

We discuss a method to filter observed data,  $u_{\text{obs}}$ , in a Bayesian framework, where  $u_{\text{obs}}$  can be spatial, temporal, or higher dimensional data. We are also not limited to cases where  $u_{\text{obs}}$  is observed over a regular grid. Our filtered estimate,  $u_{\text{post}}$ , incorporates  $u_{\text{obs}}$  and prior knowledge of the statistical properties of the underlying signal. In this sense, this work is closely related to Kriging, Kalman filtering, and, more generally, Gaussian process regression (e.g Rasmussen and Williams, 2006) . We constrain  $u_{\text{post}}$  with the observation equation

$$u_{\text{post}} = u_{\text{obs}} + \epsilon, \quad \epsilon \sim \mathcal{N}(0, \mathbf{C}_{\text{obs}}), \quad (1)$$

and the prior model

$$u_{\text{prior}} \sim \mathcal{N}(0, \mathbf{C}_{\text{prior}}), \quad (2)$$

where  $\epsilon$  and  $u_{\text{prior}}$  are considered to be Gaussian processes with zero mean and covariances  $\mathbf{C}_{\text{obs}}$  and  $\mathbf{C}_{\text{prior}}$  respectively. The solution for  $u_{\text{post}}$  minimizes the objective function

$$\|u_{\text{post}} - u_{\text{obs}}\|_{\mathbf{C}_{\text{obs}}}^2 + \|u_{\text{post}}\|_{\mathbf{C}_{\text{prior}}}^2 \quad (3)$$

and is itself a Gaussian process with a distribution described by

$$u_{\text{post}} \sim \mathcal{N}(\bar{u}_{\text{post}}, \mathbf{C}_{\text{post}}). \quad (4)$$

We use  $\bar{u}_{\text{post}}$  and  $\mathbf{C}_{\text{post}}$  to denote the mean and covariance of  $u_{\text{post}}$  respectively. Using Bayesian linear regression (e.g. Tarantola, 2005) these values are found to be

$$\begin{aligned} \bar{u}_{\text{post}} &= (\mathbf{C}_{\text{obs}}^{-1} + \mathbf{C}_{\text{prior}}^{-1})^{-1} \mathbf{C}_{\text{obs}}^{-1} u_{\text{obs}} \\ \mathbf{C}_{\text{post}} &= (\mathbf{C}_{\text{obs}}^{-1} + \mathbf{C}_{\text{prior}}^{-1})^{-1}. \end{aligned} \quad (5)$$

$\mathbf{C}_{\text{obs}}$  is presumably well known, while  $\mathbf{C}_{\text{prior}}$  needs to be chosen based on an understanding of the underlying signal which we are trying to estimate. With a judicious choice of  $\mathbf{C}_{\text{prior}}$ , eq. (5) can be made equivalent to several well established smoothing methods. In particular, we demonstrate how eq. (5) can be viewed as a low-pass filter with a well defined cutoff frequency. This is first demonstrated for filtering one-dimensional data, and the extension to higher dimensions follows naturally.

## One-dimensional smoothing

For one-dimensional data we consider a prior which can be stated implicitly as

$$\mathbf{D}_n u_{\text{prior}} = q, \quad q \sim \mathcal{N}(0, \lambda^2), \quad (6)$$

where  $D_n$  is an  $n^{\text{th}}$  order differentiation matrix, and  $q$  is white noise with constant variance  $\lambda^2$ . If we momentarily ignore the fact that  $D_n$  is not invertible then we can explicitly write our prior covariance as

$$\mathbf{C}_{\text{prior}} = \lambda^2 (\mathbf{D}_n^T \mathbf{D}_n)^{-1}, \quad (7)$$

and the filtered mean and covariance are described by

$$\begin{aligned} \bar{u}_{\text{post}} &= (\mathbf{C}_{\text{obs}}^{-1} + \frac{1}{\lambda^2} \mathbf{D}_n^T \mathbf{D}_n)^{-1} \mathbf{C}_{\text{obs}}^{-1} u_{\text{obs}} \\ \mathbf{C}_{\text{post}} &= (\mathbf{C}_{\text{obs}}^{-1} + \frac{1}{\lambda^2} \mathbf{D}_n^T \mathbf{D}_n)^{-1}. \end{aligned} \quad (8)$$

