

Autophon user guide

Danish

Engine: Montreal Forced Aligner 1.0

Model: DanFA 4.0

1 Introducing Autophon and Forced Alignment

Autophon is a free, user-friendly tool for phoneticians that performs *forced alignment* (FA) – the automated process of converting speech recordings and their transcriptions into phonetically time-stamped annotations.

Autophon leverages widely used alignment engines developed by the phonetics community, including:

- FAVE¹
- faseAlign²
- Montreal Forced Aligner, version 1.0³
- Montreal Forced Aligner, version 2.0³

The tool produces time-aligned phonetic annotations compatible with Praat⁴, based on two user inputs: (1) a speech recording and (2) its orthographic transcript.

This user guide is specifically for **Danish**, using the **Montreal Forced Aligner 1.0** engine with the **DanFA 4.0** model. Autophon may support additional engine-model combinations for this language; therefore, ensure you are using the best option for your needs.

While many forced aligners exist, they often require command-line usage and are tied to outdated or incompatible operating systems. **Autophon offers a platform-independent, intuitive alternative for phoneticians worldwide.**

2 Using the app

2.1 Aligning files without registering To align smaller files, go to the main page and click **Add files** at the bottom. A box titled *Transcription Mode: change transcription mode* will appear. Click the heading to choose one of four *Transcription Modes* (see below), then select your files.

2.2 Registering and logging in To align larger files or access the full suite, click **Sign up** to create a free account. This helps us monitor usage for funders and guard against bots. After signing up, check your email for a verification link. If it doesn't arrive, check your spambox and wait 15 minutes before contacting tech support.

2.3 Cost Autophon is free of charge.

2.4 Aligning files in a registered account Once registered and verified, go to the **Aligner** tab and click **Add files**. A box titled *Transcription Mode: change transcription mode* will appear. Click the heading to choose one of four *Transcription Modes*, then select your files.

2.5 Transcription modes Autophon supports four *Transcription Modes*, named for the fields they're commonly used in: *Experimental Linguistics A*, *Experimental Linguistics B*, *Computational Linguistics*, and *Variationist Linguistics*. Each mode can be selected via the corresponding box in Figure 1, which illustrates expected file structures and links to instructional videos.

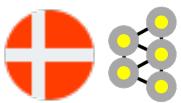
Video instructions for each transcription mode can be viewed. In addition, sample templates for each mode are available for [download here](#).

¹FAVE was built by Rosenfelder, Fruehwald, Brickhouse, Evanini, Seyfarth, Gorman, Prichard, and Yuan (2022). It relies on the Hidden Markov Toolkit (S. J. Young, Woodland, and Byrne 1993).

²faseAlign was built by Wilbanks (2022). Like FAVE, it relies on the Hidden Markov Toolkit (S. J. Young, Woodland, and Byrne 1993).

³The Montreal Forced Aligner was developed by McAuliffe, Socolof, Mihuc, Wagner, and Sonderegger (2017). It uses the Kaldi toolkit (Povey, Ghoshal, Boulianne, Burget, Glembek, Goel, Hannemann, Motlicek, Qian, Schwarz, et al. 2011).

⁴Praat is a speech analysis tool developed by Boersma and Weenink (2017).

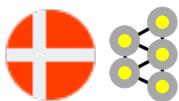


Experimental Ling A (click to see video guide)	Experimental Ling B (click to see video guide)	Computational Ling (click to see video guide)	Variationist Ling (click to see video guide)
<pre>yourzip.zip - yourtrans.xlsx/tsv/txt - file0001.wav - file0002.wav - file0003.wav ... - file9999.wav</pre> <p><i>Transcriptions in a master file absent of time stamps - as separate rows with separate audio* files for each transcription.</i></p>	<pre>yourzip.zip - yourtrans.xlsx/tsv/txt - file01.wav - file02.wav - file03.wav ... - file99.wav</pre> <p><i>Transcriptions in a master file with start and end time stamps with more than one row per audio* file.</i></p>	<pre>yourzip.zip - file0001.lab - file0001.wav - file0002.lab - file0002.wav - file0003.lab - file0003.wav ... - file9999.lab - file9999.wav</pre> <p><i>Transcriptions as separate same-name lab and audio* files, absent of time stamps.</i></p>	<pre>yourzip.zip - file01.TextGrid - file01.wav - file02.eaf - file02.wav - file03.tsv - file03.wav - file04.xlsx - file04.wav ... - file99.txt - file99.wav</pre> <p><i>Longer transcription files in TextGrid, eaf, tsv, txt, or xlsx format with same-name audio* files.</i></p>

