# Subpixel Point Spread Function Estimation from Two Photographs at Different Distances* 

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#### Abstract

In most digital cameras, and even in high-end digital single lens reflex cameras, the acquired images are sampled at rates below the Nyquist critical rate, causing aliasing effects. This work introduces an algorithm for the subpixel estimation of the point spread function (PSF) of a digital camera from aliased photographs. The numerical procedure simply uses two fronto-parallel photographs of any planar textured scene at different distances. The mathematical theory developed herein proves that the camera PSF can be derived from these two images, under reasonable conditions. Mathematical proofs supplemented by experimental evidence show the well-posedness of the problem and the convergence of the proposed algorithm to the camera in-focus PSF. An experimental comparison of the resulting PSF estimates shows that the proposed algorithm reaches the accuracy levels of the best nonblind state-of-the-art methods.


Key words. image blur, subpixel convolution kernel estimation, aliasing, inverse problems, camera quality assessment, point spread function, modulated transfer function

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1. Introduction. Light diffraction, lens aberrations, sensor averaging, and antialiasing filters are some of the inherent camera factors that unavoidably introduce blur in photographs. The blur that results from the combination of all these factors can be modeled locally as a convolution kernel known as point spread function (PSF), which corresponds to the space variant impulse response of the whole camera, including the sensor, before the final sampling.

The area enclosed by the first zero crossing of the PSF, usually called Airy pattern, is arguably the most reasonable characterization of the optical system resolution. Top camera/lens manufacturers use charts based on the PSF Fourier spectrum modulus (the modulated transfer function (MTF)) to describe their products. But accurate knowledge of the PSF is not limited to quality assessment of optical devices, and it proves to be extremely useful or even necessary for several image processing tasks such as deblurring [28], superresolution [29, 31], or shape from defocus [10].

[^0]In most typical digital cameras, both compact and high-end digital single lens reflex cameras, images are sampled at frequencies below the Nyquist critical rate. Consequently, only aliased versions of the camera PSF can be directly observed. Yet, to fully characterize the PSF, it is necessary to recover it at a subpixel resolution.

PSF estimation methods can be classified as nonblind or blind, depending on whether they use a snapshots of a specially designed calibration pattern. Blind approaches try to estimate the PSF from photographs from an unknown scene. They do assume, however, that the scene involved in the estimation follows some statistical model of sharp images or includes a significant number of geometric cues such as sharp edges. Most of these PSF approaches attempt to detect edges, which are modeled as pure step-edge functions convolved with the PSF kernel $[9,24,8,4]$. In this setting, the estimation is very ill-posed; to solve the inverse problem, the solution space has to be constrained by considering kernels with a parametric model or with strong regularity assumptions. Therefore, such blind estimation techniques do not lead to accurate PSF estimates and are normally an ingredient in image restoration problems, where precision is not the main objective. For this reason, accurate PSF estimation procedures rely on the use of specially designed calibration patterns. A local kernel estimation is performed by comparing the ideal calibration pattern to its photographs.

Several patterns have been used for PSF estimation, ranging from pin-hole, slanted-edge $[19,30,38,11]$, or arc-step-edge patterns $[21,20]$ to random noise images $[12,22,2,3,7]$. Until recently, even nonblind subpixel PSF estimation methods reported in the literature led to ill-posed inverse problems. The inversion required the imposition of simple PSF parametric models or other regularity or symmetry priors. In a recent work [14] we have shown that such a priori assumptions on the PSF are actually unnecessary and jeopardize the estimation accuracy. More precisely, by carefully modeling the image acquisition system, a calibration pattern made of a white noise realization is nearly optimal in terms of well-conditioning of the problem. This procedure leads to very accurate regularization-free subpixel PSF estimation.

The purpose of the present work is to explore the feasibility of obtaining accurate PSF estimates, while avoiding the explicit use of a calibration pattern. The motivation comes from the fact that, although very precise, the use of a calibration pattern can be sometimes tedious and impractical: these approaches rely on a careful setup, and the calibration grid has to be properly assembled where a good quality print is essential.

We show that, instead of using a photograph of a known calibration pattern, two photographs of the same scene acquired at different distances with fixed camera configuration are enough to recover a regularization-free subpixel PSF. The proposed acquisition procedure is simple and handy in comparison to a nonblind approach. Experimental evidence will show that the resulting estimates do not exhibit any significant accuracy loss compared to their best nonblind competitors. The choice of the photographed scene is important but not critical. For a wide range of everyday textured scenes, the acquired image pairs lead to well-posed inversions and highly accurate results.

This paper is written with a dual public in mind: mathematicians and/or image processing specialists. We have tried to define accurately all mathematical objects necessary to deal rigorously with image formation. An accurate formalism is needed to justify the somewhat intricate interlacement of sampling and convolution operations. This forces one to check on the compatibility of all function or distribution spaces to which the objects belong and to
verify that the formulas are mathematically consistent. Nevertheless, the application-oriented reader can skip the proofs and the functional space details at a first reading and simply focus on the standard image processing formalism and algorithms. Most proofs are placed at the end of the paper. A glossary is appended to display all notation in a single place.

The article is organized as follows: section 2 presents a mathematical model of the digital image acquisition system. This model is used in section 3, where it is shown that the camera PSF can be recovered from a pair of unknown scaled images. We define the notion of blur between such a pair of images, and we propose a method to perform its estimation. Then we prove that the camera PSF can be recovered from this interimage blur. Section 4 presents an algorithm that implements the complete PSF estimation procedure described in section 3. In section 5 we discuss a series of experiments on real and simulated images. Finally, section 6 closes with a brief recapitulation and conclusions. The paper ends with appendices containing detailed notation and complete mathematical proofs.

## 2. Image formation model.

2.1. Generalized digital pin-hole camera. An accurate estimation of the PSF requires a proper modeling of the digital image formation process. The geometric component of this process is most often modeled in computer vision by a pin-hole camera. An ideal pin-hole camera with focal length $f$, shooting at a planar scene $u$ from a distance $d$ and at frontoparallel pose, will produce an image $w(\mathbf{x})=u(\lambda \mathbf{x})$ which is just a homothecy of scale factor $\lambda=\frac{d}{f}$ of the original planar scene $u$.

If the pose is not perfectly fronto-parallel or the pin-hole camera presents noncanonical internal calibration parameters, $w$ and $u$ are related by a planar homography $D$, i.e., $w=u \circ D$. In a more accurate camera model the distortion $D$ takes the form of a more general (but regular) diffeomorphism. This is required when the scene is a regular close-to-planar surface (as is assumed here) or when the geometric distortion due to the optical system is taken into account as suggested in [38, 21, 14].

For the purpose of PSF estimation this simple model needs to be augmented with an accurate radiometric component, comprising at least the following elements.

Blurring. The PSF kernel $h$ models blur due to intrinsic camera characteristics, such as diffraction when light goes through a finite aperture, light averaging within the sensor, and lens aberration. Other blur sources such as motion, atmospheric turbulence, or defocus blur, which may change from one snapshot to another, will be minimized by the experimental procedure, and it is not the goal of the present work to estimate them. Another implicit assumption that is usually made is that as long as the camera is in focus, the PSF is independent of the focus position, i.e., the relative distance between the sensor array and the lens system. Therefore in focus images captured with the same camera configuration are affected by the same PSF.

The diffraction kernel is determined by the shape and size of the aperture, the focal length, and the wavelength of the considered monochromatic light. Under the Fraunhofer far-field approximation, for incoherent light this kernel is the squared Fourier transform modulus of the camera's aperture indicator function [15]. It follows that the PSF diffraction kernel is always nonnegative and band-limited.

Besides the kernel due to diffraction, other sources of blur inherent to the optical system are present in real cameras. These are mainly optical aberrations, and antialiasing filters (which reduce aliasing but do not completely cancel it) introduced in the system prior to sampling [35, 39]. The sampling process also introduces blur. Indeed, each photo-sensor in the rectangular sampling grid integrates the light arriving at a particular exposure time. This corresponds to a convolution with the indicator function of the photo-sensor active area. To sum up, the unknown PSF results basically from the convolution of three nonnegative kernels (diffraction, aberration and antialiasing filters, and sensor averaging), one of them being bandlimited. No parametrical model on the PSF will be adopted here. Nonetheless, the physical modeling justifies our assumption that the PSF is band-limited and nonnegative.

Sampling. We model the continuous to digital conversion at the image plane by introducing an ideal sampling operator $\mathbf{S}_{1}$ and additive noise $\mathbf{n}$ due to measurement uncertainties. Physical models of digital camera sensors, both for CCD and CMOS sensors, suggest that the readout noise $\mathbf{n}$ is a mixture of luminance independent (Gaussian, thermal) noise, and luminance dependent (Poisson or photon counting) noise [16, 34, 25]. A usual simplification of this model, which we follow here, assumes the noise is image independent, white, and Gaussian, with constant variance.

