

A COMPARISON OF CONVOLUTIONAL NEURAL NETWORKS FOR GLOTTAL CLOSURE INSTANT DETECTION FROM RAW SPEECH

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ABSTRACT

In this paper, we continue to investigate the use of machine learning for the automatic detection of glottal closure instants (GCIs) from raw speech. We compare several deep one-dimensional convolutional neural network architectures on the same data and show that the InceptionV3 model yields the best results on the test set. On publicly available databases, the proposed 1D InceptionV3 outperforms XGBoost, a non-deep machine learning model, as well as other traditional GCI detection algorithms.

Index Terms— glottal closure instant (GCI), detection, deep learning, convolutional neural network

1. INTRODUCTION

Machine learning is gaining more and more attraction in many areas of signal processing, replacing the established and refined signal processing techniques (such as autocorrelation, convolution, Fourier and wavelet transforms and many others), or speech/audio processing techniques (such as Gaussian mixture models or hidden Markov models) [1]. It is also the case of *glottal closure instant detection*, a traditional signal processing/detection task. Detection of glottal closure instants (GCIs) could be viewed as a task of determining peaks in the *voiced parts* of the speech signal that correspond to the moment of glottal closure, a significant excitation of the vocal tract during speaking.

In our previous research [2, 3], we showed that classical (“non-deep”) machine learning, and especially the one based on *extreme gradient boosting* (XGBoost), was able to perform very well and consistently outperformed traditionally used algorithms on several test datasets [3]. From the point of view of machine learning, GCI detection could be described as a two-class classification problem: whether or not a peak in a speech waveform represents a GCI [4]. Unlike the traditionally used algorithms, which usually exploit expert knowledge and hand-crafted rules and thresholds to identify GCI candidates from local maxima of various speech representations (see, e.g. [5]), the advantage of a machine-learning-based method is that once a training dataset is available and relevant features identified from raw speech, classifier parameters are set up automatically without manual tuning. On the other hand, the identification of relevant features may be time-consuming and tricky, especially when carried out by hand.

Deep learning, and especially *convolutional neural networks* (CNNs), can help solve the problem of identifying features. In general, deep learning can help in finding more complex dependencies

Table 1. Train/validation/test dataset description.

Dataset	Train	Val.	Test	Total
# utterances	3,136	32	32	3,200
length (minutes)	331.28	3.52	3.48	338.28
# peaks	2,127,650	22,644	22397	2,172,691
# GCIs	1,767,752	18,901	18687	1,805,340
# non-GCIs	359,898	3743	3710	367,351

between raw speech and the corresponding GCIs. CNNs can directly be applied to the raw speech signal without requiring any pre- or post-processing, such as feature identification, extraction, selection, dimension reduction, etc. [6, 7]. CNNs were already shown to perform very well in GCI detection [8, 9, 10, 11].

In this paper, we investigate several deep one-dimensional (1D) CNN architectures in the context of GCI detection and compare them with non-deep machine learning XGBoost and with traditional GCI detection algorithms on the same data.

2. DATA DESCRIPTION

Experiments were performed on clean 16 kHz sampled speech recordings primarily intended for speech synthesis. We used 3200 utterances from 16 voice talents (8 male and 8 female voices with 200 utterances per voice) of different languages (8 Czech, 2 Slovak, 3 US English, Russian, German, and French). Two voices were from CMU ARCTIC database [12, 13] (Canadian English JMK and Indian English KSP), the rest were our proprietary voices. For our purposes, speech waveforms were mastered to have equal loudness and negative polarity [14]. Ground truth GCIs were detected from contemporaneous EGG recordings by the Multi-Phase Algorithm (MPA) [15] and shifted towards the neighboring minimum negative sample in the speech signal. The ratio of division into train/validation/test sets was set to 98/1/1 (see Table 1 for more details). Each voice was part of the train, validation and test dataset.