Figure 1: The Transcription Mode selection menu for Autophon.

The figure shows two file trees representing Autophon outputs. The left tree shows a directory structure for 'daDK_small' containing subfolders 'X0297' and 'X0298'. Each folder contains several wav and lab files. The right tree shows the same structure but includes additional TextGrid files (e.g., X0297-dk15-09082000-1715_u0295140-1.TextGrid) under the respective subfolders.

Figure 2: Autophon outputs finished TextGrids using the same subfolder structure as the uploaded files.



Experimental linguistics A: Upload a two-column spreadsheet (Excel **xlsx**, or tab-delimited **txt/tsv**) with audio filenames in column 1 and transcriptions in column 2. No time stamps allowed. This format suits short clips and resembles CommonVoice⁵. Use zip or individual file upload.

Experimental linguistics B: Same structure as A, but with four columns: audio file name, start time, end time, and transcription. Designed for longer recordings requiring segmentation. Time stamps must be in real-number seconds (e.g., **1.23** or **1,23**); no colons or hour-minute markers are permitted (e.g., you may not use something like **00:00:01.23**).

Computational Linguistics: Upload matching audio and **lab** files (containing only the corresponding transcription). Files may be zipped with nested folders—Autophon preserves the hierarchy (Figure 2). No time stamps permitted.

Variationist Linguistics:⁶ Upload paired transcription and audio files (individually or zipped). Transcriptions may be in Praat **TextGrid**, ELAN **eaf**, or tabular format (**xlsx**, **txt**, **tsv**). Use either three or four columns:

- Four-column: speaker, start time, end time, transcription
- Three-column: start time, end time, transcription

Time stamps must be real-number seconds (comma or period decimal separators); formats with colons (e.g., **00:00:01.23**) are not supported.

2.6 File formats and codecs

If you encounter errors during upload, it's often due to unsupported file formats or codecs. The simplest fix is to re-save your files in a common format using tools like Praat or ELAN.

Transcription file formats: Autophon accepts transcription files in most standard encodings, including **UTF-8** and **UTF-16** (**Windows CRLF**). If you encounter issues, try re-saving the file or email a sample to tech support.

Audio file formats: Autophon supports a wide range of audio formats, including: **WAV**, **FLAC**, **MP3**, and more. Stereo files are not currently accepted. Therefore, convert all audio to mono first.

2.7 Transcription preparation

Regardless of the transcription mode, each entry should contain between one and 20 words. Boundary demarcations must include at least 0.01 seconds of silence before and after the speech. Figure 3 shows a five-word phrase with a 0.03-second pre-boundary and a 0.25-second post-boundary. This sort of variability is expected and handled well by Autophon.⁷

2.8 Select a language

After uploading files into the aligner, Autophon will suggest a language and language model. You may override this suggestion using the dropdown menu.

2.9 Task list

The task list displays all uploads along with file name, upload date, language, tier count, file size, word count, and an inventory of missing words. You can delete the task and start over, add words to your custom pronunciations box (described below), or proceed by clicking **Align**.

2.10 Missing words

To understand this feature, it helps to know that forced alignment maps phonemic pronunciations – defined in language-specific dictionaries – onto the speech stream using statistical models. These dictionaries contain a finite set of words. The missing words feature lists items not found in Autophon's dictionary and provides suggested pronunciations. Autophon will use these suggestions by default, but you can reject them and enter your own. The next section explains how.

