Boundary conditions for Young - van Vliet recursive filtering
Bill Triggs, Michaël Sdika

To cite this version:

HAL Id: inria-00548616
https://inria.hal.science/inria-00548616
Submitted on 20 Dec 2010

HAL is a multi-disciplinary open access archive for the deposit and dissemination of scientific research documents, whether they are published or not. The documents may come from teaching and research institutions in France or abroad, or from public or private research centers.

L’archive ouverte pluridisciplinaire HAL, est destinée au dépôt et à la diffusion de documents scientifiques de niveau recherche, publiés ou non, émanant des établissements d’enseignement et de recherche français ou étrangers, des laboratoires publics ou privés.
Boundary Conditions for Young–van Vliet Recursive Filtering

Bill Triggs, Member, IEEE, and Michael Sdika

Abstract—Young and van Vliet have designed computationally efficient methods for approximating Gaussian-based convolutions by running a recursive infinite-impulse-response (IIR) filter forward over the input signal, then running a second IIR filter backward over the first filter’s output. To transition between the two filters, they use a suboptimal heuristic that produces significant amplitude and phase distortion for all points within about three standard deviations of the right-hand boundary. In this correspondence, a simple linear transition rule that eliminates this distortion is derived.

Index Terms—Bidirectional recursive filtering, boundary conditions, Gaussian smoothing.

I. INTRODUCTION

Young and van Vliet (YvV) have developed computationally efficient forward–backward infinite-impulse-response (IIR) recursions for Gaussian filters [1], Gaussian derivatives [2], and Gabor filters [3]. (See [3] for their most recent design rules for Gaussians, and [4] for space-variant extensions and a performance comparison with other IIR Gaussian methods, including Deriche’s original method [5], [6].) Our approach also applies to the analogous recursion [7], [8] for B-spline–based signal processing. All of the YvV filters work forward, recursively calculating a running sum \( u_t \) as a linear combination of the input signal \( i_t \) and the \( k \) previous \( u \) values, then work backwards calculating a running sum \( v_t \) as a linear combination of \( u_t \) and the \( l \) previously calculated \( v \) values:

\[
\begin{align*}
  u_t &= i_t + \sum_{j=1}^{k} a_j u_{t-j}, \quad t = 1, \ldots, n \\
  v_t &= u_t + \sum_{j=1}^{l} b_j v_{t+j}, \quad t = n, \ldots, 1
\end{align*}
\]  

The final output is a scaled version of \( v_t \), and \( \{a_j, j=1, \ldots, k\} \) and \( \{b_j, j=1, \ldots, l\} \) are suitably chosen filter coefficients. For Gaussians, YvV chose \( k = l \) and \( \alpha = b \) [1], [3]. For other filters, \( i_t \) may be a linear transformation of the original input signal, e.g., a discrete derivative for derivative filters [1].

II. PROBLEM WITH HEURISTIC BOUNDARY CONDITIONS

To complete the specification (1), (2), we must fix initial conditions for \( u \) near \( t = 1 \) and for \( v \) near \( t = n \). For \( u \), we can pretend that the signal existed and took some nominal constant value \( i_1 \) (typically either 0 or \( i_1 \) ) for all \( t < 1 \). The correct initialization at \( t = 1 \) is then to set all \( u_{1-j, j=1, \ldots, k} \), to \( i_1(1 - \sum_{j=1}^{k} a_j) \), the steady-state response to an infinite stream of \( i_1 \)’s. Similarly, if we could suppose that for all \( t > n \), \( u_t \) took some constant value \( u_n \), the correct condition at \( t = n \) would be to set \( u_{n-j, j=1, \ldots, l} \) to \( u_n(1 - \sum_{j=1}^{l} b_j) \), the steady-state response to an infinite stream of \( u_n \)’s. YvV apparently do exactly this, with \( i_1 = i_1 \) and \( u_n = u_n \) (cf. [3, eq. (20) and (21)]). Another plausible choice for

Manuscript received July 19, 2004; revised June 13, 2005. This work was partly supported by European Research projects LAVA and aceMedia. The associate editor coordinating the review of this manuscript and approving it for publication was Prof. Tran D. Tran.

The authors are with GRAVIR-CNRS-INRIA, 38330 Montbonnot, France (e-mail: Bill.Triggs@inrialpes.fr; Michael.Sdika@inrialpes.fr; website: http://lear.inrialpes.fr/people/triggs/).

Digital Object Identifier 10.1109/TSP.2006.871980

Fig. 1. Impulse responses for points at various numbers of standard deviations from the right boundary, for a YvV Gaussian filter, with the standard YvV boundary heuristic \( u_\pm = u_n \) (top) and with our new boundary correction (14) (bottom). The corrected responses are much closer to the desired (truncated Gaussian) form.

