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Nuclear Counting Filter Based on a Centered Skellam Test and a Double Exponential Smoothing

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Abstract— Online nuclear counting represents a challenge due to the stochastic nature of radioactivity. The count data have to be filtered in order to provide a precise and accurate estimation of the count rate, this with a response time compatible with the application in view.

An innovative filter is presented in this paper addressing this issue. It is a nonlinear filter based on a Centered Skellam Test (CST) giving a local maximum likelihood estimation of the signal based on a Poisson distribution assumption. This nonlinear approach allows to smooth the counting signal while maintaining a fast response when brutal change activity occur. The filter has been improved by the implementation of a Brown’s double Exponential Smoothing (BES).

The filter has been validated and compared to other state of the art smoothing filters. The CST-BES filter shows a significant improvement compared to all tested smoothing filters.

Index Terms— Nuclear, Measurement, Filter, Signal Processing.

I. INTRODUCTION

Nuclear counting signal could be considered as a time-dependent random variable $x^t$ taking its value into a Poisson distribution $\mathcal{P}$ with a parameter $\lambda^t$ (see Eq. 1) [1]. The expected count number $\lambda^t$ has to be estimated at each time $t$.

$$x^t \sim \mathcal{P}(\lambda^t)$$  

(1)

If the activity is constant ($\lambda^t = \gamma_0$), the maximum likelihood estimation of $\lambda^t$ is given by the average of the measured data $x^t$. In the case of time varying activity the count rate could be estimated by a Single Moving Average (SMA) or an Exponential Moving Average (EMA). The tradeoff between precision and response time is set by the preset time $\theta$ of the SMA estimate $\bar{x}_{SMA}^t$ and the forgetting factor $\alpha$ of the EMA estimates $\bar{x}_{EMA}^t$ such as presented in Eq. 2-3.

$$\bar{x}_{SMA}^t = \frac{1}{\theta} \sum_{\ell=0}^{t} x^\ell = \bar{x}_{SMA}^{t-1} + \frac{1}{\theta} (x^t - x^{t-\theta})$$  

(2)

$$\bar{x}_{EMA}^t = ax^t + (1-a)\bar{x}_{EMA}^{t-1}$$  

(3)

These linear low-pass filters are commonly implemented into ratemeters [2-6]. Nonlinear filters are also in development since many years to address the issue of nuclear counting smoothing [7-10]. A hypothesis test is performed in order to determine where the activity is changing and then make a decision on the time constant to set into the moving average filter. For this purpose, it has been proven that the Centered Skellam Test (CST) is an efficient filter to detect changes into the Poisson counting [11-12]. By its simplicity, it could easily be implemented into embedded electronics assuming real time processing.

The data are record in a buffer with a size $M$. The integration time constant $\theta^t$ is set following the result of the test as detailed in Eq. 4-5 where $L$ is the vector of results from the test and $Q$ is the parameter of the filter. The hypothesis test carried on the Poisson statistics assumption and a property associated with the Skellam distribution. Further details are mentioned in [12].

$$L = \arg_{t} \left\{ \sum_{i=0}^{t-\theta} x^t \theta_{i-1} - \frac{\sum_{i=0}^{t-\theta} x^t \theta_{i}}{i} - Q \sqrt{\frac{\sum_{i=0}^{t-\theta} x^t \theta_{i}}{(\theta_{i-1})^2} + \frac{\sum_{i=0}^{t-\theta} x^t \theta_{i}}{i^2}} \right\} > 0$$  

(4)

$$\theta^t = \theta^{t-1} - \dim(L) + 1$$  

(5)

At the current time $t$, the count rate is estimated as presented in Eq. 6 by a SMA filter where $\theta^t$ is the time-dependent integration time. In the absence of detection, the value $\dim(L)$ is equal to zero, and then the time constant $\theta^t$ is extended. When a change in $\lambda^t$ is detected the value of $\dim(L)$ provides an indication about the credibility of the detection, and then the time constant $\theta^t$ is reduced.
\[ \hat{\lambda}_{\text{CST}} = \frac{1}{\theta t} \sum_{t-g}^{t} x^t \]  

(6)

It has been proven in [12] that the CST estimates the counting signal with a better performance compared to SMA or EMA filters at count rate above 1 cps. An improvement is considered in this new work by the adding of recursive smoothing on the CST output estimates. The method is details on next section and the filter will be benchmark with other filters in order to highlight its advantages and limitations.