2.11 Your custom pronunciations

If you disagree with either (a) Autophon's pronunciation suggestions for missing words or (b) the default dictionary entries, you can override them here. Enter your own phonemic transcriptions in this box, which will take precedence over both.

Pronunciations must be entered using the alphanumeric string specific to the language model at hand – in this case, the **DanFAbet**. Section 4 provides a key that maps the DanFAbet to its IPA⁸ equivalents.

⁵<https://commonvoice.mozilla.org>

⁶This field originally drove the development of forced alignment in the early 2000s.

⁷If your transcriptions are segmented with exact start and end times, performance may degrade and boundary shifts may occur. If you're working with such data, contact tech support—we are interested in designing a fifth transcription mode for these cases.

⁸International Phonetic Alphabet

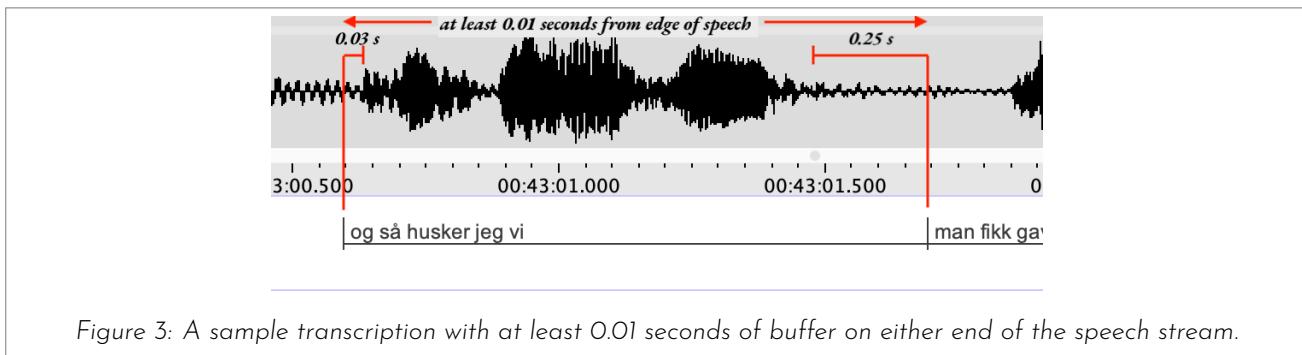
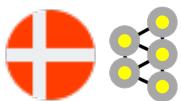


Figure 3: A sample transcription with at least 0.01 seconds of buffer on either end of the speech stream.

