

# Digital filtering

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## ABSTRACT

Notes about digital (and frequency space) current filtering on particle-in-cell simulations.

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## 1. Current filtering

See ? for a general discussion about filters. Here we will mainly focus on linear filters.

### 1.1. Fourier-space filtering

Convolution of an image  $f$  with kernel  $w$  (also known as filter mask) to produce a new image  $g$  can be expressed as

$$g = f * w = \int_{-\infty}^{\infty} f(\tau)(t - \tau)d\tau. \quad (1)$$

Discrete direct convolution,  $g = f \otimes w$ , valid for images that are defined on a grid can be expressed as

$$g_{ij} = \sum_{k=-\frac{n}{2}}^{\frac{n}{2}-1} \sum_{l=-\frac{n}{2}}^{\frac{n}{2}-1} w_{kl} f_{i+k, j+l}, \quad (2)$$

for  $i, j = 1, \dots, n$  where  $(i + k)$  is replaced by  $(i + k \pm n)$ , if it falls outside the range from 1 to  $n$ . Similar transformation needs to be applied to  $(j + l)$ . This is the so-called cyclic (wrapped) direct convolution. In this case, the design of filters reduces to finding expressions for the kernel  $w$ .

Fourier transform of  $f$ ,  $g$ , and  $w$  satisfy

$$g_{kl}^* = w_{kl}^* f_{kl}^*, \quad (3)$$

for  $k, l = 1, \dots, n$ , provided that  $w$  and  $f$  are arrays of the same size. Therefore, filtering in the spatial domain can be expressed with a simple point-by-point multiplication in the frequency domain. Typically different sources claim that it is beneficial to use Fourier domain for filtering if the kernel is bigger than  $7 \times 7$ . This allows us to design filters in the frequency domains, finding expressions for the  $w^*$ .

Ideal low-pass filter is given by setting  $w_{kl}^*$  to zero beyond certain distance  $R$  from the origin, and otherwise to unity as

$$w_{kl}^* = \begin{cases} 1 & \text{if } k^2 + l^2 < R^2, \\ 0 & \text{otherwise,} \end{cases} \quad (4)$$

for  $k, l = -\frac{n}{2}, \dots, \frac{n}{2}-1$ . This will, however, lead to ringing, as is evident from the spatial-domain weights of  $w$  that have both negative and positive values.

A filter that minimizes ringing is a Gaussian

$$w_{ij} = \frac{1}{\sqrt{2\pi}\sigma^2} \exp\left\{\frac{-i^2 + j^2}{2\sigma^2}\right\}, \quad (5)$$

where  $\sigma$  is the standard deviation or a width of the filter in units of pixels. Fourier transform of the Gaussian weights can be expressed by

$$w_{kl}^* = \exp\left\{\frac{-k^2 + l^2}{2[n/(2\pi\sigma)]^2}\right\}, \quad (6)$$

that is a scaled version of the spatial domain Gaussian, but with a different spread.

In conclusion, a good filter should be resembling a Gaussian as close as possible.

### 1.2. Digital filtering

Digital filtering describes the process of filtering a quantity on the  $x$ -space (withouth any Fourier transforms to  $k$ -space). [Hamming1977](#) [Collatz1966](#) (Birdsall & Langdon 1985)

Given a grid quantity  $X_i \equiv X(i\Delta x)$  a simple digital filtering is performed by replacing

$$X_i \leftarrow \frac{WX_{i-1} + X_i + WX_{i+1}}{1 + 2W}, \quad (7)$$

where  $W$  is the filter weight. Caution is needed to make sure that points on the right-hand side are the original values.

Different compact filters can be obtained by selecting different weightings. A filter with  $W = \frac{1}{2}$  is called a binomial filter and has a positive smoothing function in  $k$ -space. Selecting  $W < 0$  produces a so-called compensation filter as it enhances the original signal.  $W = -\frac{1}{6}$  gives a compensation that cancels second-order attenuation of a  $W = \frac{1}{2}$  filter, i.e., applying a  $W = \frac{1}{2}$  filter followed by a  $W = -\frac{1}{6}$  produces a fourth-order attenuated filter corresponding to a broader stencil with weights of  $(1/16)[-1 \ 4 \ 10 \ 4 \ -1]$ .