The whole image formation process can then be summarized in a single equation:

$$
\tilde{\mathbf{v}}=g\left(\mathbf{S}_{1}((u \circ D) * h)\right)+\mathbf{n}
$$

where $g(\cdot)$ is a monotone nondecreasing function that describes the nonlinear sensor response. If the camera is working outside the saturation zone, in the raw camera output this response can reasonably be assumed to be linear [14]. This boils down to a rescaling of the dynamics of $u$ and therefore disappears without loss of generality from the model. Hence, in what follows, the image formation model will be
(M) $\quad \tilde{\mathbf{v}}=\mathbf{S}_{1}((u \circ D) * h)+\mathbf{n}$.
2.2. Inverse problem statement in terms of digital sequences. Since in practice our data consist exclusively of discrete sequences (or digital images), the image formation model will be rewritten in terms of discrete sequences. This requires the introduction of additional notation, summarized in Table 1 (a more precise definition of each term is presented in Appendix A). It would be cumbersome to verify systematically all regularity requirements on all functions and distributions needed in the proofs. Thus, all necessary results are given in a precise form in the appendices. They will be invoked in the proofs, and the reader is invited to check that their use was licit.

Suppose that the PSF $h$ is $s$-band-limited, that is, $\operatorname{supp}(\hat{h})=[-s \pi, s \pi]^{2}$. Then, if sampled at a rate $s$, the Nyquist sampling theorem guarantees perfect reconstruction of $h$ from its samples $\mathbf{h}=\mathbf{S}_{1} H_{\frac{1}{s}} h$. We are actually interested in the case $s>1$, usual for digital cameras. This means that the images obtained from (M) may be subject to aliasing. Following Lemma C. 3 (equation (C.4)), the image formation model (M) can be written in terms of discrete sequences:

$$
\begin{align*}
\tilde{\mathbf{v}} & =\mathcal{S}_{s}\left(\overline{\mathbf{u}}_{D} * \mathbf{h}\right)+\mathbf{n} \\
& =\mathcal{S}_{s} \mathbf{C}\left[\overline{\mathbf{u}}_{D}\right] \mathbf{h}+\mathbf{n} . \tag{2.1}
\end{align*}
$$