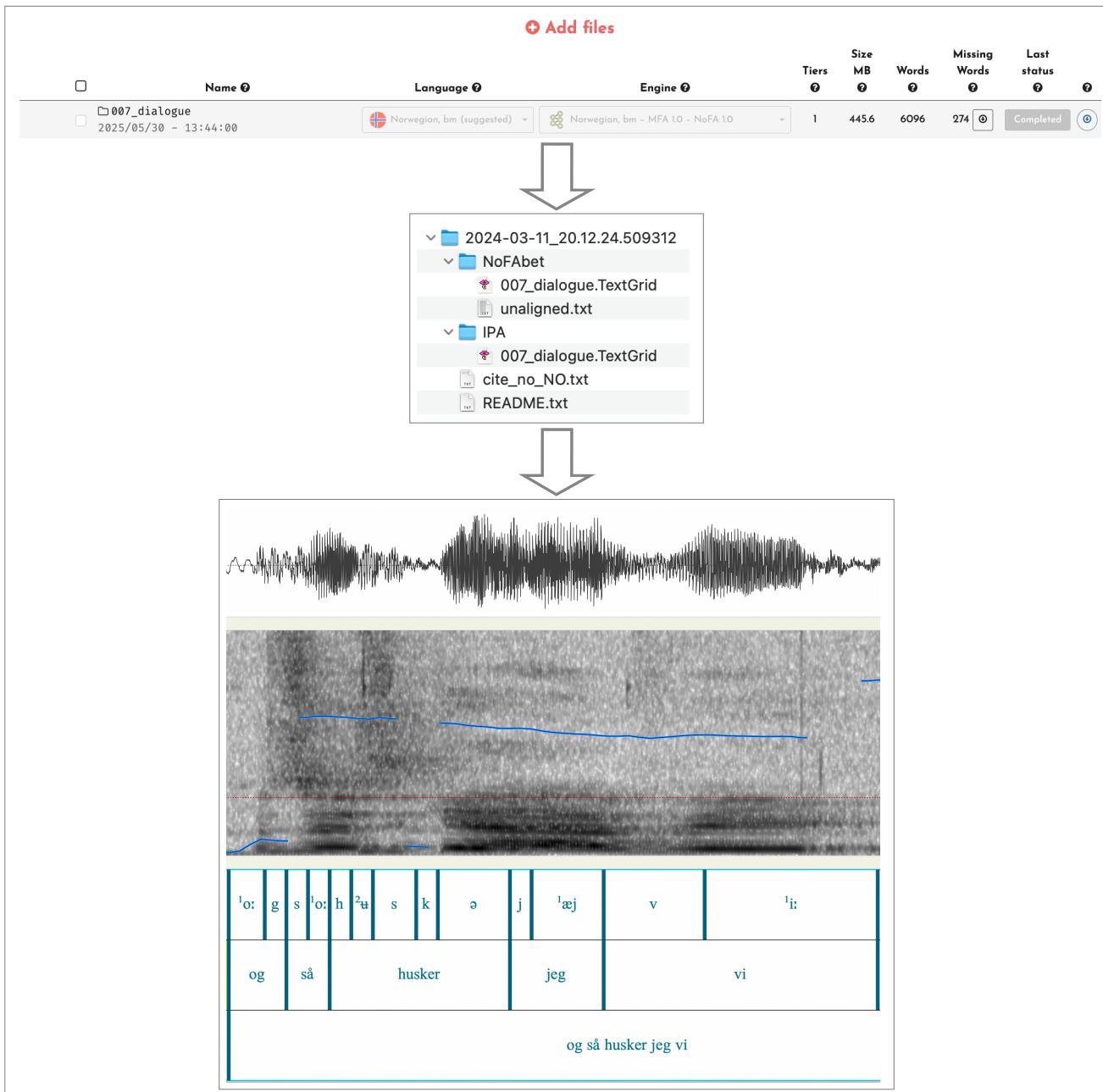
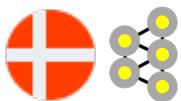


Figure 4: The alignment process, including task list, folder structure, and Praat TextGrid.



You can enter pronunciations directly in the dictionary box or upload them from a **txt** file. The maximum input length is 50 000 characters.

Entries must follow the format:

- word[space]phoneme[space]phoneme OR word[tab]phoneme[space]phoneme

Each phoneme must be separated by a space, and the lookup may not include two or more words – Autophon will interpret the second word as a phone and produce an error. You may submit multiple pronunciations for the same word by repeating the word on separate lines with different phoneme strings. Autophon will evaluate the best match for each speech instance. Refer to Figure 5 and the examples below.

Stress and/or accent **must** accompany every vowel and diphthong with a number. Consult Figure 1 for the specific digits used in this model; refer to Figure 5 to see how these are operationalized.