\( u_\pm \) would be \( i_1(1 - \sum_{j=1}^{k} a_j) \), the steady state \( u \) resulting from an infinite stream of constant input values \( i_1 \) above \( t = n \) (typically, \( i_1 \) would be either \( i_n \) or 0).

Unfortunately, neither choice for \( u_\pm \) is correct. If the forward filter were continued to \( t \to n \) with input \( i_\pm \), its output would decay smoothly from \( u_n \) to \( i_\pm(1 - \sum_{j=1}^{l} b_j) \) within a few standard deviations, and the corresponding backward filter would take all elements of this “advance warning” signal into account when calculating its response. In fact, the forward–backward process only gives the correct overall impulse response when the full double recursion is run without truncation. Incorrect truncation causes significant amplitude and phase (geometric position) distortion for all points within about three standard deviations of the boundary. Fig. 1 illustrates the extent of the problem.

III. DERIVATION OF LINEAR BOUNDARY CORRECTION

To correct for the effects of truncation, we notionally extend the forward–backward pass to \( t \to \infty \) assuming a constant input value \( i_\pm \) above \( t = n \) and calculate the coefficients \( \{v_{n+j, j=1, \ldots, l}\} \) that would result from this infinite extension, given \( i_\pm \) and the final forward filter state \( \{v_{n-j, j=0, \ldots, l}\} \). The whole process is linear so the \( v \)’s must be linear functions of the \( u \)’s and \( i_\pm \). First suppose that \( i_\pm = 0 \). Gathering
the u’s, v’s into running k, l vectors u, v, the forward and backward passes become

\[
\begin{align*}
    u_t &= A_{t-1}u_{n}, \\
    v_t &= I_u u_t + B e_{t+1},
\end{align*}
\]

where \(A^k = A \cdot A \cdot \ldots \cdot A\) (k terms) is the \(k^{th}\) power of \(A\) and see (5), shown at the bottom of the page

\[
\begin{align*}
    u_t \equiv \begin{pmatrix} u_t \\ u_{t-k+1} \end{pmatrix} A \equiv \begin{pmatrix} a_1 & \cdots & a_{k-1} & a_k \\ 1 & \cdots & 0 & 0 \\ \vdots & \ddots & \vdots & \vdots \\ 0 & \cdots & 1 & 0 \end{pmatrix}
\end{align*}
\]

\[
\begin{align*}
    v_t \equiv \begin{pmatrix} v_t \\ v_{t-k+1} \end{pmatrix} B \equiv \begin{pmatrix} b_1 & \cdots & b_{k-1} & b_k \\ 1 & \cdots & 0 & 0 \\ \vdots & \ddots & \vdots & \vdots \\ 0 & \cdots & 1 & 0 \end{pmatrix}
\end{align*}
\]

IV. SUMMARY OF METHOD

In summary, Young and van Vliet recursive filters suffer from severe amplitude and phase distortion at the right boundary unless the backward running coefficients are initialized from the forward ones as follows, where \(M\) is given by (5), (8), (10) or (12) and (13):

\[
\begin{align*}
    u_+ &= \frac{1}{1-\sum_{j=1}^{k} a_j} v_+ + \sum_{j=1}^{k} a_j v_{n+k} + \sum_{j=1}^{k} a_j u_{n+k}.
\end{align*}
\]

An implementation for two-dimensional Gaussian image filtering is available on the author’s website. J.-M. Geusebroek has also incorporated the technique in his IIR filtering package, available from his website http://www.science.uva.nl/~mark/downloads.html.

ACKNOWLEDGMENT

The authors would like to thank J.-M. Geusebroek for pointing out an error in the submitted version of (14).

\[
\begin{align*}
    M &= \frac{1}{(1 + a_1 - a_2 + a_3)(1 - a_1 - a_2 - a_3)(1 + a_2 + (a_1 - a_3) a_3) \\
    &\quad \times \begin{pmatrix}
        -a_3 a_1 + 1 - a_2^2 - a_2 \\ a_3 a_1 + 1 - a_2^2 - a_2 \\ a_3 a_1 + a_2 + a_1^2 - a_2^2 \\ a_1 a_2 + a_3 a_2 - a_1 a_3 - a_3 a_2 + a_3 - a_3 a_2 + a_3 \\ a_3 (a_1 + a_3 a_2) \\ a_3 (a_1 + a_3 a_2) 
    \end{pmatrix}.
\end{align*}
\]
REFERENCES