II. METHOD

Based on the original CST filter presented and pre-validated in [12], some optimizations are now presented in this chapter. Two figures of merits \( FOM_1 \) and \( FOM_2 \) are used as performance indicators.

The first one \( FOM_1(\hat{\lambda}_X/\hat{\lambda}_Y) \) indicated the behavior of the filter \( X \) in comparison with the filter \( Y \) when a brutal change in the signal intensity occurs. It is minimized when the precision of the measurement \( P(\hat{\lambda}_X) \) and the response times to brutal change \( \tau_i(\hat{\lambda}_X) \) and \( \tau_d(\hat{\lambda}_X) \) are the lowest (see Eq. 8). The relative standard deviation \( P(\hat{\lambda}_X) \) is defined as the ratio between the observed standard deviation \( \sigma(\hat{\lambda}_X) \) and the empirical mean \( \bar{\lambda}_X \) of the output signal during a constant activity such as: \( \hat{\lambda}^t = \gamma_0 \) (see Eq. 7). The response time to an increase step \( \tau_i(\hat{\lambda}_X) \) and to a decrease step \( \tau_d(\hat{\lambda}_X) \) are defined as times required to achieve the final value \( \lambda^t = \gamma_1 \) starting from the original count rate value \( \lambda^t = \gamma_0 \) (details on their calculation are mentioned in [12]). The value of \( \gamma_1 \) is set as a function of the Signal to Noise Ratio SNR such as: \( \gamma_1 = \gamma_0 + SNR \sqrt{\gamma_0} \).

\[ P(\hat{\lambda}_X) = \frac{\sigma(\hat{\lambda}_X)}{\bar{\lambda}_X} \]  

(7)

\[ FOM_1(\hat{\lambda}_X/\hat{\lambda}_Y) = \frac{P(\hat{\lambda}_X)(\tau_i(\hat{\lambda}_X) + \tau_d(\hat{\lambda}_X))}{P(\hat{\lambda}_Y)(\tau_i(\hat{\lambda}_Y) + \tau_d(\hat{\lambda}_Y))} \]  

(8)

The second figure of merit \( FOM_2(\hat{\lambda}_X/\hat{\lambda}_Y) \) has been built in order to account for the accuracy of the filter output with regards to the trend phenomenon. The accuracy is defined as the relative error of the output estimates compared to the expected value which following a trend such as: \( \hat{\lambda}^t = \gamma_0 + C t \) where \( C \) is the slope of the trend.

\[ A(\hat{\lambda}_X) = \frac{|\hat{\lambda}_X - \hat{\lambda}^t|}{\hat{\lambda}^t} \]  

(9)

\[ FOM_2(\hat{\lambda}_X/\hat{\lambda}_Y) = \frac{P(\hat{\lambda}_X)A(\hat{\lambda}_X)}{P(\hat{\lambda}_Y)A(\hat{\lambda}_Y)} \]  

(10)

A FOM value equal or above to 1 means that the filter \( X \) gives no significant gain in comparison to the referenced filter \( Y \). A FOM below 1 quantifies the improvement ensured by the filter \( X \).

A. Optimization of the original CST filter

An implementation of the CST filter with an infinite buffer size is considered here, to get rid of any memory limitations. It has been shown in previews works [12] that the parameter \( Q \) of the CST filter found its optimal value between 1 and 2. A more accurate estimation has been calculated and presented into the Fig. 1. A value of 1.6 seems to give optimal performance whatever the activity characteristics are. This value will be considered in the following of the study.