Table 1
Notation used in this paper.

```
\(u, v \quad\) Images defined on continuous domain \(\mathbf{x} \in \mathbb{R}^{2}\)
\(\mathbf{u}, \mathbf{v}\) Digital images are sampled on a discrete grid \(\mathbf{k} \in \mathbb{Z}^{2}\)
F Fourier transform
\(\hat{f} \quad\) Fourier transform of a function \(f\)
\(\mathbf{I}_{1} \quad\) Shannon-Whittaker interpolator: \(\mathbf{I}_{1} \mathbf{u}(\mathbf{x})=\sum_{\mathbf{k}} \mathbf{u}(\mathbf{k}) \operatorname{sinc}(\mathbf{x}-\mathbf{k})\)
\(\mathbf{S}_{1} \quad\) 1-sampling operator: \(\mathbf{u}(\mathbf{k})=\left(\mathbf{S}_{1} u\right)(\mathbf{k})=u(\mathbf{k})\)
\(W_{w} \quad\) Ideal low-pass filter that cuts the spectrum of continuous signals to \([-w \pi, w \pi]^{2}\)
\(\mathcal{S}_{s} \quad\) The \(s\)-to-1-resampling operator \(\mathcal{S}_{s}=\mathbf{S}_{1} H_{s} \mathbf{I}_{1}\)
\(H_{\lambda} \quad\) Continuous homothecy: \(H_{\lambda} u(x, y)=\lambda^{2} u(\lambda x, \lambda y) .(\lambda<1\) dilation, \(\lambda>1\) contraction)
\(\mathbf{H}_{\alpha} \quad\) Digital Nyquist homothecy operator of parameter \(\alpha: \mathbf{H}_{\alpha} \mathbf{u}:=\mathbf{S}_{1} W_{1} H_{\alpha} \mathbf{I}_{1} \mathbf{u}\)
\(\mathbf{C}[\mathbf{u}]\) Linear map associated to the convolution with a digital image \(\mathbf{u}\)
\(\mathbf{L}^{*} \quad\) Adjoint of a linear operator \(\mathbf{L}\)
\(\mathbf{L}^{+} \quad\) Pseudoinverse \(\mathbf{L}^{+}:=\left(\mathbf{L}^{*} \mathbf{L}\right)^{-1} \mathbf{L}^{*}\) of a linear operator \(\mathbf{L}\)
\(L^{1} \quad\) Integrable functions on \(\mathbb{R}^{2}\left(L^{1}\left(\mathbb{R}^{2}\right)\right)\)
\(L^{2} \quad\) Square integrable functions \(\left(L^{2}\left(\mathbb{R}^{2}\right)\right)\)
\(\mathcal{B} \mathcal{L}^{2} \quad L^{2}\) functions, band-limited in \([-\pi, \pi]^{2}\)
\(\mathcal{B L}_{0}^{2} \quad L^{2}\) functions with compactly supported Fourier transform
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The digital image $\overline{\mathbf{u}}_{D}=\mathbf{S}_{1} W_{1} H_{\frac{1}{s}} u_{D}$ is a well-sampled version of the distorted image $u_{D}=$ $u \circ D$. The value $s$ is the resampling rate from the high resolution lattice $s \times$, where the PSF estimation will take place, to the $1 \times$ sensor grid.

The numerical method will recover only a finite number of samples $\mathbf{h}$ of $h$. Strictly speaking, $h$, being band-limited, cannot be compactly supported. Nonetheless, the error introduced by assuming that the support of $h$ is bounded will prove negligible in comparison to the other sources of error: image noise, quantization, slight estimation errors of $D$, etc. Indeed, the retrieved solution $\mathbf{h}$ will prove to be experimentally independent from variations of its assumed support as long as it is large enough for errors to be negligible and small enough for the operator to still be well-conditioned.

When $\mathbf{n}$ is a zero-mean white discrete Gaussian noise, it follows from the previous formula that $\mathbf{h}_{e}=\left(\mathcal{S}_{s} \mathbf{C}\left[\overline{\mathbf{u}}_{D}\right]\right)^{+} \tilde{\mathbf{v}}$ is an unbiased estimator of $\mathbf{h}$, as long as the linear operator $\mathcal{S}_{S} \mathbf{C}\left[\overline{\mathbf{u}}_{D}\right]$ is injective. It can be shown that the estimator variance is proportional to the Hilbert-Schmidt norm of $\left(\mathcal{S}_{S} \mathbf{C}\left[\overline{\mathbf{u}}_{D}\right]\right)$ (for matrices, the Frobenius norm ${ }^{1}$ ), and that it is nearly minimal when $\overline{\mathbf{u}}_{D}$ is a white noise realization (see [14]).
3. PSF estimation from an unknown pair of scaled images. Assume that we have perfect knowledge of the latent sharp image $u$ that produced the blurry aliased observation $\tilde{\mathbf{v}}$. Under this nonblind assumption, solving for the PSF amounts to solving an inverse problem governed by the image formation model (M). Of course, this would require the use of a specially designed calibration pattern. We are now interested in investigating to what extent the use of such a pattern could be circumvented. We will propose a method that allows us to accurately estimate the PSF from a pair of snapshots of the same scene, captured from different distances. In this

[^1]method, the closest image will play a role similar to that of a calibration pattern in a classical nonblind approach.

In a previous work, we have shown that the highest accuracy in the PSF estimation is obtained by using a realization of a Bernoulli white noise as calibration target [14]. However, many highly textured scenes do exist in nature which, while not being optimal, may still lead to a well-posed inverse problem. In what follows, we prove that from two far apart snapshots of this kind of scene, complete recovery of the camera PSF is theoretically possible based on the estimation of the blur between this pair.
3.1. Relative blur between two images: The interimage kernel. Consider two digital images $\tilde{\mathbf{v}}_{1}, \tilde{\mathbf{v}}_{2}$ of the same planar scene $u$, captured from different distances in a fronto-parallel position with negligible rotation around the optical axis. Let $\lambda_{1}$ and $\lambda_{2}$ denote the corresponding scale factors between the scene and each of the images. Then,

$$
\begin{align*}
\tilde{\mathbf{v}}_{i} & =\mathbf{S}_{1} H_{\lambda_{i}} u * h+\mathbf{n}_{i} \quad \text { for } i=1,2  \tag{3.1}\\
& =\mathbf{S}_{1} v_{i}+\mathbf{n}_{i} \\
& =\mathbf{v}_{i}+\mathbf{n}_{i}
\end{align*}
$$

where $v_{i}:=H_{\lambda_{i}} u * h$ and $\mathbf{v}_{i}:=\mathbf{S}_{1} v_{i}$. We will realistically assume that $h \in L^{1} \cap \mathcal{B} \mathcal{L}_{0}^{2}$ is nonnegative with $\|h\|_{L^{1}}=1$, and $u \in \mathcal{B L}_{0}^{2}$ (details on the appropriateness of these assumptions are given in Appendix A.1). Also, it will be assumed that the acquisition distances are such that $s \lambda_{1}<\lambda_{2}$; the importance of this assumption will soon become clear.

Definition 3.1. Let $v_{1}, v_{2} \in \mathcal{B} \mathcal{L}_{0}^{2}$ be two fronto-parallel continuous views of the same scene, acquired from different distances $\lambda_{1}$ and $\lambda_{2}$, respectively. We define an interimage kernel between $v_{1}$ and $v_{2}$ as any kernel $k \in \mathcal{B L}_{0}^{2}$ satisfying

$$
v_{2}=H_{\lambda_{2} / \lambda_{1}} v_{1} * k
$$

The following lemma provides a characterization of the interimage kernel.
Lemma 3.2. Let $h \in L^{1} \cap \mathcal{B} \mathcal{L}_{0}^{2}$ be nonnegative, band-limited with $\operatorname{supp}(\hat{h}) \subset[-s \pi, s \pi]^{2}$ and $\hat{h}(0)=1$. Let $\rho$ be the largest positive number such that $|\hat{h}(\zeta)|>0$ for every $\|\zeta\|_{\infty}<\rho \pi$, and assume that $\lambda_{2} \rho>s \lambda_{1}$. Then there is an interimage kernel $k \in \mathcal{B L}_{0}^{2}$ with support in $[-s \pi, s \pi]^{2}$ between (fronto-parallel views) $v_{1}$ and $v_{2}$ that satisfies

$$
\begin{equation*}
H_{\lambda} h * k=h, \quad \text { where } \lambda=\frac{\lambda_{2}}{\lambda_{1}} . \tag{3.2}
\end{equation*}
$$

If $\hat{u}$ does not vanish inside $\left[-s \frac{\pi}{\lambda_{2}}, s \frac{\pi}{\lambda_{2}}\right]$, then the interimage kernel is unique and depends only on $h$ and $\lambda$.

Proof. If $k$ is an interimage kernel between $v_{1}$ and $v_{2}$, according to Definition 3.1 it must satisfy

$$
\mathbf{F}\left(H_{\lambda} v_{1}\right)(\zeta) \hat{k}(\zeta)=\hat{v}_{2}(\zeta)
$$

Since $v_{i}:=H_{\lambda_{i}} u * h$, the right-hand side of the previous equation is given by

$$
\hat{v}_{2}(\zeta)=\hat{u}\left(\zeta / \lambda_{2}\right) \hat{h}(\zeta)
$$

In the same way, for the left-hand side,

$$
H_{\lambda} v_{1}=H_{\lambda}\left(H_{\lambda_{2}} u * h\right) \stackrel{(\mathrm{C} .3)}{=} H_{\lambda_{2}} u * H_{\lambda} h,
$$

i.e., $\mathbf{F}\left(H_{\lambda} v_{1}\right)(\zeta) \hat{k}(\zeta)=\hat{u}\left(\zeta / \lambda_{2}\right) \hat{h}(\zeta / \lambda)$. Hence,

$$
\begin{equation*}
\hat{u}\left(\zeta / \lambda_{2}\right) \hat{h}(\zeta / \lambda) \hat{k}(\zeta)=\hat{u}\left(\zeta / \lambda_{2}\right) \hat{h}(\zeta) . \tag{3.3}