All of the filtering operations above need a temporary scratch array of the quantity being processed. A scratch-memory-free alternatives can be derived by assuming multiple passes with different filters. Simplest one can be obtained by replacing  $X_i$  with a forward-pass of  $X_i + UX_{i+1}$

and a backward-pass of  $X_i + UX_{i-1}$ , for a multi-pass filter parameter  $U$ . This is equal to the one-pass filter for

$$W = \frac{U}{1 + U^2} \quad (8)$$

or

$$U = \frac{1}{2W} \pm \sqrt{\frac{1}{(2W)^2} - 1}. \quad (9)$$

For  $U$  to be real valued, we require  $-\frac{1}{2} \leq W \leq \frac{1}{2}$ . Binomial two-pass filter (equal to  $\frac{1}{4}[1 \ 2 \ 1]$ ;  $W = \frac{1}{2}$ ) corresponds to  $U = 1$  with a forward-pass of  $\frac{1}{2}[0 \ 1 \ 1]$  and a return-pass with  $\frac{1}{2}[1 \ 1 \ 0]$ . The corresponding two-pass compensator ( $W = -\frac{1}{6}$ ) has  $U = -3 + \sqrt{8} \approx -0.171573$  to be applied after the smoothing.

Two-dimensional filters can be generated by using two one-dimensional filters (shown here for the second-order Binomial filter  $B_2$ ) as

$$B_2^{(2D)} \equiv B_2 \otimes B_2^T = \frac{1}{4}[1 \ 2 \ 1] \otimes \frac{1}{4}\begin{bmatrix} 1 \\ 2 \\ 1 \end{bmatrix} = \frac{1}{16}\begin{bmatrix} 1 & 2 & 1 \\ 2 & 4 & 2 \\ 1 & 2 & 1 \end{bmatrix}. \quad (10)$$

A scratch-memory-free version also exists in a form of a four-pass filter (consisting of  $C_{\rightarrow} \equiv \frac{1}{2}[0 \ 1 \ 1]$  and  $C_{\leftarrow} \equiv \frac{1}{2}[1 \ 1 \ 0]$ ) as

$$B_2^{(2D)} = C_{\rightarrow} \otimes C_{\leftarrow} \otimes C_{\rightarrow}^T \otimes C_{\leftarrow}^T \quad (11)$$

$$= \frac{1}{2}\begin{bmatrix} 0 & 0 & 0 \\ 0 & 1 & 1 \\ 0 & 0 & 0 \end{bmatrix} \frac{1}{2}\begin{bmatrix} 0 & 0 & 0 \\ 1 & 1 & 0 \\ 0 & 0 & 0 \end{bmatrix} \frac{1}{2}\begin{bmatrix} 0 & 1 & 0 \\ 0 & 1 & 0 \\ 0 & 0 & 0 \end{bmatrix} \frac{1}{2}\begin{bmatrix} 0 & 0 & 0 \\ 0 & 1 & 0 \\ 0 & 1 & 0 \end{bmatrix}. \quad (12)$$

A possible choice for the compensator can be (20, -1, -1) or (36, -6, 1) **check Birdsall filter notation assumed to be (center, middle, corner)**

$$\begin{bmatrix} -1 & -1 & -1 \\ -1 & 20 & -1 \\ -1 & -1 & -1 \end{bmatrix} \quad \text{or} \quad \begin{bmatrix} 1 & -6 & 1 \\ -6 & 36 & -6 \\ 1 & -6 & 1 \end{bmatrix}. \quad (13)$$

Here, one should, however, note that scratch-free versions can not be vectorized, which means that for modern processing units are probably not the optimal choice.