The distribution of the integration time constant \( \theta t \) for different activity magnitudes is now studied. The Fig. 2 shows that the time constant is comprised between 1 and 150 with a mean value equal to 47. We can deduce that the buffer size of the system does not need to be higher than 200 to ensure a convenient operation of the filter. A buffer size equal to 200 will therefore be considered in this study.
B. The exponential smoothing

The output signal provided by CST filtering cannot be described by a determined distribution. The data have then to be processed by implementing a technique for smoothing this indeterminate series containing a potential trend. The Brown\'s double Exponential Smoothing (BES) is largely used in economics, and allows forecasting futures values and then anticipating a trend [13-14]. This approach has been chosen since it is a simple and then compatible with real time and embedded requirements of ratemeter. The value of the trend is estimated by the difference between a simple exponential smoothing \( \hat{\lambda}_{EMA} \) and a double exponential smoothing \( \hat{\lambda}_{DEMA} \) as presented in the regressive formulae Eq. 12-13. The time constant of exponential smoothing \( \alpha^t \) is set with regards to the value of the CST time constant \( \theta^t \) as presented in the Eq. 11, where \( \beta \) is the parameter of the exponential smoothing part. Thus, the time constants of the final filter will be linked together within the same meaning.

\[
\alpha^t = \left[ 1 - \exp\left( -\frac{1}{\beta (\theta^t - 1)} \right) \right] \quad (11)
\]

\[
\hat{\lambda}_{EMA}^t = \alpha^t \hat{\lambda}_{CST}^t + (1 - \alpha^t) \hat{\lambda}_{EMA}^{t-1} \quad (12)
\]

\[
\hat{\lambda}_{DEMA}^t = \alpha^t \hat{\lambda}_{EMA}^t + (1 - \alpha^t) \hat{\lambda}_{DEMA}^{t-1} \quad (13)
\]

The BES estimate at the current time \( t \) is given by the following formula:

\[
\hat{\lambda}_{BES}^t = 2 \hat{\lambda}_{EMA}^t - \hat{\lambda}_{DEMA}^t \quad (14)
\]

This estimate allows a nonbiased smoothing of the signal for constant or soft slope. When a brutal change and a hard trend slope occur, a lag time and then a lack of accuracy steel appear. In order to process in a better way the signal, an alternative estimation will be done when significant changes are detected. A Moving Average Convergence Divergence \( MACD^t \) indicator is defined as the difference between the EMA and DEMA estimates. The relative trending indicator \( \rho^t \) is adapted to our problematics by a specific normalization as presented in Eq. 15.

\[
\rho^t = \frac{\hat{\lambda}_{EMA}^t - \hat{\lambda}_{DEMA}^t}{\sqrt{\hat{\lambda}_{DEMA}^t}} \quad (15)
\]

The indicator \( \rho^t \) is used to detect the presence of a trend into the signal with regards to a threshold value \( \kappa \). The final estimation \( \hat{\lambda}_{CST}^t \), is calculated as presented in Eq. 15. If \( \theta^t \) exceeds the threshold \( \kappa \), the CST output is directly corrected by adding the trending correction. A gain in accuracy and response time is attempted by the lag time reduction. If \( \theta^t \) is under the threshold \( \kappa \), the BES estimate is applied ensuring a gain in precision.

\[
\begin{align*}
\hat{\lambda}_{CST}^t &= \hat{\lambda}_{CST}^t + \rho^t \sqrt{\hat{\lambda}_{DEMA}^t} \quad \forall \rho^t \geq \kappa \\
\hat{\lambda}_{CST}^t &= \hat{\lambda}_{EMA}^t + \rho^t \sqrt{\hat{\lambda}_{DEMA}^t} \quad \forall \rho^t < \kappa
\end{align*}
\]

Fig. 4 shows the probability distribution of the \( \rho^t \) indicator in the case of constant activity and in the case of a trend contained into the signal. The threshold \( \kappa \) has to be chosen to detect a trend with a maximum of true detections and a minimum of false detections. From the Fig. 4, a value of 0.3 has been chosen as optimal.

C. Optimization of the \( \beta \) parameter

The parameter \( \beta \) used into the exponential smoothers has also to be optimized. The figure of merits \( FOM_1 \) and \( FOM_2 \) are estimated as a function of the smoother parameter \( \beta \) for different level of activities \( y_0 \), relative slope \( C \) and signal to noise \( SNR \) values. Some of these curves are reported in Fig. 6-9 and a value equal to 1.5 can be considered as an optimal value maintaining a compromise between accuracy and precision.