This filtered solution is closely tied to several well established methods of smoothing. For example, Kimeldorf and Wahba (1970) pointed out that eq. (8) is equivalent to a smoothing spline. This can be seen by ignoring data uncertainty (i.e.  $\mathbf{C}_{\text{obs}} = \mathbf{I}$ ) and noting from eq. (3) that  $\bar{u}_{\text{prior}}$  minimizes

$$\|u_{\text{obs}} - \bar{u}_{\text{prior}}\|_2^2 + \|D_n \bar{u}_{\text{prior}}\|_2^2, \quad (9)$$

which is a discretized form of the definition of a smoothing spline (e.g De Boor, 1978). As another example, we can loosely interpret our prior model in eq. (6) as Brownian motion or integrated Brownian motion when  $n = 1$  and  $n = 2$ , respectively. From this perspective, it is possible to see the connection between eq. (8) and Kalman filters, where these stochastic models are commonly used to describe time-dependent signals (e.g. Segall and Mathews, 1997).

One challenge that is common to each of these smoothing methods is selecting the hyperparameter  $\lambda$ . There are numerous ways that this could be accomplished. For example, one could use cross-validation, a trade-off

curve, or simply vary  $\lambda$  until  $\bar{u}_{\text{post}}$  looks appropriate when compared to  $u_{\text{obs}}$ . There is merit to using an entirely objective approach such as cross-validation, and this would be appropriate if the researcher has no prior knowledge of the signal's characteristic wavelength. Otherwise, it may be better to chose  $\lambda$  such that eq. (8) damps out all the high frequency oscillations which are known to be noise. We elaborate on this point by demonstrating that eq. (8) can also be viewed as a low-pass filter with a cutoff frequency determined by  $\lambda$ .

We wish to transform  $\bar{u}_{\text{post}}$  in eq. (8) to the frequency domain and in order to do so we assume that  $u_{\text{obs}}$  has a constant sampling rate. We also require that  $\epsilon$  and  $u_{\text{prior}}$  are stationary stochastic processes (i.e. their statistical properties are invariant to time shifts). We require them to be stationary so that matrix multiplication by  $\mathbf{C}_{\text{obs}}$  and  $\mathbf{C}_{\text{prior}}$  can be viewed as convolution in the time domain and thus multiplication in the frequency domain. We then consider the simplifying case where  $\epsilon$  is white noise with constant variance  $\sigma^2$  (i.e.  $\mathbf{C}_{\text{obs}} = \sigma^2 \mathbf{I}$ ), and  $\mathbf{D}_n$  is the periodic spectral differentiation matrix (e.g. Trefethen, 2000). Under a discrete Fourier transform,  $\mathbf{D}_n$  has the properties

$$\mathcal{F}[\mathbf{D}_n f] = (2\pi i \omega)^n \hat{f} \quad (10)$$

and

$$\mathcal{F}[\mathbf{D}_n^T f] = (-2\pi i \omega)^n \hat{f}, \quad (11)$$

where  $\omega$  is the frequency domain variable,  $f$  is an arbitrary vector and  $\hat{f}$  is its discrete Fourier transform. The discrete Fourier transform of  $\bar{u}_{\text{post}}$  is then

$$\hat{u}_{\text{post}}(\omega) = \frac{\frac{1}{\sigma^2}}{\frac{1}{\sigma^2} + \frac{(2\pi\omega)^{2n}}{\lambda^2}} \hat{u}_{\text{obs}}(\omega). \quad (12)$$

We make the change of variables

$$\lambda^2 = (2\pi\omega_c)^{2N} \sigma^2 \quad (13)$$

which changes the hyperparameter from  $\lambda$  to  $\omega_c$ . The reason for this change of variables becomes apparent when we simplify eq. (12) to

$$\hat{u}_{\text{post}}(\omega) = \frac{1}{1 + \left(\frac{\omega}{\omega_c}\right)^{2N}} \hat{u}_{\text{obs}}(\omega). \quad (14)$$

We can recognize eq. (14) as an  $n$ 'th order low-pass Butterworth filter with cut-off frequency  $\omega_c$ . In the limit as  $N \rightarrow \infty$  eq. (14) becomes an ideal low-pass filter which removes all frequencies above  $\omega_c$  and leaves lower frequencies unaltered. Of course, an ideal low-pass filter is often undesirable because it will tend to produce ringing artifacts in the filtered solutions. When modeling  $u_{\text{prior}}$  as Brownian motion or integrated Brownian motion, where  $n = 1$  and  $n = 2$  respectively, the frequency response is tapered across  $\omega_c$ , which ameliorates ringing in the filtered solution.