The figure shows a comparison between two parts of the Autophon interface. On the left, there is a decorative illustration of an open book with sound waves coming out of it, followed by a large arrow pointing to the right. The right side shows a screenshot of a web-based application. At the top, it says "Your Custom Pronunciations" with a "Swedish - MFA 1.0 - SweFA 2.0" dropdown. Below that is a text input field with the placeholder "Type directly into the field below or upload a text file here". Underneath the input field, there is a table of phoneme strings:

:	ackompanjera	AH0 K OAH0 M P AH0 N J EE1 R AH0
:	ackompanjerade	AH0 K OAH0 M P AH0 N J EE1 R AH0 D EH0
:	ackompanjerades	AH0 K OAH0 M P AH0 N J EE1 R AH0 D EH0 S
:	ackompanjeras	AH0 K OAH0 M P AH0 N J EE1 R AH0 D S

Below this table, there are two sections: "Correct:" and "Incorrect:". The "Correct:" section lists three examples of well-formed phoneme strings. The "Incorrect:" section lists three examples of errors: missing vowel stress numbers, missing spaces between phones, and a single line containing two look-ups.

Figure 5: Interface with dictionary entry (left) and phoneme string input (right).

2.12 Aligning files To begin alignment, click *Align* to the far right of the upload list. Alignment typically takes a few minutes, depending on server load.

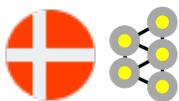
2.13 Downloading the annotations When alignment is complete, you can download the annotations as Praat TextGrids by clicking the downward arrow beside the task list. See Figure 4 for an illustration.

3 How to cite

Any dissemination or publication that makes use of this Autophon package for **Danish**, which uses **DanFA 4.0** within **The Montreal Forced Aligner 1.0** for its engine, should cite the relevant references listed below. Proper citation is essential: not only to acknowledge the “daisy chain” of technical and academic work underpinning Autophon, but also to reinforce the incentives for sharing tools with the broader community.

While space constraints may tempt you to remove references to software, we strongly encourage prioritizing these citations. If trimming is necessary, consider reducing peripheral citations in the literature review instead.

- Boersma, P., & Weenink, D. (2017). Praat: Doing phonetics by computer [softw.], ver.6.0.36. www.praat.org
- McAuliffe, M., Socolof, M., Mihuc, S., Wagner, M., & Sonderegger, M. (2017). Montreal Forced Aligner: Trainable text-speech alignment using Kaldi. *Proceedings of Interspeech*, 498–502.
- Young, N. J., & Anikwe, K. H. (2025). Autophon – Automatic phonetic annotation and online forced aligner. www.autophon.org
- Young, N. J., Falkencrone, B., & Pharao, N. (2024). DanFA 4.0 – Forced Alignment of Danish, ver. 4.0. www.autophon.org



4 Phoneme key

Autophon will output two versions of the same TextGrid for every file you align: (1) a TextGrid in the DanFAbet specific to DanFA 4.0 for The Montreal Forced Aligner 1.0 and (2) a TextGrid in the International Phonetic Alphabet (IPA). The phoneme key is located in Table 1.

DanFA	IPA	example	DanFA	IPA	example	DanFA	IPA	example	DanFA	IPA	example																						
Full vowels																																	
AA	a:	far	OEH	ø	synder	MQ	m̩	dame → dam̩	R	ɛ̄	rose																						
AE	e:	kæde	OH	o	foto	NQ	ň	håne → hän̩	S	s̄	si																						
AEE	æ:	mast	OO	o:	ro	NGQ	ŋ̄	konge → koŋ̄	SJ	ç̄	sjæl																						
AEEH	æ	mat	UH	u	guld	Consonants																											
AEH	ɛ	mæt	UU	u:	hus	B	b	bil	V	v̄	vase																						
AH	ɑ	takke	YH	y	nyt	D	d̄	dag	Z	z̄	Benz̄*																						
AO	ɔ:	morse	YY	y:	nyt̄	DX	ð̄	bad̄	ZH	ʒ̄	juh̄*																						
AOH	ɔ	vor	AXH	ʌ	råt̄	F	f̄	fin̄	Lexical stress and glottal stød																								
EE	e:	vene		G	ḡ	gul̄	os1	ɔ̄?	barn																								
EH	e	midt	Diphthongs																														
IH	i	mit̄	-J	-j̄	nej̄	J	j̄	jā	o1	ɔ̄:	barnevogn																						
II	i:	vin̄	-R	-ʁ̄	jer̄	K	k̄	kost̄	os2	ɔ̄?	spædbarn̄																						
OA	ɔ:	rå	-W	-ɥ̄	hav̄	L	l̄	land̄	o2	ɔ̄:	spildvarme																						
OAH	ɔ	rust̄	Schwa assimilation																														
OE	ø:	føre	DXQ	ð̄	bade → bað̄	NG	ŋ̄	enḡ																									
OEE	œ:	fyrre	LQ	!	tale → tal̄	P	p̄	pil̄																									