Since the classification of peaks as GCI/non-GCI is performed in a peak-by-peak manner, negative peaks were detected by zero-crossing low-pass filtered (by a zero-phase Equiripple-designed filter with 0.5 dB ripple in the pass band, 60 dB attenuation in the stop band, and with the cutoff frequency of 800 Hz) speech signal exactly in the same way as described in [16] (see also Fig. 1). It was also found that downsampling to 8 kHz prior filtering provided slightly better results than the use of 16 kHz. Thus, all compared CNNs use 8 kHz internally.

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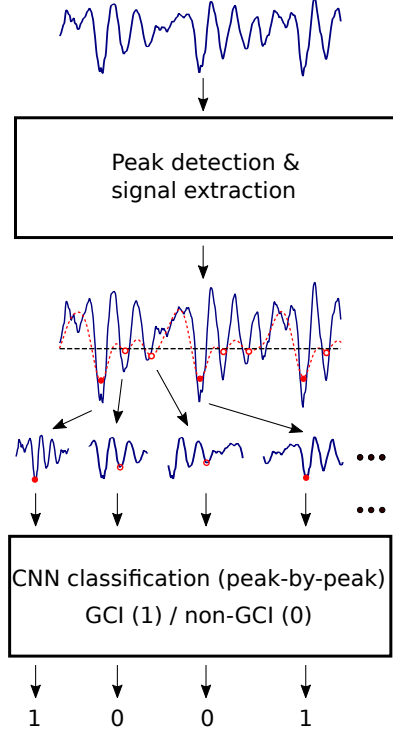


Fig. 1. A simplified scheme of a CNN-based GCI detection.

3. EXPERIMENTS

Unlike classical (non-deep) machine learning algorithms, CNNs have the unique ability to fuse feature extraction and classification into a single learning body, and thus eliminate the need for fixed and hand-crafted features. Typically, each CNN consists of a series of *convolutional layers* (convolving their input with learnable kernels and computing feature maps) interleaved with *pooling layers* (downsampling the learned feature maps), followed by one or more dense layers which perform the actual classification. Conventional (2D) CNNs were originally introduced to perform object recognition/detection tasks for 2D signals (images or video frames), and since then they became the state-of-the-art technique for many computer vision tasks [17, 18].

In this section, we compare several 1D CNN architectures on the same data. For this purpose, we implemented 1D versions of well-known 2D architectures primarily proposed for image processing (LeNet-5 [19], various VGG networks [20], Inception [21], ResNet [22], Xception [23], NASNet [24], ShuffleNet [25]) in the Keras framework [26]. We also included two 1D architectures proposed directly for audio processing: “Yang2018” (a VGG10-like network [8]) and SwishNet [27].

Due to the unbalanced number of GCIs and non-GCIs in our data (see Table 1), the comparison was made with respect to $F1$, $recall$ (R), and $precision$ (P) scores.

3.1. Initial Comparison

The purpose of the initial comparison of different architectures/models was to identify the capabilities of the models in the context of GCI detection, and to discard some less powerful models from further evaluation.

Table 2. Initial comparison of GCI detection performance using 1D CNNs on the validation set (in percents) including the approximate number of learnable parameters. VGG4/6 are simple VGG networks with only 4/6 layers including only one dense layer with 32 neurons.

CNN	$F1$	R	P	# params
InceptionV3 [28]	98.77	98.86	98.67	12.3M
NASNet-Mobile [24]	98.77	98.81	98.72	4.0M
VGG13 [20]	98.76	98.73	98.79	34.6M
ResNet101 [22]	98.73	98.73	98.74	28.3M
Xception [23]	98.73	98.48	98.97	20.7M
NASNet-Large [24]	98.72	98.77	98.68	83.6M
Inception-ResNetV2 [29]	98.71	98.66	98.76	44.7M
ResNet152 [22]	98.67	98.73	98.62	38.5M
ResNet50 [22]	98.66	98.56	98.77	16.0M
VGG11 [20]	98.66	98.32	99.00	34.5M
VGG16 [20]	98.65	98.37	98.94	36.4M
VGG19 [20]	98.61	98.58	98.64	38.2M
SwishNet-Wide [27]	98.50	98.39	98.61	17k
ShuffleNet [25]	98.49	98.32	98.67	0.5M
SwishNet-Slim [27]	98.49	98.42	98.56	4k
LeNet-5 [19]	98.44	98.28	98.60	0.5M
VGG4 [20]	98.07	97.62	98.52	0.1M
VGG6 [20]	98.02	97.58	98.46	0.1M
Yang2018 [8]	97.55	96.68	98.43	2.9M