By demonstrating that eq. (8) can be made equivalent to a low-pass filter, it may not be clear why we would use eq. (8) when it is far more efficient to filter in the frequency domain through the Fast Fourier Transform. In order to make use of the Fast Fourier Transform, the observations must be made at a constant sampling rate and the observation noise must be white with constant variance. In contrast, these conditions do not need to be met in order to evaluate eq. (8). The question is then whether eq. (8) still effectively acts as a low-pass filter when the idealized conditions are not met.

We answer this question with two demonstration. In the first demonstration we generate XXX samples of  $u_{\text{obs}}(t)$  with a constant sampling rate over the interval  $0 < t < 1$ . Our samples of  $u_{\text{obs}}(t)$  are white noise where the variances for each observation are randomly selected from a uniform distribution ranging from 0.1 to 10. In our second demonstration we generate XXX samples of  $u_{\text{obs}}(t)$  at values of  $t$  which have a uniform random distribution ranging from 0 to 1. the values for  $u_{\text{obs}}(t)$  are again white noise but we use a constant variance of 1. In both demonstrations,  $u_{\text{prior}}$  is treated as integrated Brownian motion (i.e.  $N = 2$ ). We compute  $\mathbf{D}_2$  with a first-order accurate finite difference scheme. We then compute  $\bar{u}_{\text{post}}$  and  $\mathbf{C}_{\text{post}}$  from eq. (8) and select  $\lambda^2$  to be consistent with a cut-off frequency of XXX. For the first demonstration, where the variance is not constant, we modify eq. 13 so that  $\lambda^2$  is in terms of a characteristic variance,  $\bar{\sigma}^2$ , so that

$$\lambda^2 = (2\pi\omega_c)^{2N} \bar{\sigma}^2 \quad (15)$$

where

$$\frac{1}{\bar{\sigma}^2} = \frac{1}{P} \text{tr}(\mathbf{C}_{\text{obs}}^{-1}) \quad (16)$$

and  $P$  is the number of observations. We show the filtered time series and covariances at select positions in figures X. In order to assess whether the filtered solution is acting as a low-pass filter we make use of the Wiener-Khinchin theorem, which states that the power spectral density and autocovariance of a stationary

random process are Fourier transform pairs. We would then expect each row of the covariance matrix to resemble

$$c(t) = \mathcal{F}^{-1} \left[ \left| \frac{1}{1 + \left( \frac{\omega}{\omega_c} \right)^{2N}} \right|^2 \right] \quad (17)$$

after an appropriate time shift.

then compare the inverse Fourier transform of the squared frequency response in eq. 14 to the covariances for our demonstrations. If the two covariance functions resemble each other then we can conclude that eq.

We assess whether eq. 8

$u_{\text{obs}}(t) \sim \mathcal{N}(0, 1)$ . XXX samples of observations with uniform spacing from 0 to 1. Each o create XXX samples of white noise , each observations made with a uniform random distribution over the time interval (0,1). The variance of the observations are also randomly chosen from a uniform distribution over the interval (0.1,10). We use integrated Brownian motion for our prior model (i.e.  $N = 1$ ), and  $D_1$  is a first-order accurate finite difference operator (e.g. Fornberg).

The value of eq. (8) is that it offers a way to perform a low-pass filter when the conditions required for a discrete Fourier transform are not met. In the preceding Fourier analysis, we needed to make assumptions which are generally not true for real world applications. Namely, we assume This raises the question of whether eq. 8 will still effectively act as a low-pass filter when the idealized conditions are not met. In this example, we use observations made with a uniform random distribution over the time interval (0,1). The variance of the observations are also randomly chosen from a uniform distribution over the interval (0.1,10). We use integrated Brownian motion for our prior model (i.e.  $N = 1$ ), and  $D_1$  is a first-order accurate finite difference operator (e.g. Fornberg).

For example, if  $u_{\text{obs}}$  has an irregular sample rate or if the observation noise does not have a constant variance then we can no longer work in the frequency domain but we can still evaluate eq. (8). We can also still use eq. (13) as an intuitive guide for choosing  $\lambda$ . If the observation noise does not have a constant variance then  $\sigma$  in eq. (13) needs to be replaced with a characteristic variance,  $\bar{\sigma}^2$ , which we choose to be

$$\frac{1}{\bar{\sigma}^2} = \frac{1}{P} \text{tr} (\mathbf{C}_{\text{obs}}^{-1}), \quad (18)$$

where  $P$  is the number of observations. We provide an example to demonstrate that eq. (8) effectively acts as a low-pass filter when the data is aperiodically sampled and the observation noise does not have a constant variance. We compare each row in the correlation matrix for this example, determined from eq. 8, to the autocorrelation predicted function

illustrate filtering irregularly sampled data time non-constant variance.