* multiethnolect

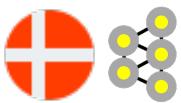
Table 1: DanFAbet, IPA, and lexical examples. The prosodic denotation means that any DanFAbet vowel or diphthong must **always** be followed by the numbers **1** (primary stress), **2** (secondary stress), or **0** (unstressed). Note also that **s** before stress denotes **stød**.

Numerical stress and glottal stød Every DanFAbet vowel is followed by a numerical code that denotes lexical stress. A lower-case **-s** is included between the upper-case base form and the numerical stress to denote the inclusion of lexical glottal stød. For example, the character string **ø0** refers to lexically unstressed vowels; **o1** – primary lexical stress (IPA: **ɔ̄:**); **os1** – primary lexical stress with stød (IPA: **ɔ̄?**); **o2** – secondary lexical stress (IPA: **ɔ̄:**); **os2** – secondary lexical stress with stød (IPA: **ɔ̄?**).

Diphthongs Every DanFAbet vowel can be combined with **J**, **R**, or **W** to assemble diphthongs with the offglides /-j̄/, /-ʁ̄/, and /-ɥ̄/, respectively. These must always be combined with numerical stress **0**, **1**, or **2**. To include glottal stød, we add a lower-case **-s** between the upper-case base form and the numerical stress. For example, the /'yɣ̄/ in **fyrste** combines **YH** + rhotic offglide **R** + primary stress **1** to render **YHR1**. Similarly, the /'yɣ̄?/ in **fyr** combines **YH** + rhotic offglide **R** + lexical glottal stød **s** + primary stress **1** to render **YHRS1**.

5 Acoustic model and pronunciation dictionary

This specific Autophon package for **Danish** uses DanFA 4.0 within The Montreal Forced Aligner 1.0, which was trained on spontaneous and read-aloud Copenhagen Danish from the DanPASS corpus⁹, read-aloud speech from the NST Danish ASR database¹⁰, and read-aloud speech from the Nordic Dialect Corpus¹¹. The pronunciation dictionary is an adapted version of the NST Pronunciation Lexicon for Danish¹², within which a series of changes were made, including schwa assimilation.



		DanFA 1.0				DanFA 2.0				DanFA 3.0/4.0 (current)			
		<i>n</i> boundaries		Median onset difference (ms)		<i>n</i> boundaries		Median onset difference (ms)		<i>n</i> boundaries		Median onset difference (ms)	
Received Copenhagen	Speaker 1	1128	14	34	62	1163	12	44	67	1163	10	47	72
	Speaker 2	1031	16	32	57	1265	10	49	70	1265	10	51	77
	Speaker 3	1062	17	34	55	1474	9	53	73	1474	10	50	76
Working-class Copenhagen	Speaker 1	639	17	30	55	638	15	36	59	638	11	44	69
	Speaker 2	804	13	37	63	806	14	41	63	806	10	45	73
	Speaker 3	740	18	30	53	750	11	47	66	750	10	46	75
Copenhagen Multiethnolect	Speaker 1	845	19	28	51	872	2	59	76	872	10	46	68
	Speaker 2	743	17	32	55	726	18	35	54	726	13	41	61
	Speaker 3	1011	17	32	54	1063	10	48	66	1063	10	43	69
<i>all</i>		8003	16	33	57	8757	11	47	67	8757	10	47	72

