

# Nós-TTS: a Web User Interface for Galician Text-to-Speech

Carmen Magariños<sup>1</sup>, Alp Öktem<sup>2</sup>, Antonio Moscoso Sánchez<sup>1</sup>, Marta Vázquez Abuín<sup>1</sup>, Noelia García Díaz<sup>1</sup>, Adina Ioana Vladu<sup>1</sup>, Elisa Fernández Rei<sup>1</sup>, María Baqueiro Vidal<sup>3</sup>

<sup>1</sup>Instituto da Lingua Galega (ILG), Universidade de Santiago de Compostela, Spain

<sup>2</sup>Col-lectivaT, Spain

<sup>3</sup>Centro Singular de Investigación en Tecnoloxías Intelixentes (CiTIUS),  
Universidade de Santiago de Compostela, Spain

## Abstract

Speech synthesis, also known as text-to-speech (TTS), aims to generate human-like speech from text. The recent emergence of end-to-end deep learning TTS models has led to impressive natural-sounding results. Nevertheless, expanding these models to multiple languages and speakers poses challenges, especially for low- or limited-resource languages. In this context, we introduce Nós-TTS, a user-friendly web interface for Galician TTS developed under the Nós project. The proposed interface offers a choice among three distinct voices trained on diverse conditions regarding data quantity, training approach, and input modality. Although in an experimental stage, informal listening tests have shown a satisfactory performance of the models.

## 1 Introduction

Speech synthesis, also referred to as text-to-speech (TTS), is the automated generation of human-like speech by machines or computers (Dutoit, 1997; Taylor, 2009). More specifically, TTS techniques aim to synthesize intelligible and natural speech from input text. Over the years, different TTS approaches have been proposed (Tabet and Boughazi, 2011), the most prominent being concatenative unit-selection (Black and Campbell, 1995; Hunt and Black, 1996) and statistical parametric synthesis (Black et al., 2007; Zen et al., 2009).

In recent years, deep learning-based TTS systems have emerged as a powerful alternative to traditional synthesis (Ning et al., 2019; Tan et al., 2021). These systems use deep neural networks (DNNs) as the model backbone and have shown the ability to produce high-quality natural-sounding speech. However, these models often rely on massive single-speaker datasets (20-40 hours) for optimal performance. This high data demand poses a significant drawback, particularly for languages with limited resources, such as Galician, since ac-

quiring this data can be costly and time-consuming. While various techniques like transfer learning, multilingual training, or zero-shot learning have been applied to alleviate this problem (Casanova et al., 2022), their effectiveness still depends on factors such as the data quality, quantity, and the specific traits of the target languages.

Within the framework of the Nós project (Vladu et al., 2022; de Dios-Flores et al., 2022), we present Nós-TTS, a user-friendly web interface for Galician text-to-speech conversion. Nós-TTS allows users to input a text in Galician and synthesize the corresponding speech using one of three distinct voices: Celtia, Sabela, or Icíá. The underlying voice models are built on Variational Inference with adversarial learning for end-to-end Text-to-Speech (VITS) architecture (Kim et al., 2021) and were trained using the Coqui-TTS library (Eren and The Coqui TTS Team, 2021) on different Galician TTS datasets.

The following sections briefly describe the proposed system, discuss the evaluation strategy and main findings, and summarize future work.

## 2 System Description

Nós-TTS<sup>1</sup> is a web user interface for speech synthesis in Galician. As shown in Figure 1, the proposed interface takes an input text in Galician (up to 1000 characters) and generates the corresponding synthesized speech by clicking the “Xerar voz” button. Once available, the audio will automatically start playing, and audio controls will be shown, including play, pause, volume, and position bar. Depending on the browser, the playback speed may also be adjusted. Other available features are download the synthesized audio (down arrow button) and clear the input text (“Borrar texto” button).

<sup>1</sup><https://tts.nos.gal/>

## 2.1 Voice models

In its current version, the TTS system integrates three voice models with different characteristics:

- **Celtia.** Female voice model trained from scratch on a subset of the Nos\_Celtia-GL corpus (Vázquez Abuín et al., 2023). This corpus, created under the Nós project, comprises a total of 20,000 sentences recorded by a professional voice talent. Specifically, a subset of 13,000 sentences were used to train the model, which corresponds to 15.5 hours of speech. The Celtia model (Magariños, 2023) was trained directly on grapheme inputs and includes a text normalization step based on the front-end of Cotovía (Rodríguez Banga et al., 2012).
- **Sabela.** Female voice model trained from scratch on the Sabela corpus within the CRPIH UVigo-GL-Voices dataset (CRPIH and GTM, 2023). This corpus comprises 10,000 sentences recorded by a professional radio broadcaster, amounting to approximately 14 hours of speech. The Sabela model (Öktem and Magariños, 2023) was trained on phonemes and incorporates the Cotovía front-end for text normalization and grapheme-to-phoneme conversion.
- **Icía.** Female voice model fine-tuned from the previously described Celtia model using the Icía corpus within the CRPIH UVigo-GL-Voices dataset (CRPIH and GTM, 2023). The Icía corpus comprises around 3,000 sentences, equivalent to approximately 4 hours of speech, recorded by an amateur speaker. Icía is a phoneme-based model which integrates the front-end of Cotovía for both text normalization and phonetic transcription.

