NAViDAd: A No-Reference Audio-Visual Quality Metric Based on a Deep Autoencoder

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Abstract—The development of models for quality prediction of both audio and video signals is a fairly mature field. But, although several multimodal models have been proposed, the area of audio-visual quality prediction is still an emerging area. In fact, despite the reasonable performance obtained by combination and parametric metrics, currently there is no reliable pixel-based audio-visual quality metric. The approach presented in this work is based on the assumption that autoencoders, fed with descriptive audio and video features, might produce a set of features that is able to describe the complex audio and video interactions. Based on this hypothesis, we propose a No-Reference Audio-Visual Quality Metric Based on a Deep Autoencoder (NAViDAd). The model visual features are natural scene statistics (NSS) and spatial-temporal measures of the video component. Meanwhile, the audio features are obtained by computing the spectrogram representation of the audio component. The model is formed by a 2-layer framework that includes a deep autoencoder layer and a classification layer. These two layers are stacked and trained to build the deep neural network model. The model is trained and tested using a large set of stimuli, containing representative audio and video artifacts. The model performed well when tested against the UnB-AV and the LiveNetflix-II databases.

Index Terms—audio-visual, quality metrics, no-reference, distortions, autoencoder, NAViDAd

I. INTRODUCTION

The great progress achieved by communications technology in the last twenty years is reflected by the amount of multimedia services currently available, such as digital television, IP-based video transmission, and mobile services. Among the most popular multimedia services are IP-based transmission, including video conference (Skype, Google Hangout, Facebook Video, FaceTime) and on-demand streaming media (Netflix, iTunes, Hulu, Amazon). Yet, it is understood that the success of these kind of services relies on their trustworthiness and the provided quality of experience [1]. Therefore, the development of efficient real-time monitoring quality tools, which can quantify the audio-visual experience, is key to the success of any multimedia service or application. Although the research in audio and video quality assessment (tackled as individual modalities) is fairly mature [2], there are still several issues to be solved in the area of audio-visual quality.

Audio and visual descriptive features have been used in several applications, such as speech intelligibility and pattern recognition [3], [4]. Their performance naturally relies on how good these features are able to describe the signal characteristics, specially in terms of human perception. In the quality assessment area, there are several audio and video quality metrics that achieve very good performances using audio and video features, respectively, to predict the perceived quality [2]. But, currently there is no feature-based quality metric that estimates the quality of an audio-visual signal, taking into consideration the characteristics the audio and visual components and their important interactions.

Considering these issues, Machine Learning (ML) paradigms arise as an appealing option to tackle the audiovisual quality assessment problem from a different perspective. Ouality assessment methods based on ML are capable of mimicking human reactions to media distortions, instead of explicitly modelling it. Soni et al. used a deep autoencoder strategy to design a non-intrusive speech quality assessment method [5]. The proposed metric adopts a two-layer approach to treat speech background noise distortions, using audio features in the form of spectrograms. In the first layer, a speech spectrogram is passed by a two-layer autoencoder to produce a low-dimensional set of new features. A mapping function between the features and subjective scores is found using an artificial neural network (ANN). Results showed that this deep autoencoder approach produced better descriptive features than Filterbank Energies (FBEs) and more accurate speech quality predictions. We believe that deep autoencoders techniques can be used for the particular task of finding ways to describe the audio and video components interact.

In this work, we design a No-reference Audio-VIsual Quality metric based on a Deep AutoencoDer (NAViDAd). The proposed model seeks to blindly estimate audio-visual quality for streaming multimedia applications. To estimate the audiovisual quality, the model uses both audio and video descriptive features. This way, it takes into account not only for the quality of the individual components, but also their interactions. First, a set of features that describe the characteristics of the audio and video components are computed. Next, the set of features are passed to a trained model that is composed of two layers:

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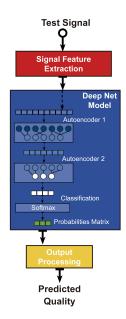


Fig. 1. Block diagram of the Audio-VIsual Quality metric based on a Deep AutoencoDer (NAViDAd).

an autoencoder layer and a classification layer. The autoencoder layer produces a low-dimensional set of features. At this stage, it is expected that these low-dimensional set of features are able to describe the complex interaction between audio and video stimulus. The classification layer is responsible of mapping the features into audio-visual quality scores. Finally, the last stage takes the model output to processed it and calculate the overall audio-visual quality. Figure 1 depicts the block diagram of NAViDAd, depicting all three stages of the proposed metric.