Key

<i>n</i> boundaries	number of boundaries tested against the manual gold standard (g.s.)
median onset difference (ms)	median difference between aligner boundaries and manual g.s. boundaries
pct 10 ms or less	percentage of aligner boundaries that fall within 10 milliseconds of manual g.s. boundaries
pct 20 ms or less	percentage of aligner boundaries that fall within 20 milliseconds of manual g.s. boundaries

Table 2: Accuracy metrics for DanFA version 1.0 (N. J. Young 2017), DanFA version 2.0 (N. J. Young 2021), DanFA version 3.0 (N. J. Young 2023), and the current DanFA 4.0 (N. J. Young, Falkencrone, and Pharao 2024)

6 Performance metrics

DanFA 4.0 is quite accurate and exceeds most of the benchmarks established in the forced-alignment literature. Its accuracy is measured here by comparing alignments of approximately 1000 phonemes in spontaneous speech, each from nine male teenage speakers from the Copenhagen region. The alignments are compared with manual segmentation in Table 2.

7 Data security and GDPR compliance

Files uploaded to Autophon are encrypted and transmitted to a secure server hosted by Digital Ocean within the European Union (Frankfurt and Amsterdam). Transcriptions and audio files are automatically deleted immediately after alignment. This approach enhances privacy while also reducing storage costs. By contrast, finished TextGrids remain available in your account until you choose to delete them. Once deleted, they are permanently removed from our servers.

If you upload files but do not initiate alignment by clicking *Align*, the files will be automatically purged at 3:00 AM GMT¹³.

Autophon adheres to the principles of the European Union’s General Data Protection Regulation (GDPR). We collect only four pieces of user information: name, title, affiliation, and email address. Once a file is aligned, the corresponding audio is permanently deleted. Deleting a file from your task list also permanently removes the transcription and filename metadata. You may delete your account at any time, which will erase all associated

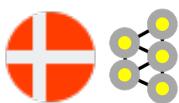
⁹Grønnum (2006)

¹⁰<https://www.nb.no/sprakbanken/en/resource-catalogue/oai-nb-no-sbr-55/>

¹¹Johannessen, Priestley, Hagen, Nøklestad, and Lynum (2012)

¹²<https://www.nb.no/sprakbanken/en/resource-catalogue/oai-nb-no-sbr-26/>

¹³Users working near this cutoff time—e.g., at 2:55 AM GMT—should be aware that their files may disappear if alignment is not initiated in time.



personal data. However, we **do** retain anonymized alignment metadata – such as a randomly assigned alphanumeric user ID and summary usage statistics – to demonstrate the platform’s utility to funders.

8 Features and limitations

What Autophon is: Autophon is a web-based application designed to simplify forced alignment workflows and expand access for users with minimal technical background. It is particularly useful for research on under-resourced languages and non-standard varieties, and emphasizes ease of use, format flexibility, and language model diversity. The backend relies on existing forced alignment technologies developed over the past decades, wrapped in a modern frontend that facilitates fast, OS-independent processing.

Key features include:

1. Fully web-based and platform-independent (OS-agnostic).
2. No programming or installation required.
3. Accepts a wide range of transcription and audio formats.
4. Capable of processing low-resource and non-standard language varieties.
5. Supports user-defined pronunciation dictionaries and multiple transcription modes.

What Autophon is not: Important caveats to bear in mind:

1. Alignment quality depends on transcription accuracy, recording quality, and language characteristics.
2. Performance may vary across languages, dialects, and speaking styles.
3. Benchmarking accuracy is ongoing and not available for all models.
4. Core updates to underlying alignment engines may not be immediately reflected, due to the complexity of the Autophon backend.

9 Budget and funding

Autophon costs approximately 25 000 SEK (2 300 EUR) per year to run. Founded by Dr. Nate Young (who is the sole copyright holder), the project has since received support from the University of Helsinki, Linnaeus University, The Swedish Academy, the Department of Linguistics and Scandinavian Studies at the University of Oslo, and the European Union’s Horizon 2020 research and innovation programme under the Marie Skłodowska-Curie grant agreement No. 892963. Additional funding for language model development has come from The National Library of Norway¹⁴.

