Guinea
apu	Apurinã	Arawakan
bmi	Bagirmi	Central Sudanic
bsn	Barasano	Tucanoan
gry	Grebo	Niger-Congo
eus	Basque	Basque
ape	Arapesh (Mountain)	Torricelli
bsk	Burushaski	Burushaski
ram	Canela-Krahô	Macro-Ge
tzm	Berber (Middle Atlas)	Afro-Asiatic
cha	Chamorro	Austronesian
ckt	Chukchi	Chukotko-Kamchatkan
zoc	Zoque (Copainalá)	Mixe-Zoque
dgz	Daga	Dagan
hae	Oromo (Harar)	Afro-Asiatic
arz	Arabic (Egyptian)	Afro-Asiatic
fij	Fijian	Austronesian
fin	Finnish	Uralic
gni	Gooniyandi	Bunuban
khk	Khalkha	Altaic
hau	Hausa	Afro-Asiatic
heb	Hebrew (Modern)	Afro-Asiatic
hin	Hindi	Indo-European
hix	Hixkaryana	Cariban
hnj	Hmong Njua	Hmong-Mien
qvi	Quechua (Imbabura)	Quechuan
imn	Imonda	Border
ind	Indonesian	Austronesian
kal	Greenlandic (West)	Eskimo-Aleut
kyh	Karok	Karok
gyd	Kayardild	Tangkic
kio	Kiowa	Kiowa-Tanoan
cku	Koasati	Muskogean
kor	Korean	Korean
ses	Koyraboro Senni	Songhay
kgo	Krongo	Kadu
kut	Kutenai	Kutenai
lkt	Lakhota	Siouan
laj	Lango	Eastern Sudanic
lvk	Lavukaleve	Solomons East Papuan
lez	Lezgian	Nakh-Daghestanian
dni	Dani (Lower Grand Valley)	Trans-New Guinea