Table 3. Optimal hyper-parameter values of the selected models and the best $F1$ score achieved on the validation set.

CNN	FS	W	LR	BS	$F1$ (%)
Inception-ResNetV2	80	rect.	0.001	64	98.97
NASNet-Mobile	80	rect.	0.001	256	98.91
InceptionV3	80	hamm.	0.001	128	98.90
VGG13	48	rect.	0.001	128	98.88
ResNet101	80	rect.	0.0001	64	98.87
Xception	48	rect.	0.0001	16	98.86

The architecture of each model (the number of layers, the number of filters and their sizes, the usage of pool layers, batch normalization, etc.) was used the same as proposed by the authors of each model. In all our experiments, the networks were trained to minimize a *binary cross-entropy loss* using *mini-batch gradient descent* with the *Adam optimizer*. *ReLU activations* were applied in all inner layers, whereas a *sigmoid activation* was used in the last (dense) layer. For the initial comparison, the mini-batch size was 128 and the learning rate was 0.001. To speed up the training, it was stopped when the validation loss did not improve for 10 epochs and the maximum number of epochs was set to 100. The length of the input speech segment extracted around each negative peak was fixed to 30 ms (i.e., the frame size was 240 samples).

Based on the results shown in Table 2, the following networks were chosen for further fine-tuning: InceptionV3, NASNet-Mobile, VGG13, ResNet101, Xception, and Inception-ResNetV2.

3.2. Model Tuning

In this stage, we focused on the selected models and tuned their hyper-parameters on the validation set. The following hyper-parameters were taken into account in our comparison: the size of the frame around each negative peak (FS: 30–128 ms, i.e. 240–1024 samples) and windowing (W: rectangular or Hamming), learning rate (LR:

Table 4. Comparison of the finalized GCI detection models (trained for the given number of epochs (# ep.) on train and validation sets, including the number of learnable parameters # par.) on the test set (left) and the corresponding statistical significance according to McNemar’s test [30] (right). The symbols \gg and $>$ mean that the row model is significantly better at the significance level $\alpha = 0.01$ and $\alpha = 0.05$, respectively, than the column model. The symbol $=$ means that the respective models perform the same.

CNN	F1	R	P	# ep.	# par.		INC	XCE	INR	RSN	VGG	NNM
InceptionV3 (INC)	98.94	98.94	98.94	8	12.3M	INC	=	>	>	>	>	>
Xception (XCE)	98.85	98.81	98.89	7	20.7M	XCE	<	=	=	=	=	>
Inception-ResNetV2 (INR)	98.84	99.15	98.92	9	44.7M	INR	<	=	=	=	=	>
ResNet101 (RSN)	98.84	98.93	98.74	10	28.3M	RSN	<	=	=	=	=	>
VGG13 (VGG)	98.81	98.94	98.69	17	34.6M	VGG	<	=	=	=	=	=
NASNet-Mobile (NNM)	98.78	98.91	98.65	7	4.0M	NNM	<	<	<	<	=	=

0.0001–0.1), and mini-batch size (BS: 16–512). The optimal hyperparameter values are shown in Table 3.

3.3. Model Testing

Finally, the tuned models were finalized, i.e., trained on both train and validation datasets for the number of epochs found during the model tuning phase in Section 3.2 (see Table 4), and evaluated on the test set. Due to the stochastic nature of neural network training algorithms (especially when using GPU), we repeated the training five times. A more robust final comparison was then achieved by evaluating the models’ performance over all runs.

