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# Blind Quality Estimation for Corrupted Source Signals Based on A-Posteriori Probabilities

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**Abstract** — A novel approach is presented for assessing the quality of transmission systems, comprising quantized source signals and APP source decoders, via Monte-Carlo simulation. A-posteriori probabilities are exploited in order to obtain an unbiased estimate of both the symbol error probability and the expected distortion for the transmission system; knowledge of the transmitted source signal is not necessary. Compared to the conventional method this blind quality estimation has a smaller estimation variance.

## SUMMARY

The bit error rate estimation based on a-posteriori probabilities (APPs) was shown to be superior to the conventional one based on hard decisions [1]. In this paper this method is extended to symbol error rate (SER) and distortion estimation.

Let us assume a simplified transmission model, where a real-valued source signal<sup>1</sup>  $U$  is quantized to quantization indices  $I$ ,  $i \in \mathcal{I}$ , which are transmitted over a communication channel. Based on the channel observations  $Y$  the receiver generates APPs  $Pr(I=i|\mathbf{y})$  [2], which are exploited to obtain estimates  $\hat{U}$  and  $\hat{I}$  of  $U$  and  $I$ .

Typically quality evaluation via Monte-Carlo simulation is based on a comparison of the transmitted source data  $(u, i)$  to their reconstructed versions  $(\hat{u}, \hat{i})$  with respect to appropriate quality measures, such as the symbol error rate  $P_s$  or distortion  $D$ , defined as  $P_s := Pr(I \neq \hat{I})$  and  $D := E\{(U - \hat{U})^2\}$ . Accordingly the conventional approach for measuring  $P_s$  and  $D$  can be summarized as follows:

**Method H:** Let us define the *hard SER sample*  $z_H := Pr(I \neq \hat{I} | I = i, \hat{I} = \hat{i})$ ,  $z_H \in [0, 1]$ , indicating whether a symbol error occurred or not, and let us define the *hard distortion sample*  $d_H := (u - \hat{u})^2$ , taking into account the contribution to the reconstruction error due to estimate  $\hat{u}$ . For a transmission of  $K$  source symbols, the corresponding quality samples can be used to compute the *hard SER estimate*  $z_H^{(K)}$  and *hard distortion estimate*  $d_H^{(K)}$  as

$$z_H^{(K)} := \frac{1}{K} \sum_{k=1}^K z_{Hk} \quad \text{and} \quad d_H^{(K)} := \frac{1}{K} \sum_{k=1}^K d_{Hk}.$$

Obviously,  $z_H^{(K)}$  and  $d_H^{(K)}$  rely on the knowledge of  $u$  and  $i$ , from which it follows that the conventional Method H is not suitable for application in practical transmission systems.

Thus, we consider now the case, where knowledge of  $u$  and  $i$  is not available. I.e., only the source statistics, the estimates  $\hat{u}$  and  $\hat{i}$ , and the set of APPs  $p_{A_k} = \{Pr(I_k = i|\mathbf{y}) | i \in \mathcal{I}\}$  may be used for quality estimation. These restrictions lead to a novel approach for the evaluation of  $P_s$  and  $D$ , referred to as *Method S* in the following:

**Method S:** We define the *soft SER sample* as  $z_S := Pr(I \neq \hat{I} | I = \hat{i}, P_A = p_A)$ , which can be computed as  $z_S = 1 - Pr(I = \hat{i} | \mathbf{y})$ , and we define the *soft distortion sample*  $d_S := E\{(U - \hat{U})^2 | P_A = p_A\}$ , which is given by the *a-posteriori* expectation of the mean-squared error according to  $d_S = \sum_{i \in \mathcal{I}} E\{(u - \hat{u})^2 | I = i\} \cdot Pr(I = i | \mathbf{y})$  for

a given  $\hat{u}$ . Considering again the transmission of  $K$  source symbols, the *soft SER estimate*  $z_S^{(K)}$  and the *soft distortion estimate*  $d_S^{(K)}$  for Method S are given by

$$z_S^{(K)} := \frac{1}{K} \sum_{k=1}^K z_{Sk} \quad \text{and} \quad d_S^{(K)} := \frac{1}{K} \sum_{k=1}^K d_{Sk}.$$

For comparison of both methods we regard the hard and the soft SER and distortion samples as random variables  $Z_H$ ,  $Z_S$  and  $D_H$ ,  $D_S$ . From their definitions and since the estimates are sample means, it follows that  $\mu_Z = E\{Z_H\} = E\{Z_S\} = P_s$  and  $\mu_D = E\{D_H\} = E\{D_S\} = D$ . Thus, both estimates are unbiased for both the SER and the distortion estimation.

An appropriate figure-of-merit is the estimation variance. The variance of the hard SER sample  $Z_H$  can be written as

$$\sigma_{Z_H}^2 = E\{Z_H^2\} - \mu_Z^2 = E\{Z_H\} - \mu_Z^2 = \mu_Z(1 - \mu_Z), \quad (1)$$

where the identity  $Z_H^2 = Z_H$  was applied. The variance of the soft SER sample  $Z_S$ , respectively, can be written as

$$\sigma_{Z_S}^2 = E\{Z_S^2\} - \mu_Z^2 \leq E\{Z_S\} - \mu_Z^2 = \mu_Z(1 - \mu_Z) \quad (2)$$

and is upper bounded by  $\mu_Z(1 - \mu_Z)$ , since from  $Z_S \in [0, 1]$  it follows that  $Z_S^2 \leq Z_S$ , and thus  $E\{Z_S^2\} \leq E\{Z_S\}$ . Equality holds for the uninteresting cases  $Z_S = 0$  and  $Z_S = 1$  ( $\sigma_{Z_S}^2 = 0$ ). For all other cases we have a lower bound on the ratio of variances  $\sigma_{Z_H}^2$  and  $\sigma_{Z_S}^2$  of the SER samples:

$$\frac{\sigma_{Z_H}^2}{\sigma_{Z_S}^2} > 1, \quad (3)$$

resulting directly from (1) and (2).

A similar bound on the ratio of variances  $\sigma_{D_H}^2$  and  $\sigma_{D_S}^2$  of the distortion samples can be derived by applying Jensen's inequality to the a-posteriori expectation of  $D_H^2$ :

$$E\{D_H^2 | P_A = p_A\} \geq E\{D_H | P_A = p_A\}^2 = D_S^2, \quad (4)$$

where the identity  $D_H = (u - \hat{u})^2$  and the definition of the soft distortion sample  $D_S$  is exploited. It follows from (4) that  $E\{D_H^2\} \geq E\{D_S^2\}$ , where again equality holds for  $\sigma_{D_S}^2 = 0$  (see above), and otherwise

$$\frac{\sigma_{D_H}^2}{\sigma_{D_S}^2} > 1, \quad (5)$$

which represents a lower bound on the ratio of variances  $\sigma_{D_H}^2$  and  $\sigma_{D_S}^2$  of the distortion samples.

The bounds in (3) and (5) prove that the hard SER sample as well as the hard distortion sample have always (except for  $P_s = 0$ ) a larger variance than the soft SER sample and the soft distortion sample, respectively. This reveals the superiority of the proposed Method S to the conventional Method H. In numerical results for Gauss-Markov sources the gain with respect to the estimation variance turned out to be even larger than predicted.

## REFERENCES

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<sup>1</sup>Random variables are denoted by uppercase letters, their realizations by lowercase letters. Indices are omitted for convenience, whenever this can be done without ambiguity.