From the numerical point-of-view the filtering is always a balance between number of passes and size of the filter stencil. It also matters in what order we are asking the variables from the array; a modern processor has a prefetcher that can try to guess our memory access patterns. A suitable selection for the 2D seems to be a two-pass filtering (up and down) with the compact 3-point binomial (requiring 3 values,  $X_{i-1}$ ,  $X_i$ , and  $X_{i+1}$ ). A modern prefetcher might even be able to cope with a one-pass compact Binomial (requiring 9 values in 2D).

Note also that thanks to the usage of tiles (with a length of 3 buffer zones for the current arrays), it can make sense to combine multiple compact Binomial passes into a fewer passes with an extended stencil (supporting a 6 order stencil with 7 [3+1+3] points with one MPI communication round).

### 1.3. Binomial filters

Binomial filters form a compact approximation for the (discretized) Gaussian.<sup>1</sup> **Hamming 1977** Binomial coefficients

<sup>1</sup> [http://www.cse.yorku.ca/~kosta/CompVis\\_Notes/binomial\\_filters.pdf](http://www.cse.yorku.ca/~kosta/CompVis_Notes/binomial_filters.pdf)

are given by

$$\binom{N}{n} = \frac{N!}{(N-n)!n!}, \quad (14)$$

where  $n = 0, \dots, N$  and  $N$  is the order of the binomial filter  $B_N$ . It forms a rapid approximation for the Gaussian distribution as  $\sigma \approx \sqrt{N}/2$ .<sup>2</sup> First few are given as

$$B_0 = [1] \quad (15)$$

$$B_1 = \frac{1}{2}[1 \ 1] \quad (16)$$

$$B_2 = \frac{1}{4}[1 \ 2 \ 1] \quad (17)$$

$$B_3 = \frac{1}{8}[1 \ 3 \ 3 \ 1] \quad (18)$$

$$B_4 = \frac{1}{16}[1 \ 4 \ 6 \ 4 \ 1] \quad (19)$$

$$B_5 = \frac{1}{32}[1 \ 5 \ 10 \ 10 \ 5 \ 1] \quad (20)$$

$$B_6 = \frac{1}{64}[1 \ 6 \ 15 \ 20 \ 15 \ 6 \ 1] \quad (21)$$

$$\vdots \quad (22)$$

The binomial filter coefficients are applied for a quantity  $X$  (defined on a grid as  $X_i \equiv X(i\Delta x)$  as

$$X_i^{f_2} = B_2\{X_i\} = \frac{1}{4}X_{i-1} + \frac{1}{2}X_i + \frac{1}{4}X_{i+1}, \quad (23)$$

where  $X_i^{f_2}$  the filtered quantity and  $B_2\{X\}$  denotes an application of a  $B_2$  filter on the original quantity. Applying higher-order binomial filters generalizes naturally from here.

Discrete-time Fourier transform of the binomial filter  $B_2 = \frac{1}{4}[1 \ 2 \ 1]$  is proportional to a single period of a cosine as

$$B_2^*[k] = \sum_n B_2[n] \cos[kn] = \frac{1}{2} + \frac{1}{2} \cos[k]. \quad (24)$$

For higher-order filters of rank  $N$  we similarly have

$$B_N^*[k] = \left( \frac{1}{2} + \frac{1}{2} \cos(k) \right)^N. \quad (25)$$

### 1.4. Temporal filters

Another possibility, besides usage of spatial filters, is to filter electric field in time. This is known as the Friedman filter, defined as

$$\mathbf{E}^{(n)} = \left(1 + \frac{\theta}{2}\right)\mathbf{E}^{(n)} - \left(1 - \frac{\theta}{2}\right)\mathbf{E}^{(n-1)} + \frac{1}{2}(1 - \theta)^2\mathbf{E}^{(n-2)}, \quad (26)$$

where  $\mathbf{E}^{(n-1)} = \mathbf{E}^{(n-2)} + \theta\mathbf{E}^{(n-3)}$ , and the filtering parameter  $\theta \in [0, 1]$ .

## References

Birdsall, C. & Langdon, A. 1985, Plasma physics via computer simulation, The Adam Hilger series on plasma physics (McGraw-Hill, New York)

<sup>2</sup> Alternatively, binomial filters correspond to the rows of the Pascal's triangle.