\end{equation*}
$$

It follows that a sufficient condition for $k$ to be an interimage kernel is $\hat{h}(\zeta / \lambda) \hat{k}(\zeta)=\hat{h}(\zeta)$. Since $h \in L^{1}, \hat{h}$ is continuous. It follows that $\rho$ is necessarily positive, since $\hat{h}(0)=1>0$. In addition, as $\lambda>\frac{s}{\rho}$ by hypothesis, $\mathbf{F}\left(H_{\lambda} h\right)(\zeta)=\hat{h}(\zeta / \lambda)$ does not vanish inside $[-s \pi, s \pi]^{2}$ and

$$
\begin{equation*}
\hat{k}(\zeta)=\frac{\hat{h}(\zeta)}{\hat{h}(\zeta / \lambda)} \tag{3.4}
\end{equation*}
$$

is well defined all over its support, $\operatorname{supp}(\hat{k}) \subset[-s \pi, s \pi]^{2}$. Finally, if $\hat{u}\left(\zeta / \lambda_{2}\right)$ does not vanish within the support of $\hat{h}$, from (3.3) $k$ is unique.

Remark 1. In Lemma 3.2 it is assumed that the PSF $h$ is the same for the two images. This has at least two practical implications. First, we assume that both images are taken in perfect focus through proper refocusing. The only camera parameter allowed to change is the focus (i.e., aperture and focal distance remain unchanged). Second, the common area between $\mathbf{v}_{1}$ and $\mathbf{v}_{2}$ covers an important part of $\mathbf{v}_{1}$, and consequently its PSF may exhibit some space variance that may degrade the estimation. This can be avoided by taking snapshots with their common area covering only the central part of the frame, where the kernel does not change significantly.
3.2. Estimation of the interimage kernel. The next goal is to estimate the interimage kernel $k$. Since $k$ is an $s$-band-limited function, we will work with its $s \times$ samples $\mathbf{k}=\mathbf{S}_{1} H_{\frac{1}{s}} k$. We will show that under reasonable conditions, $\mathbf{k}$ can be recovered from the noisy aliased observations $\tilde{\mathbf{v}}_{1}$ and $\tilde{\mathbf{v}}_{2}$. Let us first build up some intuition on how to derive the proposed estimator. In what follows, $\dot{\mathbf{v}}_{1}=\mathbf{S}_{1} W_{1} H_{\frac{\lambda}{s}} v_{1}$ denotes a well-sampled homothecy of parameter $\lambda / s$ of $v_{1}$.

Proposition 3.3. Under the assumptions of Lemma 3.2,

$$
\begin{equation*}
\mathbf{v}_{2}=\left(\mathcal{S}_{s} \mathbf{C}\left[\dot{\mathbf{v}}_{1}\right]\right) \mathbf{k} \tag{3.5}
\end{equation*}
$$

Proof. Since $k$ is an interimage kernel between $v_{1}$ and $v_{2}$, it satisfies (3.2). Then,

$$
\begin{aligned}
\mathbf{v}_{2} & =\mathbf{S}_{1}\left(v_{2}\right) \\
& =\mathbf{S}_{1}\left(H_{\lambda} v_{1} * k\right) .
\end{aligned}
$$

Since $k$ is $s$-band-limited, it follows that

$$
\begin{equation*}
\mathbf{v}_{2} \stackrel{(\mathrm{C} .1)}{=} \mathbf{S}_{1}\left(W_{s} H_{\lambda} v_{1} * W_{s} k\right) . \tag{3.6}
\end{equation*}
$$

Using the Nyquist-Shannon theorem for a band-limited signal and a set of properties detailed in Appendices B and C yields

$$
\begin{aligned}
& \mathbf{v}_{2} \stackrel{s>0}{=} \mathbf{S}_{1} H_{s} H_{\frac{1}{s}}\left(W_{s} H_{\lambda} v_{1} * W_{s} k\right) \\
& \stackrel{(\mathrm{C} .3)}{=} \mathbf{S}_{1} H_{s}\left(H_{\frac{1}{s}} W_{s} H_{\lambda} v_{1} * H_{\frac{1}{s}} W_{s} k\right) \\
& \stackrel{(\mathrm{C} .2)}{=} \mathbf{S}_{1} H_{s}\left(W_{1} H_{\frac{\lambda}{s}} v_{1} * W_{1} H_{\frac{1}{s}} k\right) \\
& \stackrel{(\mathrm{B} .5)}{=} \mathbf{S}_{1} H_{s} \mathbf{I}_{1} \mathbf{S}_{1}\left(W_{1} H_{\frac{\lambda}{s}} v_{1} * W_{1} H_{\frac{1}{s}} k\right) \\
& \stackrel{(\mathrm{C} .4)}{=} \mathbf{S}_{1} H_{s} \mathbf{I}_{1}\left(\mathbf{S}_{1} W_{1} H_{\frac{\lambda}{s}} v_{1} * \mathbf{S}_{1} W_{1} H_{\frac{1}{s}} k\right) \\
& \stackrel{\text { def }}{=} \mathbf{S}_{1} H_{s} \mathbf{I}_{1}\left(\stackrel{\circ}{\mathbf{v}}_{1} * \mathbf{k}\right) \\
& \stackrel{\text { def }}{=} \mathcal{S}_{s}(\stackrel{\mathbf{v}}{1} * \mathbf{k}) \\
& \stackrel{\text { def }}{=} \mathcal{S}_{s} \mathbf{C}\left[\stackrel{v}{0}_{1}\right] \mathbf{k} .
\end{aligned}
$$

Of course, in practice we do not have access to $\stackrel{\mathbf{v}}{1}$ or to $\mathbf{v}_{2}$, but only to their noisy, aliased versions $\tilde{\mathbf{v}}_{1}$ and $\tilde{\mathbf{v}}_{2}$. Thus $\mathbf{k}$ cannot be directly estimated from (3.5). However, a relationship between $\dot{\mathbf{v}}_{1}$ and $\tilde{\mathbf{v}}_{1}$ can be established as follows:

$$
\begin{align*}
\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1} & =\mathbf{H}_{\frac{\lambda}{s}}\left(\mathbf{v}_{1}+\mathbf{n}_{1}\right)+\dot{\mathbf{v}}_{1}-\dot{\mathbf{v}}_{1} \\
& =\stackrel{\circ}{\mathbf{v}}_{1}+\underbrace{\mathbf{S}_{1} W_{1} H_{\frac{\lambda}{s}}\left(\mathbf{I}_{1} \mathbf{v}_{1}-v_{1}\right)}_{\mathbf{r}}+\mathbf{H}_{\frac{\lambda}{s}} \mathbf{n}_{1}, \tag{3.7}
\end{align*}
$$

where the last equality results from the definition of the discrete homothecy operator. The term $\mathbf{r}$ is a consequence of aliasing when sampling $v_{1}$ and introduces an unknown bias in the estimation of $\mathbf{k}$. While this bias cannot be fully controlled, its impact can be mitigated. Indeed, since

$$
\begin{aligned}
& \mathbf{r}=\mathbf{S}_{1} W_{1} H_{\frac{\lambda}{s}}\left(\mathbf{I}_{1} \mathbf{v}_{1}-v_{1}\right) \\
& \stackrel{(\mathrm{C}, 2)}{=} \mathbf{S}_{1} H_{\frac{\lambda}{s}} W_{\frac{s}{\lambda}}\left(\mathbf{I}_{1} \mathbf{v}_{1}-v_{1}\right),
\end{aligned}
$$

the aliasing term $\mathbf{r}$ will be nonzero only if there are aliasing components in the frequency interval $\left[-\frac{s}{\lambda} \pi, \frac{s}{\lambda} \pi\right]^{2}$. This allows us to choose $v_{1}=H_{\lambda_{1}} u$ such that $\operatorname{supp}\left(\hat{v}_{1}\right) \subset\left[-2 \pi+\frac{s}{\lambda} \pi, 2 \pi-\frac{s}{\lambda} \pi\right]^{2}$ (see Figure 1). Thus, to minimize the impact of the aliasing term, the images should be acquired from a pair of fronto-parallel locations as far as possible one from the other, since that amounts to increasing the value of $\lambda$.

From now on, we assume that the snapshots are acquired following the previous considerations. Therefore, we can ignore the aliasing term in (3.7), which leads to

$$
\mathbf{v}_{2}=\left(\mathcal{S}_{s} \mathbf{C}\left[\dot{\mathbf{v}}_{1}\right]\right) \mathbf{k}=\left(\mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \mathbf{v}_{1}-\mathbf{H}_{\frac{\lambda}{s}} \mathbf{n}_{1}\right]\right) \mathbf{k}
$$

that is,

$$
\left(\mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}-\mathbf{H}_{\frac{\lambda}{s}} \mathbf{n}_{1}\right]\right) \mathbf{k}=\tilde{\mathbf{v}}_{2}-\mathbf{n}_{2} .
$$



Figure 1. Neglecting the aliasing. The estimation will be affected by aliasing only if there are aliasing components in the interval $\left[-\frac{s}{\lambda} \pi, \frac{s}{\lambda} \pi\right]^{2}$. Hence, to avoid aliasing one can choose $v_{1}=H_{\lambda_{1}} u$ such that $\operatorname{supp}\left(\hat{v}_{1}\right) \subset\left[-2 \pi+\frac{s}{\lambda} \pi, 2 \pi-\frac{s}{\lambda} \pi\right]^{2}$.

One could be tempted to solve for $\mathbf{k}$ in the previous equation using a total least squares based approach:

$$
\text { (TLS) } \underset{\mathbf{k}, \boldsymbol{\delta}, \boldsymbol{\epsilon}}{\arg \min }\|\boldsymbol{\delta}\|+\kappa\|\boldsymbol{\epsilon}\| \text { subject to } \mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}+\boldsymbol{\delta}\right] \mathbf{k}=\tilde{\mathbf{v}}_{2}+\boldsymbol{\epsilon} .
$$

However, the particular structure of the operator $\mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{\mathbf{1}}+\boldsymbol{\delta}\right]$ makes this problem a difficult one. Instead we prefer to follow a simpler approach, which results from neglecting the noise term $\mathbf{H}_{\frac{\lambda}{s}} \mathbf{n}_{1}$. This yields to the least squares estimation problem

$$
\text { (LS) } \underset{\mathbf{k}, \boldsymbol{\epsilon}}{\arg \min }\|\boldsymbol{\epsilon}\| \text { subject to } \mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}\right] \mathbf{k}=\tilde{\mathbf{v}}_{2}+\boldsymbol{\epsilon}
$$