Fig. 6. Figures of merit obtained as a function of the parameter \( \beta \) for a base line count rate \( y_0 = 10 \text{ cps} \) and for different value of the slope \( C \).
III. RESULTS

The filter CST* has been defined in the previous chapter and is now compared with some other smoothing filters implemented for nuclear counting. The studied filters are listed below:

- The SMA filter described in Eq. 2 with a time constant $\theta = 100$.
- The EMA filter described in Eq. 3 with a time constant $\alpha = 0.7$.
- The CST filter described in [12] and I. with a parameter $Q = 1.6$.

The figure of merit relatively to one obtained by SMA filter has been calculated as a function of the baseline signal amplitude $\gamma_0$. SMA is chosen as the reference filter because of its usual implementation in ratemeters. Fig. 10 shows the values of $FOM_1(\hat{\lambda}_X/\hat{\lambda}_{SMA})$ for a brutal change in amplitude corresponding to a $SNR = 10$. The CST filters exhibit better performances than SMA and EMA filters as already demonstrated in [12]. The CST* allows an improvement to be made in comparison to the original version of the CST filter when the activity falls bellow 10 cps. This gain at low count rate is also observed at lower SNR as seen in Fig.11 where the CST* maintains a lower FOM while CST has poorer performances in this case.

Fig. 12 shows the values of $FOM_2(\hat{\lambda}_X/\hat{\lambda}_{SMA})$ for a signal containing a trend with a slope $C=0.2$ and 0.3. As already seen with $FOM_1$, the performance below 10 cps are improved without any discrepancy observed at higher count rate.
Fig. 12. Figures of merit obtained as a function of the baseline count rate for a trend with a slope $C$ equal to 0.2.

Fig. 13. Figures of merit obtained as a function of the baseline count rate for a trend with a slope $C$ equal to 0.3.

The CST* version allows to process the counting signal on a larger range of count rate levels. The superiority of nonlinear filters as the CST* compared to traditional linear filtering is confirmed by those results.

As an illustration, temporal profiles are presented in the Fig. 14-16 in the case of very low count rates. In Fig. 13, the CST* exhibits a faster response to the increase step while smoothing the signal with a better precision. In Fig. 14, the trend is followed by the CST* without any lag times giving performance closed to CST ones. Finally a constant count rate is presented in the Fig. 15. The CST* provides a more precise estimate than the CST and exhibits a behavior closer to the highly destructive SMA low-pass filter.

Fig. 14. Example of temporal profiles obtained by SMA, EMA, CST and CST* for a brutal change set of with an SNR equal to 5 and a baseline count rate of 0.1

Fig. 15. Example of temporal profiles obtained by SMA, EMA, CST and CST* for a trend set of with an slope equal to 0.15 and a baseline count rate of 0.1

Fig. 16. Example of temporal profiles obtained by SMA, EMA, CST and CST* for a constant count rate of 0.1

IV. DISCUSSION

The nonlinear CST* filter has recently been tested for spectroscopy applications. It is applied to each energy channel $j$ of a temporal spectrum $\varphi_j$ as seen in Eq. 16 where $\psi_j$ is the output smoothed spectrum.

$$\forall j \in [1, n], \quad \psi_j = CST^*(\varphi_j)$$ (16)

Fig. 16 shows a row temporal gamma spectrum obtained during a clad failure experiment conducted at the OSIRIS testing reactor where an elementary integration time equal to 5 s is considered [15-16]. The large fluctuations induced difficulties for the fitting of the Gaussian peak required to quantify isotope activities. Fig. 15 shows the same temporal gamma spectrum after being processed by the CST* filter. The time preserving smoothing ensured by the CST* filter makes the analysis of the temporal spectrum easier.
suitable for the accurate and precise estimation of the count rate whatever the trend of the signal or any brutal change occurring. It is well suited for embedded and real time requirements associated with nuclear instruments.

V. CONCLUSION

The filter will also be implemented as a critical building block into current R&D projects such as an innovative compensation system addressing neutron detection [17-18] and into a system allowing moving sources to be detected by a sensor network [19-20]. The stability of the CST* filter with regards to drifts in the Poisson assumption (pile-up, correlation) will be studied in future works.

REFERENCES