The structure of this paper is organized as follows. In Section II, the architecture of the proposed metric is described. In this Section we detail the extraction of the audio and video features as well as the proposed audio-visual quality assessment model. In Section III we present the results obtained with the proposed model. Finally, in Section IV some conclusions and final comments are presented.

II. PROPOSED ARCHITECTURE

In this section, we describe the architecture of the proposed NR audio-visual quality metric, which includes: feature extraction, model training and testing, and output processing.

A. Extraction of Audio-Visual Features

Natural Scene Statistics (NSS) and the spatial and temporal information were used as visual features. We used the feature extract function from the Diivine image quality metric to extract a total of eighty-eight (88) features [6], resulting in an 88-by-n matrix (n is the number of video frames) that represents the NSS features. To capture the spatial and temporal characteristics of the video, we used the algorithm proposed by Ostaszewska and Kloda [7] to compute the spatial and temporal information, helping characterize important visual distortions (e.g. freezing and packet loss distortions). Spatial

and temporal values are computed for each video frame, resulting in a 2-by-n matrix that represents the spatial and temporal features. Both sets of features are merged to form the *visual* features of a video sequence, represented by a 90-by-n matrix.

ViSQOL [8] and ViSQOLAudio [9] are full reference speech and audio quality metrics based on the same underlying platform. They use an intensity spectrogram representation of the audio signal, i.e. a time-frequency intensity representation of the audio activity, as the audio feature source for quality prediction. In this work, we use the feature extraction functionality of ViSQOL to process the audio signal and obtain a 25-by-m matrix, where 25 is the number of frequency bands and m is the number of audio frames in the signal.

B. Combination of Audio-Visual Features

Once the visual features (90-by-n matrix) and the audio features (25-by-m matrix) are obtained, they are merged together resulting in a total of 115 descriptive features. However, given that the number of video frames (n) and the number of audio samples (m) are not necessarily the same, a scaling process is required to match these two sets before merging them. To uniformize the length of the two matrices, we simply replicated the values of the matrix that has the shorter length, so that it matches the size of the larger matrix. Since the number of frame videos (n) is generally smaller than the number of audio samples (m), values of the visual feature set are replicated to match the audio feature set. Once the length of both sets matches, they are merged to form a 115-by-m matrix, denoted as the audiovisual feature set (115 is the sum of the 90 visual features plus the 25 audio features).

Additionally, a *target set* is built using the subjective scores associated with each video under analysis. This set contains the target quality scores used during the model training. Since in an ACR quality scale there are 4 quality groups, which represent the quality intervals assigned to the stimuli, the target set is a 4-by-m matrix composed of zeros and ones, where 4 is the number of quality groups and m is the length of the features matrix (115-by-m). The target set is built by taking the average subjective score associated with the stimuli and setting to one the corresponding quality group. For example, a sequence that has a subjective score of 1.65 is assigned the quality group 1 since the score is in the interval [1, 2), while a sequence with a subjective score of 3.52 is assigned to the quality group 3 since the score is in the interval [3, 4). In summary, the row corresponding to the corresponding quality group is set to one and the rest of the rows are set to zero. Considering that each column represents a video sample, this setup guarantees that each sample has only one quality group associated to it. During the model training, this target set is used to map the corresponding quality group of each sample.

Then, the feature and target matrices for all the audiovisual signals of the dataset are concatenated to produce two large global sets. The number of rows of the global feature and target matrices are 115 and 4, respectively. Meanwhile, the columns of the matrices are denoted by M, which represents

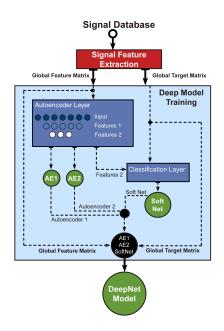


Fig. 2. Simplified block diagram of the No-Reference Audio-Visual Quality Metric Based on a Deep Autoencoder (NAViDAd).

the sum of all number of video samples from the training dataset. These two global sets served as input for the model training at different stages.

C. NAViDAd Training

A basic block diagram of the proposed No-Reference Audio-Visual Quality Metric Based on a Deep Autoencoder (NAViDAd) is presented in Figure 2. The training phase consists of two main layers: 1) the autoencoder layer, that receives as input the global feature set, and 2) the classification layer, which receives the global target set and a low-dimensional set of features. Once trained, the resulting models are stacked and trained to build the final deep audiovisual quality model.