whose solution is given by

$$
\begin{equation*}
\mathbf{k}_{e}=\left(\mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}\right]\right)^{+} \tilde{\mathbf{v}}_{2} \tag{3.8}
\end{equation*}
$$

If the noise $\mathbf{n}_{1}$ is small compared to $\mathbf{v}_{1}$, this solution would be very close to the one that would be obtained from (TLS). If, in addition, $\mathbf{n}_{2}$ is small compared to $\mathbf{v}_{2}$, both solutions would be close to the actual interimage kernel $\mathbf{k}=\left(\mathcal{S}_{s} \mathbf{C}\left[\mathbf{v}_{1}\right]\right)^{+} \mathbf{v}_{2}$. This follows directly from the continuity and injectivity assumptions on $\mathcal{S}_{s} \mathbf{C}\left[\stackrel{\vee}{\mathbf{v}}_{1}\right]$, as a consequence of Lemma D.1. This being said, we will consider the estimator of the interimage kernel in (3.8).

Remark 2. If $\lambda<s$, the convolution between $\mathbf{k}$ and $\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}$ is not invertible so the operator $\mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}\right]$ will not be injective. This constraint on $\lambda$ is necessary but not sufficient to make $\mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}}^{s} \tilde{\mathbf{v}}_{1}\right]$ invertible. In addition, it is required that the spectrum of the image $\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}$ exhibits slow decay. Indeed, it is shown in [14] that the flatter the spectrum of the image scene is, the better conditioned is the inverse problem. For that reason, in order to obtain accurate estimates of $\mathbf{k}$, it is desirable that the chosen scene $u$ exhibits white noise characteristics.
3.3. From relative to absolute blur. Now that we have a method for estimating the interimage kernel $\mathbf{k}$, we will concentrate on how to recover the camera PSF. Notice that $h$ is related to $k$ by $H_{\lambda} h * k=h$, and therefore its derivation is not straightforward. However, as we prove in Appendix C (Lemma C.5), it holds that

$$
\begin{equation*}
h=\lim _{n \rightarrow \infty} H_{\lambda^{n-1}} k * \cdots * H_{\lambda} k * k \tag{3.9}
\end{equation*}
$$

This equality shows that it is possible to recover the camera PSF $h$ from the interimage kernel $k$. Recall that in practice we have access only to discrete sequences; therefore it is convenient to derive a discrete equivalent of the previous limit. Since $\mathbf{k}$ is $s$-band-limited,

$$
\begin{aligned}
\mathbf{S}_{1} H_{\frac{1}{s}}\left(H_{\lambda} k * k\right) & \stackrel{(\mathrm{C} .1)}{=} \mathbf{S}_{1} H_{\frac{1}{s}}\left(W_{s} H_{\lambda} k * k\right) \\
& \stackrel{(\mathrm{C.} 3)}{=} \mathbf{S}_{1}\left(H_{\frac{1}{s}} W_{s} H_{\lambda} k * H_{\frac{1}{s}} k\right) \\
& \stackrel{(\mathrm{C} .2)}{=} \mathbf{S}_{1}\left(W_{1} H_{\frac{\lambda}{s}} k * H_{\frac{1}{s}} k\right) \\
& \stackrel{(\mathrm{C} .5)}{=} \mathbf{S}_{1} W_{1} H_{\frac{\lambda}{s}} k * \mathbf{S}_{1} H_{\frac{1}{s}} k \\
& \stackrel{\text { def }}{=} \mathbf{H}_{\lambda} \mathbf{k} * \mathbf{k}
\end{aligned}
$$

Iteratively applying this result to (3.9) yields

$$
\begin{equation*}
\mathbf{h}=\lim _{n \rightarrow \infty} \mathbf{H}_{\lambda^{n-1}} \mathbf{k} * \cdots * \mathbf{H}_{\lambda} \mathbf{k} * \mathbf{k} \tag{3.10}
\end{equation*}
$$

4. The complete PSF estimation procedure. This section describes the algorithmic steps that lead to local subpixel PSF estimates. The complete chain is summarized in the block diagram of Figure 2. The next paragraphs present brief summaries for each block.

Image alignment. In order to estimate the geometric transformation between both images, they need to be precisely aligned. This alignment can be obtained by matching SIFT descriptors [23], which have the advantage of being scale invariant. We chose to use the ASIFT descriptor [27], more precisely, the implementation [37] that includes rejection of false matches by means of the optimized random sampling algorithm (ORSA).

Geometric transform estimation. The complete geometric transformation from one image to the other was approximated with thin-plate splines [6,32] from the matched SIFT pairs. This permits the correction of small nonaffine distortions, and allows for deviations from the fronto-parallel assumption in the acquisition. Of course, if the distortion is significant the assumed interimage kernel, (3.2) will not be accurate. The thin-plate representation as affine + nonaffine parts of the geometric distortion is especially helpful in estimating the relative scale $\lambda=\left(\lambda_{x}, \lambda_{y}\right)$ between both views, since this can be directly estimated from the affine part.

Gray level adjustment. Both snapshots should be acquired with exactly the same camera configuration and constant scene illumination. This ensures that there is no contrast change between them.


Figure 2. Algorithm description. Both captured images are aligned via SIFT feature matching followed by the estimation of a smooth geometric distortion through thin-plate approximation of matched features. The relative geometric distortion and gray-level corrections are applied to a low-pass (unaliased) version of the finest scale image $\tilde{\mathbf{v}}_{1}$. Then the interpolated image $\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}$ and image $\tilde{\mathbf{v}}_{2}$ are compared to obtain the interimage kernel $k$, which is later iteratively updated to obtain the absolute camera PSF $h$.

Resampling and distortion correction of $\tilde{\mathbf{v}}_{1}$. The generation of the rescaled samples $\mathbf{H}_{\underline{\lambda}} \tilde{\mathbf{v}}_{1}$ requires the interpolation of $\tilde{\mathbf{v}}_{1}$ at the desired scale $\lambda / s$. This is done by using the estimated geometric transformation with bicubic interpolation. Notice that since $\tilde{\mathbf{v}}_{1}$ is not very aliased, one can correctly interpolate it without introducing artifacts.

Numerical methods for interimage kernel estimation. Suppose that the image $\tilde{\mathbf{v}}_{2}$ has size $m \times n$. The goal is to estimate $k$ at $s \times$ the resolution of $\tilde{\mathbf{v}}_{2}$ (camera sensor resolution). Also suppose that the estimated support of the interimage kernel $k$ is contained in an $r \times r$ patch. Then the matrix $\mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}\right]$ is of size $m n \times r^{2}$. A simple least squares procedure yields the interimage kernel estimator:

$$
\mathbf{k}_{e}=\underset{\mathbf{k}}{\arg \min }\left\|\mathcal{S}_{s} \mathbf{C}\left[\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1}\right] \mathbf{k}-\tilde{\mathbf{v}}_{2}\right\|^{2}
$$

Transforming the kernel: From $\mathbf{k}$ to $\mathbf{h}$. Recovering the samples of the camera PSF $\mathbf{h}$ amounts to evaluating the limit in (3.10). Directly working with the digital sequences requires some care in how the successive convolutions are computed. Since $\lambda>1$, the application of $\mathbf{H}_{\lambda}$ would require a low-pass filter to avoid aliasing artifacts. To bypass this inconvenience one can restate the limit convolution as follows:

$$
\mathbf{h}=\lim _{n \rightarrow \infty} \mathbf{H}_{\lambda^{n}}\left(\mathbf{k} * \mathbf{H}_{\frac{1}{\lambda}} \mathbf{k} * \cdots * \mathbf{H}_{\frac{1}{\lambda^{n}}} \mathbf{k}\right) .
$$

If implemented in this way, the successive discrete convolutions can be computed without any special care. To apply the discrete homothecy operator to $\mathbf{k}$, we need to resample $\mathbf{k}$ using the Shannon-Whittaker interpolator. Because of its slow decay, in order to reduce ringing and other windowing effects, we opted to use bicubic interpolation. The whole derivation of $\mathbf{h}$ from $\mathbf{k}$ is summarized as follows.

This algorithm converges after a few iterations since $\lambda^{n}$ grows very fast. In practice, we set $\lambda_{\max }=50$, since the convolution with a $50 \times$ scale-contracted interimage kernel produces a negligible change in the final result (it amounts to convolving with a Kronecker delta).