We find that an appropiare which we choose to be  $\bar{\sigma} = 1/(1/\sigma)_{\text{rms}}$ .

It is worth pointing out that the solution for  $u_{\text{smooth}}$  in eq. ?? is identical to the solution that would be obtained through Kalman filtering followed by smoothing. Kalman filtering and smoothing are recursive algorithms that operate along the time dimension. It is thus difficult to envision how a Kalman filter can be extended to smoothing data in higher dimensions. In contrast, extending eq. ?? to higher dimensions comes simply and naturally as we describe in the next section.

## Smoothing in higher dimensions

We expand our discussion from one-dimensional smoothing to smoothing in two-dimensional space. While the remainder of this paper discusses smoothing in two-dimensions, the extension to smoothing in higher dimensions will prove to be trivial. We now allow  $u_{\text{prior}}$  to have non-zero covariance along multiple dimensions. In other words, we now treat  $u_{\text{prior}}$  as a random field rather than a stochastic process. The smoothed solution that we seek is still given by eq. 5 and our task is once again to find an appropriate choice of  $\mathbf{C}_{\text{prior}}$ . Below we consider the case where the  $M$ 'th order Laplacian of  $u_{\text{prior}}$  is white noise. That is to say

$$\Delta_M u_{\text{prior}}(x_1, x_2) = q, \quad q \sim \mathcal{N}(0, \lambda^2) \quad (19)$$

where

$$\Delta_M = \frac{\partial^{2M}}{\partial x_1^{2M}} + \frac{\partial^{2M}}{\partial x_2^{2M}}. \quad (20)$$

The corresponding covariance matrix is then

$$\mathbf{C}_{\text{prior}} = \lambda^2 (\Delta_M^T \Delta_M)^{-1}. \quad (21)$$

We again assume that the observation noise is uncorrelated with constant variance  $\sigma^2$ . Using the change of variables

$$\lambda = (2\pi\omega_c)^{2M}\sigma \quad (22)$$

we obtain the solution

$$u_{\text{smooth}}(x_1, x_2) = \left( \mathbf{I} + \left( \frac{1}{2\pi\omega_c} \right)^{4M} \mathbf{\Delta}_M^T \mathbf{\Delta}_M \right)^{-1} u_{\text{obs}}(x_1, x_2), \quad (23)$$

which in the two-dimensional frequency domain is

$$\hat{u}_{\text{smooth}}(\omega_1, \omega_2) = \frac{1}{1 + \left( \frac{\omega_1^{2M} + \omega_2^{2M}}{\omega_c^{2M}} \right)^2} \hat{u}_{\text{obs}}(\omega_1, \omega_2). \quad (24)$$

The transfer function in eq. (24) can once again be recognized as a low-pass filter. Namely, in the limit as  $M \rightarrow \infty$  all the frequency pairs,  $(\omega_1, \omega_2)$ , where  $|\omega_1| > \omega_c$  or  $|\omega_2| > \omega_c$  are removed. That is, the transfer function becomes a two-dimensional box centered at  $(0, 0)$  with width  $2\omega_c$ . It is apparent from the transfer function why our prior model must be defined in terms of even order derivatives. If eq. (20) contained odd order derivatives, then all frequency pairs where  $\omega_1 = -\omega_2$  would be unattenuated, leaving undesirable artifacts in the smoothed solution.

We have demonstrated with eq. (?? and 23) that eq. 5 can effectively act as a low pass filter with a judicious choice of  $\mathbf{C}_{\text{prior}}$ . One may then wonder why eq. (?? and 23) would ever be used to smooth data when it is far more efficient to perform the equivalent operation through the Fast Fourier Transform domain. The power in eq. (?? and 23) resides in the fact that they makes no assumption about when or where the observations have been made. In order to filter data through the Fast Fourier Transform, the observations must be evenly spaced on a regular grid, which is often not the case for geophysical data. Indeed, the focus of this paper is to smooth data which has been observed at scattered locations. We can use eq. (?? and 23) regardless of where the observations were made provided that we are able to create the differentiation matrix. We use the recently developed Radial Basis Function Finite-Difference method to form the differentiation matrices and we describe the procedure in the follow section.

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