1) Autoencoder Layer: The autoencoder layer produces a low-dimensional set of features that are able to describe the audio and video characteristics, as well as the distortions associated with the signal. Two sub-layers (autoencoders) are used in this phase. It is worth mentioning that this is a demonstrative structure, further tests might add more sub-layers depending on the requirements of the model. The first layer receives the global 115-by-m matrix containing the audiovisual features from the training stimuli. This first auteoncoder is trained using a hidden layer of size 60, generating as output a 60by-m matrix denoted as *Features 1*. Another output of this layer is a trained autoencoder, denoted as Autoencoder 1. The second autoencoder has a hidden layer of size 25 and is trained using as input the Features 1. It generates as output a 25-bym matrix, denoted as Features 2, and a trained autoencoder, denoted as Autoencoder 2. The training parameters for the two autoencoders are the following: a linear transfer function is set for the decoder, the L2 weight regularizer is 0.001, the sparsity regularizer is 4, and the sparsity proportion is 0.05.

In summary, the autoencoder produces: two trained autoencoders (*Autoencoder 1* and *Autoencoder 2*) and two sets of features (*Features 1* and *Features 2*). From these outputs, only *Features 2* is used as input to the following layer: the classification layer. *Autoencoder 1* and *Autoencoder 2* are used during the overall training of the deep neural network model.

2) Classification Layer: This layer has the goal of finding a mapping function between the input feature set of the training stimuli and the corresponding subjective scores. To obtain this mapping, a softmax function is used to discover the quality group corresponding to the set of features. This layer receives the *Features 2* set obtained in the previous layer and a 4-by-*m* target set and performs the training of the classification function. The resulting function, denoted as Soft Net, is trained to generate a matrix containing the probabilities that a certain sample belongs to each quality group. After the autoencoders (*Autoencoder 1* and *Autoencoder 2*) and the classification function (Soft Net) are trained, they are stacked to form the deep neural network (Deep Autoencoder Network). Finally, the network is trained using the global feature and the global target sets gathered in the previous layers.

A more detailed description of the entire training procedure and information regarding the parameters used for training the model can be found at [10].

D. NAViDAd Testing

To extract the audio-visual features of the test stimuli, the testing stage uses the same procedure used in the training stage. The extracted set of features (again, a 115-by-mmatrix) are, then, passed to the deep autoencoder network. Next, the output (4-by-m matrix) is processed to compute the audiovisual predicted quality. First, the maximum value of each column and its corresponding row index in the 4by-m matrix are computed. Then, a 1-by-m vector is built by adding the index and the maximum value, i.e. for each column the corresponding quality group index is summed to the corresponding probability value resulting in a quality (real) number in the interval [1, 5]. Finally, the quality scores of all m columns are averaged and the overall audiovisual quality score is computed.

III. PERFORMANCE ANALYSIS

The NAViDAd model was trained and tested using sequences from the UnB-AVQ database. The UnB-AVQ is a large dataset of audio-visual stimuli (video sequences with accompanying audio) with their corresponding quality scores [11]. In this work, we used the third part (Experiment 3) of this dataset, which contains a total of 800 sequences with combinations of audio and video distortions [11]. The video distortions were Bitrate compression, Packet-Loss, and Frame-Freezing. The audio distortions were: Background noise, Chop, Clip, and Echo. Detailed information regarding the experiment procedure and the distortion parameters can be found in our previous work [11].

We used a 10-fold cross-validation method to train and test the proposed metric. We compared NAViDAd with the following quality metrics:

- FR visual metrics: SSIM and PSNR;
- NR visual metrics: DIIVINE, VIIDEO, BIQI, NIQE, and BRISQUE;
- FR audio metrics: VISQOLAudio, PEAQ, and VISQOL (speech);
- NR Audio metric: P.563 (speech);
- NR Audio-visual metrics: Linear, Minkowski, and Power models, using DIIVINE and P.563.

It is worth pointing out that these metrics were designed for a variety of contexts and were trained with different content. With respect to the audio-visual metrics, given the limitation of space, we have chosen to combine the outputs of the best performing NR audio and video quality metrics.

Table I presents the Pearson and Spearman correlation coefficients (PCC and SCC, respectively), along with the root mean square error (RMSE) obtained for the considered visual and audio-visual quality metrics. Results are organized according to the type of video distortion, given that the visual quality metrics cannot differentiate audio degradations in the stimuli. From Table I, it can be observed that NAViDAd has the best overall accuracy performance (All), when compared to the other visual quality metrics. As for the audio-visual combination models, NAViDAd also shows a clear advantage. When we consider the type of visual distortion, NAViDAd presents the best performance for both frame freezing (0.91)and packetloss (0.86) distortions. Notice that several visual (SSIM, NIQE, BRISQE) and audio-visual quality metrics present a lower performance for one type of visual distortion, while NAViDAd achieves a consistent performance.