```
Algorithm 1. From interimage kernel to PSF.
    1. Initialize \(\mathbf{w}_{0}=\mathbf{k}, n=1\).
    2. Compute \(\mathbf{H}_{1 / \lambda^{n}} \mathbf{k}\) by using \(\lambda=\left(\lambda_{x}, \lambda_{y}\right)\) (from thin-plates affine part).
    3. Calculate \(\mathbf{u}_{n}=\mathbf{H}_{1 / \lambda^{n}} \mathbf{k} * \mathbf{w}_{n-1}\).
    4. If \(\min \left\{\lambda_{x}^{n}, \lambda_{y}^{n}\right\}>\lambda_{\max }\) go to 5 . Else update \(n:=n+1\) and repeat from 2 .
    5. Calculate \(\mathbf{h}=\mathbf{H}_{\lambda}^{n} \mathbf{w}_{n}\).
```

In theory, as we already stated, the estimated PSF should be nonnegative. In practice, small negative values may be observed, due to deviations from model assumptions and numerical artifacts. To correct for these deviations, we simply set all negative values to zero.
5. Experimental results. Since there is no PSF ground truth available, the validation of the proposed method was carried out by simulations and by comparing the results with state-of-the-art methods [21, 20, 14, 18]. Comparison was made only to nonblind, target-based methods, as the accuracy of blind methods is significantly lower. A complete algorithmic description, an online demo facility, and a reference source code can be found at the IPOL workshop [13].
5.1. Simulations as a sanity check. A synthetic random image $u$ was generated and reinterpolated $4 \times$ in order to get the "continuous" sharp homothecy of the image $u$. Next, both images were convolved with a PSF-like kernel (in this case a Gaussian isotropic kernel) and downsampled to get the respective observed digital images at the camera resolution (i.e., $1 \times)$. The kernel was chosen so that the low resolution image presents aliasing artifacts. By generating the views of $u$ in this way, there are no aliasing artifacts in the closest image. This experiment was done as a sanity check of the proposed method. A $4 \times$ kernel was estimated from the observed image pair. The results are shown in Figures 3 and 4.

The procedure was tested for both automatic SIFT-based registration and the ideal (known) alignment. Although both estimates are significantly accurate, the automatic registration introduces a small misalignment, as shown in the difference images. See the captions of Figures 3 and 4 for details.
5.2. Real camera examples. The behavior of the proposed approach was tested for several different image pairs and for superresolution estimations ranging from $1 \times$ to $4 \times$. The experiments were performed using a Canon EOS 400D camera equipped with a Tamron AF $17-50 \mathrm{~mm}$ F/2.8 XR Di-II lens. The focal length was fixed to 50.0 mm .

Two-scale versus nonblind, target-based method. In [14] we proposed a nonblind method that uses a realization of white noise as calibration pattern. It was proved that, until now, this method is the one that estimates the PSF with highest accuracy. Therefore, the PSF resulting from this method will be used here as ground truth. Figure 5 shows the $4 \times$ PSF estimated by the proposed two-scale method from a pair of views of a wall shown in Figure 6. The estimation was conducted for one of the green channels (half of the green pixels of the Bayer matrix), with the camera aperture set to $\mathrm{f} / 5.7$. The estimated PSF is quite close to the one obtained by using our random target-based approach. In particular their sizes are similar, and their corresponding MTFs present zeros at the same locations.


Figure 3. Synthetic example: $4 \times P S F$ estimation for simulated data. Top row: the closest and farthest images. Middle row: the simulated PSF (ground truth) and the respective PSF estimations using the automatic SIFT points/thin-plate alignment and the ideal alignment. Both estimations are accurate. However, as shown in the difference images, the automatic registration introduces a small misalignment. This can also be seen in the phase and modulus of the PSF Fourier transform vertical profile, shown in the bottom row. Bottom row (right): comparison of the interimage and PSF kernels. Since both input images are simulated at distances in a ratio of $\lambda=4 \times, h$ is very close to $k$.

farthest image

image difference, ideal alignment

image difference, SIFT alignment

Figure 4. Synthetic example: $4 \times$ PSF estimation for simulated data, residual image. From left to right: farthest image and residual images $\mathcal{S}_{s}\left(\mathbf{H}_{\frac{\lambda}{s}} \tilde{\mathbf{v}}_{1} * \mathbf{k}\right)-\tilde{\mathbf{v}}_{2}$ with the estimated kernel from ideal and automatic alignment. The residual in the automatic alignment case is significantly larger than in the ideal alignment case. However, the difference in the PSFs seems to be negligible up to a subpixel translation as shown in Figure 3.

Color filter array estimations. Two pictures of another textured wall shown in Figure 7 were used to estimate the PSF of the four color Bayer channels (raw camera output). This wall texture presents characteristics similar to those of white noise. The results for the $4 \times \mathrm{PSF}$ estimated at the image center are shown in Figure 7. Notice that the red channel PSF is wider than the green and the blue one, as expected from the physics of diffraction-limited optical systems, since the wavelengths associated to red light are larger than the rest. The differences between the dominant orientations of the red, green, and blue PSFs can be explained by the sensor shape and layout. In fact, each sensor active zone is usually L-shaped, and the red and blue sensors are rotated $90^{\circ}$ with respect to the green ones (see, for example, [36]). These rotations are consistently observed in the PSFs and MTFs estimated with our two-scale method. This clearly illustrates the accuracy of the proposed approach.

Different kinds of scenes. The wall images in the previous experiments are well adapted for our two-scale PSF estimation method, since their spectra show slow decay. A priori one would think that images from pure white noise would yield better estimates, since this is what happens in our previous target-based approach [14]. But for our two-scale approach, this would be true if both snapshots could be precisely aligned, which is not the case in practice. Indeed, SIFT descriptors are not stable in the presence of aliasing. Hence, there is a trade-off between having accurate SIFT matches and textures with high frequency information. The texture shown in Figure 7 is an example of an appropriate trade-off.

Figure 8 shows two snapshots of a photograph in a magazine, with the corresponding $1 \times-$ $4 \times$ PSF estimations for the first green channel. The estimation was performed at the image center for the camera working at an $\mathrm{f} / 5.7$ aperture. All the subpixel estimations are consistent: their MTFs exhibit good overlap in common regions. While these newspaper images produce accurate SIFT points, their spectra decay faster than those of the wall images. Consequently, the high frequencies in the PSF estimate are noisier. This can be readily seen by comparing both estimates at $4 \times$ resolution.

What kind of textures should be used? It follows from the previous analysis that, in order to simultaneously produce good SIFT points and a sufficiently slow frequency decay, textures composed of elements with different sizes are to be preferred. Three-dimensional (3D) textures like those shown in Figure 9 can be problematic for this approach. Even though they respect the two previous conditions, their 3D nature produces disparities and occlusions which change


Figure 5. Wall image: two-scale versus white noise target-based estimation. Estimation at $4 \times$ PSF resolution for one of the green channels from the camera raw output. Top row: two distant, parallel views of a textured wall. Middle row: the PSF estimated with the proposed algorithm and the one estimated using the random target method. Bottom row: vertical profile of the MTF. Both estimations are close. In particular the associated airy disks have similar sizes, and the MTFs vanish in the same locations.
the image beyond a simple zoom. Likewise, non-Lambertian surfaces and dynamical scenes are not appropriated either.

Comparison to other methods. In this experiment we compare the performance of the two-scale method proposed here with three state-of-the-art nonblind methods: that of Joshi and colleagues [21, 20], Imatest commercial software [18], and our previous random target


Figure 6. Wall image: two-scale versus white noise target-based estimation. Two distant, parallel views of a textured wall.
method [14]. All the estimates were computed at the image center, with aperture f/5.7. For the two-scale approach, we used the wall image pair shown in Figure 7. Joshi and colleagues and Imatest use two different kinds of slanted-edge calibration patterns. The algorithm by Joshi requires setting a regularization parameter; we show the results obtained for three different levels of regularization.

Figure 10 shows the MTF profiles of the obtained PSF estimates. The proposed two-scale method performs at least as well as the nonblind methods under comparison. The method of Joshi and colleagues shows similar performance for a carefully, manually chosen regularization parameter. See the caption of Figure 10 for details.
6. Conclusion. In this work we presented an algorithm for the subpixel estimation of the point spread function (PSF) of a digital camera from aliased photographs. The procedure is based on taking two fronto-parallel photographs of the same flat textured scene, from different distances leading to different geometric scales, and then estimating the kernel blur between them.