It could be seen in Table 4 that the 1D InceptionV3 model outperforms all other models at the statistical significance level $\alpha = 0.05$. Other models performed about the same except for 1D NASNet-Mobile which achieved the worst results.

4. COMPARISON WITH OTHER METHODS

To compare the proposed Inception model with different GCI detection algorithms, standard GCI detection measures that concern the *reliability* and *accuracy* of the GCI detection algorithms were used [31]. The former includes the percentage of glottal closures for which exactly one GCI is detected (*identification rate*, IDR), the percentage of glottal closures for which no GCI is detected (*miss rate*, MR), and the percentage of glottal closures for which more than one GCI is detected (*false alarm rate*, FAR). The latter includes the percentage of detection with the identification error $\zeta \leq 0.25$ ms (*accuracy to ± 0.25 ms*, A25) and the standard deviation of the identification error ζ (*identification accuracy*, IDA). In addition, we use a more *dynamic evaluation measure* [32]

$$E10 = \frac{N_{GT} - N_{\zeta > 0.1T_0} - N_M - N_{FA}}{N_{GT}} \quad (1)$$

that combines the reliability and accuracy in a single score and reflects the local *pitch period* T_0 pattern (determined from the ground truth GCIs). N_{GT} stands for the number of ground truth GCIs, N_M is the number of missing GCIs (corresponding to MR), N_{FA} is the number of false GCIs (corresponding to FAR), and $N_{\zeta > 0.1T_0}$ is the number of GCIs with the identification error ζ greater than 10% of the local pitch period T_0 . For the alignment between the detected and ground truth GCIs, dynamic programming was employed [32].

4.1. Compared methods

We compared the proposed 1D convolutional network InceptionV3 with a traditional machine learning-based algorithm XGBoost [3] and with six existing state-of-the-art GCI detection methods shown in

Table 5. We used the implementations available online; no modifications of the algorithms were made. Since all algorithms (except REAPER) estimate GCIs also during unvoiced segments, their authors recommend filtering the detected GCIs by the output of a separate voiced/unvoiced detector. We applied an F_0 contour estimated by the REAPER algorithm for this purpose. There is no need to apply such post-processing on GCIs detected by InceptionV3-1D and XGBoost since the voiced/unvoiced pattern is used internally in these methods. To obtain consistent results for all methods, the detected GCIs were shifted towards the neighboring minimum negative sample in the speech signal.

4.2. Test datasets

Two voices, a US male (BDL) and a US female (SLT) from the CMU ARCTIC database [12, 13], were used as a test material. Each voice consists of 1132 phonetically balanced utterances of total duration ≈ 54 minutes per voice. Additionally, KED TIMIT database [13], comprising 453 phonetically balanced utterances (≈ 20 min.) of a US male speaker, was also used for testing. All these datasets comprise clean speech. Ground truth GCIs were detected from contemporaneous EGG recordings in the same way as described in Section 2 (again shifted towards the neighboring minimum negative sample in the speech signal)¹. Original speech signals were downsampled to 16 kHz and checked to have the same polarity as described in Section 2. It is important to mention that none of the voices from these datasets was part of the training dataset used to train InceptionV3-1D and XGBoost models.

4.3. Results

The results in Table 5 show that the Inception network performs very well for all tested datasets². It generally outperforms XGBoost and other algorithms. It excels in terms of *reliability*, especially with respect to the identification (IDR) and false alarm (FAR) rates. As for the *accuracy*, it also performed reasonably well as it often achieved the second-best results (behind the GEFBA algorithm which, however, tends to miss GCIs quite often) in terms of identification accuracy (IDA) and of the smallest number of timing errors higher than 0.25 ms (A25). The proposed InceptionV3-1D also achieved the best results with respect to the combined dynamic evaluation measure (E10).