Table II presents the results for audio and audio-visual metrics. This time, the results are separated by the audio distortions, given that the audio metrics cannot differentiate video distortions in the stimuli. From Table II, it can be observed that NAViDAd has the best overall accuracy performance (All), when compared to the audio and speech quality metrics. This advantage was expected since audio and speech metrics use only the audio-visual combination models, NAViDAd also shows a clear advantage. Regarding the audio distortions, interestingly, NAViDAd presents a better performance for chop and echo distortions (0.92 and 0.90). With regard to the audio-visual combination models, it is clear that NAViDAd performs better and shows a clear advantage.

For a better visualization of the results, Figure 3 (a) and (b) depicts bar plots of the overall PCC and SCC values (over the 10 folds) for all metrics. Besides the high correlation values presented by the NAViDAd metric, it can observed that results presented a small variation on both PCC and SCC coefficients. This shows that NAViDAd's results are very consistent compared to the rest of the literature metrics.

To validate the proposed method, we performed a crossvalidation test that consists of testing the method on an independent database, for which no training was performed. With this goal, we tested NAViDAd on the LiveNetflix-II Database, provided by the Laboratory for Image and Video Engineering (LIVE) of the University of Texas at Austin [19].

TABLE I PCC, SCC, RMSE FOR THE FR AND NR VISUAL AND AUDIO-VISUAL QUALITY METRICS, TESTED ON THE UNB-AVQ DATABASE.

PSNR		Packet-Loss	Frame-Freezing	All
	PCC	0.8997	0.8629	0.7694
	SCC	0.9455	0.8833	0.7368
	RMSE	19.2054	16.5837	18.0728
SSIM [12]	PCC	0.8563	0.3899	0.3620
	SCC	0.8500	0.3727	0.3579
	RMSE	2.7378	2.2027	2.4579
DIIVINE [6]	PCC	-0.8071	-0.8647	-0.8344
	SCC	-0.8182	-0.5167	-0.7519
	RMSE	2.4662	2.9484	2.6939
VIIDEO [13]	PCC	-0.7968	-0.9883	-0.8496
	SCC	-0.6729	-0.9234	-0.7834
	RMSE	2.2337	2.6804	2.4449
BIQI [14]	PCC	-0.8575	-0.9022	-0.8310
	SCC	-0.9382	-0.6000	-0.8799
	RMSE	34.8427	32.6918	33.8917
NIQE [15]	PCC	-0.7608	-0.9332	-0.8394
	SCC	-0.7798	-0.7289	-0.7195
	RMSE	2.9388	2.4057	2.7119
BRISQUE [16]	PCC	-0.7094	-0.9525	-0.8395
	SCC	-0.6360	-0.9662	-0.7728
	RMSE	45.1371	41.4226	43.5049
AV-Linear	PCC	0.3919	0.5501	0.4431
	SCC	0.2455	0.6333	0.3368
	RMSE	10.5249	11.0035	10.7430
AV-Minkowski	PCC	0.2912	0.4594	0.3422
	SCC	0.2091	0.6333	0.3143
	RMSE	1.9879	2.4289	2.1973
AV-Power	PCC	-0.6273	-0.6938	-0.6616
	SCC	-0.6727	-0.4333	-0.6075
	RMSE	24.2614	23.7806	24.0462
NAViDAd	PCC	0.8638	0.9167	0.8819
	SCC	0.8773	0.9050	0.8904
	RMSE	0.5931	0.5718	0.5850

PCC, SCC, AND RMSE FOR THE FR AND NR AUDIO AND AUDIO-VISUAL QUALITY METRICS, TESTED ON THE UNB-AVQ DATABASE.