We continue to seek funding and welcome collaboration. If you are experienced in grant writing or interested in supporting the project, please reach out via the support page.

Acknowledgements

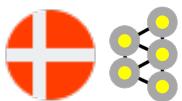
Numerous individuals have contributed to Autophon. We especially thank Michael McGarrah for strategic guidance and Kaosi Anikwe for extensive backend and frontend development. Ismail Raji Damilola helped implement a bootstrapping function to expand phoneme inventories. Additional contributions in the early stages came from Nabil Al Nazi, Zamanat Abbas Naqvi, and Santiago Recoba.

We also wish to acknowledge the people who helped make the DanFA model possible. Michael McGarrah and Joe Fruehwald helped develop DanFA version 1.0. Nicolai Pharao and Bjørn Falkencrone offered detailed feedback on how to improve the pronunciation dictionary from its original roughshod form. Nina Grønnum was kind enough to make her DanPASS recordings accessible, upon which versions 2.0 and 3.0 were trained. Nicolai Pharao, Christophe Zerakitsky Vies, and Rebecca Borg assisted with validation and testing.

References

Boersma, P., & Weenink, D. (2017). Praat: Doing phonetics by computer [softw.], ver.6.0.36. www.praat.org

¹⁴<https://www.nb.no/sprakbanken/ressurskatalog/oai-nb-no-sbr-59/>



- Grønnum, N. (2006). DanPASS - A Danish Phonetically Annotated Spontaneous Speech Corpus. *Proceedings of the Fifth International Conference on Language Resources and Evaluation (LREC'06)*. http://www.lrec-conf.org/proceedings/lrec2006/pdf/4_pdf.pdf
- Johannessen, J. B., Priestley, J., Hagen, K., Nøklestad, A., & Lynum, A. (2012). The Nordic Dialect Corpus. *Proceedings of the Eighth International Conference on Language Resources and Evaluation (LREC'12)*, 3387–3391. http://www.lrec-conf.org/proceedings/lrec2012/pdf/773_Paper.pdf
- McAuliffe, M., Socolof, M., Mihuc, S., Wagner, M., & Sonderegger, M. (2017). Montreal Forced Aligner: Trainable text-speech alignment using Kaldi. *Proceedings of Interspeech*, 498–502.
- Povey, D., Ghoshal, A., Boulian, G., Burget, L., Glembek, O., Goel, N., Hannemann, M., Motlicek, P., Qian, Y., Schwarz, P., et al. (2011). *The Kaldi speech recognition toolkit* (tech. rep.). IEEE Signal Processing Society. Piscataway.
- Rosenfelder, I., Fruehwald, J., Brickhouse, C., Evanini, K., Seyfarth, S., Gorman, K., Prichard, H., & Yuan, J. (2022). FAVE (Forced Alignment and Vowel Extraction) Program Suite v2.0.0 [Zenodo].
- Wilbanks, E. (2022). fseAlign (Version 1.1.14). <https://github.com/EricWilbanks/fseAlign>
- Young, N. J. (2017). DanFA 1.0 - Forced Alignment of Danish, ver. 1.0. <https://github.com/mcgarragh/LG-FAVE>
- Young, N. J. (2021). DanFA 2.0 - Forced Alignment of Danish, ver. 2.0. <https://www.nateyoung.se/dissemination>
- Young, N. J. (2023). DanFA 3.0 - Forced Alignment of Danish, ver. 3.0. www.autophon.org
- Young, N. J., Falkencrone, B., & Pharao, N. (2024). DanFA 4.0 - Forced Alignment of Danish, ver. 4.0. www.autophon.org
- Young, S. J., Woodland, P. C., & Byrne, W. J. (1993). HTK: Hidden Markov Model Toolkit V1.5. Cambridge Univ. Eng. Dept. Speech Group; Entropic Research Lab. Inc.