¹The ground truth GCIs and other data relevant to the described experiments are available online [38].

²A possible explanation of lower performance metrics (cf. e.g. [5, 31]) is the use of different ground truth GCIs, a different strategy of GCI filtering in unvoiced segments, and perhaps also a different implementation of GCI computation evaluation (also available in [38]).

Table 5. Comparison of GCI detection of the proposed InceptionV3-1D CNN with other algorithms.

Dataset	Method	IDR (%)	MR (%)	FAR (%)	IDA (ms)	A25 (%)	E10 (%)
BDL	InceptionV3-1D	94.34	3.99	1.67	0.53	98.89	93.37
	XGBoost [3]	93.85	2.37	3.78	0.41	98.34	92.36
	SEDREAMS [33]	91.80	3.03	5.16	0.45	97.37	90.02
	MMF [34]	90.42	4.63	4.95	0.56	97.15	87.87
	DYPSA [31]	89.43	4.38	6.19	0.54	97.13	86.89
	REAPER [35]	93.24	4.39	2.37	0.56	98.01	91.47
	GEFBA [36]	87.93	10.05	2.02	1.02	99.11	87.18
SLT	PSFM [37]	87.05	9.65	3.30	0.71	96.95	84.50
	InceptionV3-1D	96.84	1.36	1.80	0.17	99.73	96.59
	XGBoost [3]	96.05	0.57	3.38	0.17	99.71	95.78
	SEDREAMS [33]	94.66	1.13	4.21	0.17	99.67	94.36
	MMF [34]	92.44	5.29	2.26	0.40	99.17	91.78
	DYPSA [31]	93.25	2.91	3.84	0.32	99.39	92.75
	REAPER [35]	95.57	1.66	2.77	0.19	99.67	95.27
KED	GEFBA [36]	94.85	2.62	2.53	0.17	99.76	94.63
	PSFM [37]	86.95	10.46	2.60	0.45	99.26	86.42
	InceptionV3-1D	96.22	2.71	1.08	0.24	99.60	95.87
	XGBoost [3]	95.70	1.29	3.02	0.25	99.64	95.37
	SEDREAMS [33]	92.30	6.03	1.66	0.29	99.12	91.76
	MMF [34]	90.16	7.16	2.68	0.35	98.99	89.52
	DYPSA [31]	90.27	7.07	2.65	0.30	99.25	89.72
TOTAL	REAPER [35]	91.05	8.18	0.78	0.28	99.47	90.67
	GEFBA [36]	88.51	10.36	1.13	0.21	99.74	88.30
	PSFM [37]	89.47	9.59	0.94	0.39	99.22	88.85
	InceptionV3-1D	95.87	2.46	1.68	0.35	99.41	95.35
	XGBoost [3]	95.22	1.30	3.48	0.29	99.21	94.49
	SEDREAMS [33]	93.37	2.34	4.29	0.31	98.79	92.51
	MMF [34]	91.47	5.25	3.29	0.46	98.41	90.12
TOTAL	DYPSA [31]	90.27	7.07	2.65	0.30	99.25	89.72
	REAPER [35]	94.25	3.34	2.41	0.37	99.05	93.40
	GEFBA [36]	91.66	6.14	2.20	0.62	99.53	91.26

5. CONCLUSIONS

In this paper, we followed up on our previous work concerning the use of machine learning to detect GCIs from raw speech. We compared several deep one-dimensional CNN architectures on the same data and selected InceptionV3-1D that achieved the best results on the test set ($F1 = 98.94\%$). The InceptionV3-1D model outperforms other traditional state-of-the-art algorithms on several publicly available test datasets.

InceptionV3-1D also outperforms XGBoost, a non-deep machine learning model. It is a good finding because, thanks to its convolutional structure, InceptionV3-1D can directly be applied to the raw speech signal without requiring any pre- or post-processing (such as feature identification, extraction, selection, dimension reduction, etc.) which makes the use of classical machine-learning algorithms more difficult.

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