Metric	Measure	Noise	Chop	Clip	Echo	All
VISQOLAudio [9]	PCC	0.7945	0.9909	0.7429	0.6844	0.6008
	SCC	0.7000	1.0000	0.4928	0.5218	0.4781
	RMSE	2.4702	2.2047	2.0815	2.2300	2.2464
VISQOL [8]	PCC	0.6102	0.9915	0.5084	0.4963	0.4236
	SCC	0.7000	1.0000	0.4928	0.5218	0.4645
	RMSE	2.6143	2.2045	2.1639	2.3136	2.3341
PEAQ [17]	PCC	0.7573	0.9347	0.8261	0.7096	0.7689
	SCC	0.2000	1.0000	0.3189	0.3479	0.3437
	RMSE	6.3196	5.1643	5.9748	6.0418	5.9704
P.563 [18]	PCC	0.7305	0.9964	0.9413	0.7752	0.7037
	SCC	0.8000	1.0000	0.8407	0.4638	0.6367
	RMSE	1.3415	1.3252	1.2310	1.2004	1.2650
AV-Linear	PCC	0.4520	0.9649	0.7718	0.0409	0.4431
	SCC	0.6000	1.0000	0.3143	-0.2571	0.3368
	RMSE	10.9449	10.7825	10.6525	10.6429	10.7430
AV-Minkowski	PCC	0.3032	0.9109	0.6881	-0.2842	0.3422
	SCC	0.6000	1.0000	0.1429	-0.2571	0.3143
	RMSE	2.3585	2.2612	2.0770	2.1419	2.1973
AV-Power	PCC	-0.7187	-0.6990	-0.5271	-0.8383	-0.6616
	SCC	-0.6000	-0.5000	-0.6000	-0.7714	-0.6075
	RMSE	23.7961	24.0376	24.2251	24.0783	24.0462
NAViDAd	PCC	0.8879	0.9252	0.8794	0.9044	0.8819
	SCC	0.9200	1.0000	0.8629	0.9086	0.8904
	RMSE	0.5764	0.6125	0.5406	0.6013	0.5850

This database is composed of 420 sequences, with video components at a Full HD resolution (1920×1080 , 4:2:0, 24 fps). The database contains a diverse content, which includes action, documentary, video games, and sports videos. The source videos are processed with 7 different network conditions and with 4 bitrate adaptation strategies. No audio degradations were included in the database. A total of 65 subjects rated the overall audiovisual quality of the sequences.

Table III shows the LiveNetflix-II results. Since the database does not include audio degradations, we only show the results for the FR and NR visual quality metrics. It is worth pointing out that, as there are no audio degradations, the

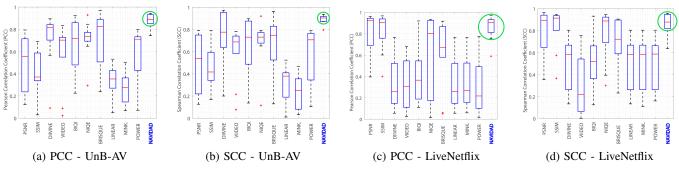


Fig. 3. Bar plot of PCC and SCC gathered from testing the litarature metrics on the UnB-AV database and the Live Netflix database.

TABLE III PCC, SCC, AND RMSE GATHERED FROM TESTING THE FR AND NR VISUAL QUALITY METRICS ON THE LIVENETFLIX-II DATABASE.

-	Full-Reference		No-Reference					
Measure	PSNR	SSIM	DIIVINE	VIIDEO	BIQI	NIQE	BRISQUE	NAViDAd
PCC	0.6981	0.7333	-0.8364	-0.6598	-0.4263	-0.7550	-0.7271	0.8611
SCC	0.6911	0.7123	-0.8106	-0.7153	-0.4724	-0.7701	-0.7115	0.8599
RMSE	32.2445	2.3024	2.6126	2.5265	38.3084	3.8324	56.2907	0.5929

visual quality metrics have a clear advantage in this database. Unfortunately, up to our knowledge, there are no audiovisual quality databases that include both audio and video degradations. Nevertheless, the proposed method performed better than the visual (and audio-visual) quality metrics, with correlation coefficients of around 0.86. Figures 3 (c) and (d) present the bar plots for the average PCC and SCC values (over the 10 folds) for the tested metrics. As with the UnB-AV database, results show that the NAViDAd correlation values varied very little across the simulations, which shows the consistency of the metric. We believe NAViDAd can be used in real-time streaming environments, specially in cases where audio distortions are expected to happen.

IV. CONCLUSIONS

In this work, we proposed a no-reference audio-visual quality metric, which is base on a deep autoencoder technique. The proposed model used a set of audio and video features to estimate the overall audiovisual quality. The model is formed by a 2-layer framework that includes a deep autoencoder layer and a classification layer. These two layers are stacked and trained to build the deep neural network model. These feature sets were passed to a two-layer autoencoder that produced a set of features with lower dimension. Then, a classification function mapped these features into subjective scores. Results showed that the proposed approach estimates the perceived quality of audio-visual sequences with good accuracy. The model presented a significant advantage when compared to several video, audio, and audio-visual objective metrics from the literature. The model was also tested on a external database and also presented a good performance. We believe better results can be achieved with the refinement of the network parameters. In addition, further tests (e.g. an ablation study) can be made in order to determine the relative importance of audio and video features in the proposed system.

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