All the models are openly available in Hugging Face<sup>2</sup>.

## 2.2 Models' Architecture

The incorporated voice models are based on VITS (Kim et al., 2021), a fully end-to-end TTS model that leverages cutting-edge deep-learning techniques like adversarial learning (Goodfellow et al., 2014), normalizing flows (Rezende and Mohamed, 2015), variational auto-encoders (Kingma and

<sup>2</sup><https://huggingface.co/proxectonos>

Welling, 2014) and transformers (Vaswani et al., 2017; Shaw et al., 2018) to achieve results comparable to ground truth. Its architecture combines the Glow-TTS encoder (Kim et al., 2020) and HiFi-GAN vocoder (Kong et al., 2020) within the same training pipeline. By jointly learning the acoustic model and the vocoder, VITS overcomes some issues of the two-stage models. It also incorporates a stochastic duration predictor that allows synthesizing speech with different rhythms from the same input text.

## 2.3 Text Processing Module

The proposed TTS system includes a text processing module based on the Cotovía front-end. Depending on the selected voice, as described in Section 2.1, the text processing module performs one or both of the following functions: (1) text normalization; (2) phonetic transcription with stress marks.

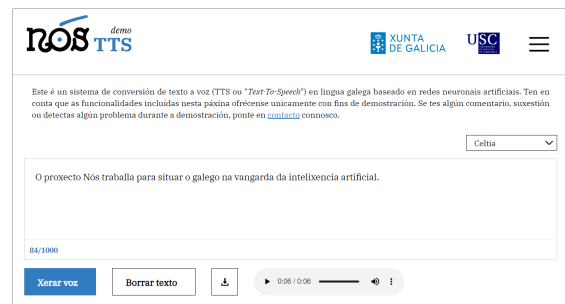


Figure 1: View of the NÓS-TTS user interface.

## 3 Evaluation and Discussion

Traditionally, the quality of TTS systems is assessed through perceptual listening tests conducted by humans. These tests commonly employ perceptual metrics, such as Mean Opinion Score (MOS) (Ling et al., 2021), to rate speech characteristics including overall quality, naturalness, or similarity to the target voice.

This form of subjective measures are the gold standard for the speech synthesis task, yet it proves to be time-consuming and demanding in terms of test preparation and listener recruitment. Consequently, models are typically initially assessed through informal listening tests, with more extensive formal evaluations reserved for final models.

While the models currently integrated into the NÓS-TTS interface are experimental, they have demonstrated competent performance in informal

listening tests, showcasing high naturalness and quality. For each voice, the models exhibiting superior performance in these informal evaluations will undergo subsequent formal evaluations involving a statistically significant number of listeners.

Nevertheless, these informal evaluations reveal insightful findings regarding the performance of the three voice models. Notably, the Celtia model stands out as the undisputed leader in terms of overall quality. Its superior results in audio quality, choice of pronunciation, precision of phonemes and naturalness of prosody, position it as the most robust and satisfactory option. This outstanding performance is directly attributed to the quality of the corpus used in its training, which was meticulously designed and developed to ensure balanced and representative textual content, voice talent with good vocal characteristics, and high-quality recordings.

Continuing in the ranking, the Icíá model is positioned as a solid alternative, despite the limited amount of data and the speaker being non-professional. In this case, the applied fine-tuning techniques have mitigated the data limitations, resulting in a synthetic voice with noticeably more natural prosody compared to the Sabela model.

On the other hand, the Sabela model faces significant challenges, primarily related to the lack of naturalness in prosody. This limitation is evident both in the original recordings used for training and in the generated synthetic voice. The main reason for this lack of naturalness seems to be the particular style and rigid prosody associated with typical news readings on radio and television used during the recordings. This finding underscores the importance of considering not only the quantity but also the quality and diversity of training data to achieve optimal results in speech synthesis.

Another important consideration when comparing the different models pertains to the input modality for training, namely graphemes versus phonemes. A model trained on phonemes is expected to converge more rapidly, and using phonemes as input is also anticipated to aid in disambiguating the pronunciation of specific graphemes. An example of this is the grapheme <x>, which in Galician can be pronounced as [ks] or [j] depending on the word. In this particular case, we have observed that the Celtia model, trained on graphemes, mispronounces this grapheme in some words (e.g., <x> as [j] instead of [ks] in *boxeo* and *axila*), whereas the Icíá and Sabela models, trained

on phonemes, correctly differentiate between the two pronunciations. We aim to address this minor drawback of the Celtia model by training a new model based on phonemes. This final comparison reveals the importance of having a proficient text processing module to achieve precise phonetic transcriptions.

## 4 Future Work

The proposed system is in a continuous improvement stage, with ongoing efforts to perfect the quality of the models and expand the voice catalog. Future work will involve testing new architectures for the existing voices and training new models with additional speakers, including male voices. As mentioned in Section 3, formal evaluations will be conducted on models achieving the best-perceived performance in informal listening tests. We also plan to improve the text processing module by implementing changes in the Cotovía front-end.

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