The estimation method is regularization-free. In that sense, the technique is closely related to our recent nonblind estimation method [14], which uses a random noise pattern. This later paper [14] shows that with such patterns the estimation problem is well-posed and leads to accurate regularization-free estimates. The main difference is that nonblind methods can estimate the PSF directly using the perfect knowledge of the pattern. In the proposed twoscale method the question is far more intricate because only the blur between the acquisitions can be estimated. Thus a mathematical analysis and new algorithms have been introduced proving how the PSF can be recovered from the interimage kernel.

To reach high accuracy, images of textured scenes with sufficiently flat spectra are preferred. It was experimentally verified that many textures found in nature are well adapted to these requirements. A comparison of the resulting PSF estimates with other subpixel PSF estimation methods shows that the proposed algorithm reaches accuracy levels similar to those of state-of-the-art methods, with the advantage of not requiring any special acquisition setup or calibration pattern and thus being much more practical.


Figure 7. Different color channels: PSF estimation at $4 \times$ resolution for the four Bayer channels (two greens, red, and blue). Top row: two distant, parallel views of a concrete wall. Middle row: the $4 \times P S F$ estimated for the four channels. Bottom row: their corresponding Fourier spectrum moduli. The estimation was performed with images captured at aperture $f / 5.7$. The red PSF is larger than the blue and green ones. This is consistent with the diffraction phenomenon: the red wavelengths are larger than the rest; thus their diffraction kernel is wider. Also notice the differences between the shape of the red, blue, and green PSF spectra (bottom row). Red and blue MTFs are rotated $90^{\circ}$ with respect to the green ones. This symmetric behavior is consistent with the layout of L-shaped sensors [36].

## Appendix A. Mathematical framework and physical modeling.

## Functional spaces and other notation.

- $\mathbb{R}^{2}$ is the set of pairs of real numbers $\mathbf{x}=\left(x_{1}, x_{2}\right)$, and $\mathbb{Z}^{2}$ is the set of pairs of integers $\mathbf{k}=\left(k_{1}, k_{2}\right) . L^{1}\left(\mathbb{R}^{2}\right)$ is the set of integrable functions on $\mathbb{R}^{2}, L^{2}\left(\mathbb{R}^{2}\right)$, the set of square integrable functions, $C_{b}^{0}\left(\mathbb{R}^{2}\right)$ is the set of continuous bounded functions, $C^{\infty}\left(\mathbb{R}^{2}\right)$ is the set of infinitely differentiable functions, $\mathcal{S}\left(\mathbb{R}^{2}\right)$ is the Schwartz class of $C^{\infty}$ functions whose derivatives of all orders have fast decay, $\mathcal{S}^{\prime}\left(\mathbb{R}^{2}\right)$ is its dual, the space of tempered distributions, and $\mathcal{E}^{\prime}$ is the subset of $\mathcal{S}^{\prime}\left(\mathbb{R}^{2}\right)$ of compactly supported distributions. We shall use the properties of the convolution $L^{1} * L^{2} \subset L^{2}, L^{1} * L^{1} \subset L^{1}, L^{2} * L^{2} \subset C^{0}$, $\mathcal{E}^{\prime} * \mathcal{S}^{\prime} \subset \mathcal{S}^{\prime}$.


Figure 8. Magazine image: $1 \times, 2 \times, 3 \times$, and $4 \times$ estimations for the first green channel from a pair of photographs of a newspaper image. The estimation was done at the image center for the camera working at f/5.7 aperture. All the estimations are consistent: their MTFs show good overlap. The $4 \times$ PSF estimation is noisier than the one produced from the wall images. The main reason is that the spectrum of the magazine image decays faster.

- We denote by $\mathcal{B} \mathcal{L}^{2}\left(\mathbb{R}^{2}\right)$ (or $\mathcal{B} \mathcal{L}^{2}$ for short) the set of $L^{2}$ functions that are band-limited inside $[-\pi, \pi]^{2}$. More generally, $\mathcal{\mathcal { B }} \mathcal{L}_{0}^{2}$ denotes the space of $L^{2}$ functions with compactly supported Fourier transform.
The following conventions and notation will be used in what follows:
- $\mathbf{F}$ is the Fourier transform operator defined on $\mathcal{S}^{\prime} ; \mathbf{F}(f)(\zeta)=\hat{f}(\zeta)=\int e^{-i \mathbf{x} \cdot \zeta} f(\mathbf{x}) d \mathbf{x}$


Figure 9. Examples of textures which are not adapted to the two-scale approach. Their 3D nature produces disparities, and little changes in the angle-of-view would result in accuracy loss. Non-Lambertian surfaces and dynamical scenes are not appropriated either.


Figure 10. Comparison of PSF/MTF estimation methods. Our implementation of the PSF estimation algorithm [21, 20] of Joshi and colleagues, Imatest commercial software [18], our previous random target method [14], and the two-scale method proposed in this work (applied to the images of the wall shown in Figure 7). On the low frequencies all algorithms produced very similar estimates, while on the higher frequencies the estimation by Joshi and coauthors depends strongly on the regularization level. Although much effort was made to get a noisefree MTF estimation from the Imatest software, the final estimation is quite noisy. The Imatest estimation is done from a slanted-edge image and only gives an estimation for the MTF at the slanted-edge orthogonal direction. The proposed two-scale algorithm is the one presenting an estimation closest to the nonblind estimation from [14], considered as ground truth by virtue of its high accuracy.
defines it for a function $f \in L^{1}\left(\mathbb{R}^{2}\right)$ in a point $\zeta=\left(\zeta_{1}, \zeta_{2}\right)$. This formula is still valid for functions belonging to $L^{p}\left(\mathbb{R}^{2}\right)$ with $1<p \leq 2$ (see, e.g., [33, 5]).

- Continuous images are defined for $\mathbf{x} \in \mathbb{R}^{2}$, whereas digital images are sampled on a discrete $\operatorname{grid} \mathbf{k} \in \mathbb{Z}^{2}$.
- $\mathbf{S}_{1}: C_{b}^{0} \rightarrow \ell^{\infty}\left(\mathbb{Z}^{2}\right)$ is the 1-sampling operator such that $\mathbf{u}(\mathbf{k})=\left(\mathbf{S}_{1} u\right)(\mathbf{k})$. From the distribution viewpoint $\mathbf{S}_{1}$ is the product by a Dirac comb $\Pi_{s}:=\sum_{\mathbf{k}} \delta_{s \mathbf{k}}$ with $s=1$, namely $\mathbf{S}_{1} u=\Pi_{1} \cdot u$, where $u$ must be a continuous function. Both representations of the sampling operator will be identified, and it will be clear from the context which representation is intended.
- A digital image $\mathbf{u}$ will be represented either as a sequence $(u(\mathbf{k}))_{\mathbf{k}}$ in $\ell^{\infty}\left(\mathbb{Z}^{2}\right)$ or as the corresponding Dirac comb $\mathbf{u}:=\sum_{\mathbf{k} \in \mathbb{Z}^{2}} \mathbf{u}(\mathbf{k}) \delta_{\mathbf{k}}$.
- The operator $\mathbf{I}_{1}: \ell^{2}\left(\mathbb{Z}^{2}\right) \rightarrow \mathcal{B} \mathcal{L}^{2}\left(\mathbb{R}^{2}\right)$ denotes the Shannon-Whittaker interpolator, defined by $\mathbf{I}_{1} \mathbf{u}(\mathbf{x})=\sum_{\mathbf{k} \in \mathbb{Z}^{2}} \mathbf{u}(\mathbf{k}) \operatorname{sinc}(\mathbf{x}-\mathbf{k})$, where $\operatorname{sinc}(\mathbf{x})=\frac{\sin (\pi x)}{\pi x} \frac{\sin (\pi y)}{\pi y}$. We therefore have $\mathbf{I}_{1} \mathbf{u}=\mathbf{F}^{-1}\left(\sum_{\mathbf{k}} \mathbf{u}(\mathbf{k}) e^{-i \mathbf{k} \cdot \xi} \mathbf{1}_{[-\pi, \pi]^{2}}\right)$. When $\mathbf{u} \in \ell^{2}, \mathbf{F}\left(\mathbf{I}_{1} \mathbf{u}\right)$ belongs to $L^{2}$ and is compactly supported. Thus $\mathbf{I}_{1} \mathbf{u} \in \mathcal{B} \mathcal{L}^{2}$, and we have $\mathbf{S}_{1} \mathbf{I}_{1}=I d$.
- The filter $W_{w} u=\mathbf{F}^{-1}\left(\hat{u} \cdot \mathbf{1}_{[-w \pi, w \pi]^{2}}\right)$ is an ideal low-pass filter that cuts the spectrum of $u$ to $[-w \pi, w \pi]^{2}$. It is defined if $\hat{u}$ is a function. Note that if $u \in L^{1} \cup L^{2}$, then $W_{1} u$ is in $\mathcal{B} \mathcal{L}^{2}$.
- $H_{\lambda} u(\mathbf{x})=\lambda^{2} u(\lambda \mathbf{x})$ is the continuous homothecy (i.e., $\lambda>1$ is a contraction); the rationale for its normalization is to preserve the image mean (its zero-frequency coefficient). In the Fourier domain $\mathbf{F}\left(H_{\frac{\lambda}{u}}\right)(\zeta)=\hat{u}\left(\frac{\zeta}{\lambda}\right)$, so if $u$ is $\alpha$-band-limited, then $H_{\frac{1}{\alpha}} u$ is band-limited.
- $\mathcal{S}_{s}: \ell^{2}\left(\mathbb{Z}^{2}\right) \rightarrow \ell^{2}\left(\mathbb{Z}^{2}\right)$ denotes the $s$-to-1-resampling operator $\mathcal{S}_{s}=\mathbf{S}_{1} H_{s} \mathbf{I}_{1}$ (i.e., $s>1$ is a subsampling by $s$ ).
- $\mathbf{C}[\mathbf{u}]: \ell^{2}\left(\mathbb{Z}^{2}\right) \rightarrow \ell^{2}\left(\mathbb{Z}^{2}\right)$ denotes the linear map associated to the convolution with a digital image $\mathbf{u}$. The convolved sequence belongs to $\ell^{2}\left(\mathbb{Z}^{2}\right)$ which in general is satisfied if $\mathbf{u} \in \ell^{1}\left(\mathbb{Z}^{2}\right)$.
- The digital Nyquist homothecy operator $\mathbf{H}_{\alpha}: \ell^{2}\left(\mathbb{Z}^{2}\right) \rightarrow \ell^{2}\left(\mathbb{Z}^{2}\right)$ is defined by $\mathbf{H}_{\alpha} \mathbf{u}:=$ $\mathbf{S}_{1} W_{1} H_{\alpha} \mathbf{I}_{1} \mathbf{u}$. It is a digital contraction if $\alpha>1$.
- Let $\mathbf{L}$ be a bounded linear operator over a Hilbert space. $\mathbf{L}^{*}$ is its adjoint, and $\mathbf{L}^{+}$(if it exists) is its pseudoinverse, i.e., the minimum-norm solution of $\left(\mathbf{L}^{*} \mathbf{L}\right) \mathbf{L}^{+}:=\mathbf{L}^{*}$.
A.1. Physical and mathematical modeling of continuous images. Continuous images will be assumed to be functions in $\mathcal{B L}_{0}^{2}\left(\mathbb{R}^{2}\right)$. This choice is consistent since these functions are continuous (actually $C^{\infty}$ ) and the sampling is well defined. Moreover, as suggested in [26] and later in [1, Appendix A], this choice is sufficiently general to model the continuous landscape observed by a camera just before sampling takes place at the sensors.

In fact, even if the raw physical image before blur and sampling is, realistically, a positive Radon measure $O$ (due to the photon-counting nature of sensitive digital systems) with compact support (imposed by the finite number of photons), it will still be blurred by the camera PSF $h$ which will be regular enough for $h * O$ to be in $\mathcal{B} \mathcal{L}_{0}^{2}$.

How regular can it realistically be assumed to be? The kernel $h$ originates in several physical phenomena, namely diffraction, antialiasing filtering, and sensor integration. Each one of these phenomena, and their combination as well, lead to model $h$ as a nonnegative function with finite mass $\int h=1$ (normalized to 1 ). In addition the diffraction part ensures that $\hat{h}$ is compactly supported. From this one deduces that $h \in \mathcal{B} \mathcal{L}_{0}^{2} \cap L^{1}$.

We now turn to the problem of simplifying $O$ to a more manageable function $u$, which is
indistinguishable from $O$ after convolution with the PSF $h$. Let $B=\operatorname{supp}(\hat{h})$ be the (compact) spectral support of the PSF $h$. Hence $h$ can be idempotently written as $h=h * h_{0}$, where $h_{0} \in \mathcal{S}^{\prime}$ has a compactly supported spectrum satisfying $\hat{h_{0}}(\eta)=1$ for $\eta \in B$. The function $\hat{h}_{0}$ can easily be constructed by an explicit formula as a $C^{\infty}$ and compactly supported function satisfying $\hat{h}_{0}(\eta)=1$ on $B$. Then its inverse Fourier transform has all required properties.

So we have

$$
v=h * O=h * u, \quad \text { where } u=h_{0} * O .
$$

Consequently, the observed landscape can be assumed without loss of generality to be $u=$ $h_{0} * O$ instead of $O$. Being the convolution of a compactly supported positive Radon measure $O \in \mathcal{E}^{\prime}$ with $h_{0} \in \mathcal{B} \mathcal{L}_{0}^{2} \cap L^{1}, u$ also belongs to $\mathcal{B} \mathcal{L}_{0}^{2}$, and its convolution with $h \in \mathcal{B} \mathcal{L}_{0}^{2} \cap L^{1}$ is the observed image $v \in \mathcal{B} \mathcal{L}_{0}^{2}$.

Appendix B. Standard results from Fourier analysis. The following two main results from standard Fourier analysis and distribution theory are stated without proof. The reader is referred to, e.g., $[33,17]$ for the proofs in the particular setting chosen here.

Proposition B. 1 (convolution through Fourier transform). The relation

$$
\begin{equation*}
\mathbf{F}(f * g)=\mathbf{F}(f) \cdot \mathbf{F}(g) \tag{B.1}
\end{equation*}
$$

is valid in either of these cases:

1. $g \in L^{1}\left(\mathbb{R}^{2}\right)$ and $f \in L^{p}\left(\mathbb{R}^{2}\right)$ for $1 \leq p \leq 2$. Then $f * g$ belongs to $L^{p}\left(\mathbb{R}^{2}\right)$ (see [33, Theorem 2.6]).
2. $g \in \mathcal{E}^{\prime}$ and $f \in \mathcal{S}^{\prime}$. Then $f * g$ belongs to $\mathcal{S}^{\prime}$ (see [17, Theorem 7.1.15]).

Applying the Fourier transform on both sides of (B.1) and recalling that the squared Fourier transform operator $\mathbf{F}^{2}(u)=(2 \pi)^{2}[\mathbf{x} \mapsto u(-\mathbf{x})]$ is almost the identity (except for flipping and a constant factor), we obtain the following corollary.

Corollary B. 2 (product through Fourier transform). The relation

$$
\begin{align*}
\mathbf{F}(f \cdot g) & =\frac{1}{(2 \pi)^{2}} \mathbf{F}(f) * \mathbf{F}(g),  \tag{B.2}\\
\mathbf{F}^{-1}(f \cdot g) & =\mathbf{F}^{-1}(f) * \mathbf{F}^{-1}(g)
\end{align*}
$$

holds when $\hat{g} \in \mathcal{E}^{\prime}$ and $f \in \mathcal{S}^{\prime}$. Then $f \cdot g$ belongs to $\mathcal{S}^{\prime}$.
Proposition B. 3 (Poisson formula in $\mathbb{R}^{2}$ for tempered distributions [17]).

$$
\begin{equation*}
\hat{\Pi}_{1}=(2 \pi)^{2} \Pi_{2 \pi} \tag{B.3}
\end{equation*}
$$

Lemma B.4. If $\hat{u} \in \mathcal{E}^{\prime}$, then

$$
\begin{equation*}
\mathbf{F}\left(\Pi_{1} \cdot u\right)=\Pi_{2 \pi} * \hat{u} . \tag{B.4}
\end{equation*}
$$

Proof. We can apply the first form of Corollary B.2, where $f=\Pi_{1} \in \mathcal{S}^{\prime}$ and $\hat{g}=\hat{u} \in \mathcal{E}^{\prime}$, to obtain

$$
\mathbf{F}\left(\Pi_{1} \cdot u\right)=(2 \pi)^{-2} \hat{\Pi}_{1} * \hat{u}=\Pi_{2 \pi} * \hat{u}
$$

where the last equality is deduced from the Poisson formula (B.3).

The Shannon-Whittaker sampling theorem is then a direct consequence of the two previous results.

Proposition B. 5 (Nyquist-Shannon theorem). If $u \in \mathcal{B} \mathcal{L}^{2}\left(\mathbb{R}^{2}\right)$, then

$$
\begin{equation*}
u=\mathbf{I}_{1} \mathbf{S}_{1} u \tag{B.5}
\end{equation*}
$$

Proof. We can apply Lemma B. 4
Multiplying both sides of (B.4) by $\mathbf{F}$ (sinc) $=\mathbf{1}_{[B]}$, we obtain

$$
\begin{aligned}
\mathbf{F}(\operatorname{sinc}) \cdot \mathbf{F}\left[\mathbf{S}_{1} u\right] & =\mathbf{F}(\operatorname{sinc}) \cdot\left[\Pi_{2 \pi} * \hat{u}\right] \\
& =\sum_{\mathbf{k} \in \mathbb{Z}^{2}} \hat{u}(\cdot+2 \pi \mathbf{k}) \mathbf{1}_{[B]} \\
& =\hat{u},
\end{aligned}
$$

where in the right-hand side the only nonnull term is for $\mathbf{k}=0$ because $u$ is band-limited in $B=[-\pi, \pi]^{2}$ and $\mathbf{F}(\operatorname{sinc})=\mathbf{1}_{[B]}$. Finally, using the second form of Corollary B.2, we obtain

$$
\operatorname{sinc} *\left(\mathbf{S}_{1} u\right)=u
$$

and the left term is by definition $\mathbf{I}_{1} \mathbf{S}_{1} u$.
Corollary B.6. If $u \in L^{2}$ is $s$-band-limited, then

$$
\begin{equation*}
u=H_{s} \mathbf{I}_{1} \mathbf{S}_{1} H_{\frac{1}{s}} u \tag{B.6}
\end{equation*}
$$

## Appendix C. Proof of main results of sections 2 and 3.

Common hypotheses. According to the discussion in Appendix A, and in order to justify all of the lemmas and propositions, we will require that

$$
\begin{aligned}
& -h \in \mathcal{B L}_{0}^{2} \cap L^{1}\left(\mathbb{R}^{2}\right) \text {, nonnegative, } \hat{h}(0)=1 ; \\
& -u \in \mathcal{B} \mathcal{L}_{0}^{2} .
\end{aligned}
$$

This ensures that the convolution $u * h=v$ is well defined with $u \in \mathcal{B L}_{0}^{2}$.
For the uniqueness of the interimage kernel we shall additionally assume that $\hat{u}$ does not vanish inside $\left[-\frac{s}{\lambda_{2}} \pi, \frac{s}{\lambda_{2}} \pi\right]$.

Main results. We now prove several properties that are used throughout the paper.
Lemma C.1. If $u \in \mathcal{B L}_{0}^{2}$, then $\mathbf{S}_{1} u \in \ell^{2}\left(\mathbb{Z}^{2}\right)$.
Proof. As $u$ is in $\mathcal{B L}_{0}^{2}$, there exists $s>0$ such that $\hat{u} \subset[-s \pi, s \pi]$. Furthermore, since $\hat{u} \in \mathcal{E}^{\prime}$, applying (B.4) we have $\mathbf{F}\left(\mathbf{S}_{1} u\right)=(2 \pi)^{2} \Pi_{2 \pi} * \hat{u}$. Since $u$ belongs to $L^{2}$, then $\hat{u}$ is again in $L^{2}$. Thus, $\Pi_{2 \pi} * \hat{u}$ is the $2 \pi$-periodic version of an $L^{2}$ function in $[-s \pi, s \pi]$. Consequently the inverse Fourier transform of $\Pi_{2 \pi} * \hat{u}$ is a Dirac comb whose coefficients are the Fourier series coefficients of $\hat{u}$. Thus the coefficients of $\mathbf{S}_{1} u$ form an $\ell^{2}$ sequence.

Proposition C.2. Let $h \in L^{1}\left(\mathbb{R}^{2}\right)$ and $u, v \in L^{1} \cup L^{2}\left(\mathbb{R}^{2}\right)$. The following equalities hold:

$$
\begin{align*}
& W_{1}(h * v)=W_{1} h * v=h * W_{1} v  \tag{C.1}\\
& W_{1} H_{\lambda} v=H_{\lambda} W_{\frac{1}{\lambda}} v  \tag{C.2}\\
& H_{\alpha}(u * v)=H_{\alpha} u * H_{\alpha} v . \tag{C.3}
\end{align*}
$$

Proof. This is the proof of (C.1). In the Fourier domain,

$$
\mathbf{F}\left(W_{1}(h * v)\right) \stackrel{\text { def }}{=} \mathbf{F}(h * v) \cdot \mathbf{1}_{[-\pi, \pi]^{2}} \stackrel{(\mathrm{~B}, 1)}{=} \mathbf{F}(h) \cdot \mathbf{F}(v) \cdot \mathbf{1}_{[-\pi, \pi]^{2}} .
$$

Thus,

$$
\mathbf{F}(h) \cdot \mathbf{F}(v) \cdot \mathbf{1}_{[-\pi, \pi]^{2}}=\mathbf{F}(h) \cdot \mathbf{1}_{[-\pi, \pi]^{2}} \cdot \mathbf{F}(v) \cdot \mathbf{1}_{[-\pi, \pi]^{2}},
$$

and all results are deduced from this last statement.
Proof. This is the proof of (C.2). Since

$$
\mathbf{F}\left(H_{\lambda} v\right)=\lambda^{2} \mathbf{F}(v(\lambda \cdot))=\lambda^{2} \frac{1}{\lambda^{2}} \hat{v}(\dot{\dot{\lambda}})=\lambda^{2} H_{\frac{1}{\lambda}} \hat{v},
$$

we have

$$
\mathbf{F}\left(W_{1} H_{\lambda} v\right) \stackrel{(\mathrm{B} .1)}{=} \mathbf{F}\left(H_{\lambda} v\right) \cdot \mathbf{1}_{[-\pi, \pi]^{2}}=\lambda^{2} H_{\frac{1}{\lambda}} \hat{v} \cdot \mathbf{1}_{[-\pi, \pi]^{2}} .
$$

On the other hand,

$$
\begin{aligned}
\mathbf{F}\left(H_{\lambda} W_{\frac{1}{\lambda}} u\right) & =\lambda^{2} H_{\frac{1}{\lambda}} \mathbf{F}\left(W_{\frac{1}{\lambda}} v\right) \\
& \stackrel{(\text { B.1) }}{=} \lambda^{2} H_{\frac{1}{\lambda}}\left(\hat{u} \cdot \mathbf{1}_{\left[-\frac{\pi}{\lambda}, \frac{\pi}{\lambda}\right]}\right) \\
& =\lambda^{2}\left(H_{\frac{1}{\lambda}} \hat{v}\right) \cdot \mathbf{1}_{[-\pi, \pi]^{2}} .
\end{aligned}
$$

Proof. This is the proof of (C.3). The proof is a mere change of variables:

$$
\begin{aligned}
H_{\alpha}(u * v)(\mathbf{x}) & =\alpha^{2} \int u(\mathbf{s}) v(\alpha \mathbf{x}-\mathbf{s}) \mathrm{d} \mathbf{s} \\
& =\alpha^{4} \int u(\alpha \mathbf{s}) v(\alpha \mathbf{x}-\alpha \mathbf{s}) \mathrm{d} \mathbf{s} \\
& =\left(H_{\alpha} u * H_{\alpha} v\right)(\mathbf{x}) .
\end{aligned}
$$

Lemma C.3. Let $u, v \in \mathcal{B L}_{0}^{2}\left(\mathbb{R}^{2}\right)$. If either $u$ or $v$ is band-limited, then

$$
\begin{equation*}
\mathbf{S}_{1}(u * v)=\mathbf{S}_{1} \bar{u} * \mathbf{S}_{1} \bar{v} \tag{C.4}
\end{equation*}
$$

where we have called $\bar{u}=W_{1} u$ and $\bar{v}=W_{1} v$.
Proof. We will prove this statement in the tempered distribution sense. We will consider $\mathbf{S}_{1} u=\Pi_{1} \cdot u=\sum_{\mathbf{k}} \delta_{\mathbf{k}} \cdot u$ as a Dirac comb. The application of $\mathbf{S}_{1}$ to $\bar{u}, \bar{v}$, and $u * v$ is well defined as all functions are in $\mathcal{B} \mathcal{L}^{2}\left(\mathbb{R}^{2}\right)$ and by consequence they are in $C^{\infty}$. Recall that if $u \in \mathcal{D}^{\prime}$ and $f$ is $C^{\infty}$, then $f \cdot u \in \mathcal{D}^{\prime}$; thus in this framework we need a function to be in $C^{\infty}$ to be sampled.

From Lemma C. 1 we know that the sequences of coefficients from $\mathbf{S}_{1} \bar{u}$ and $\mathbf{S}_{1} \bar{v}$ are in $\ell^{2}\left(\mathbb{Z}^{2}\right)$. Thus $\left(\mathbf{S}_{1} \bar{u}\right) *\left(\mathbf{S}_{1} \bar{v}\right)$ is a bounded sequence, and therefore every term is well defined.

Finally $\mathbf{F}\left(\mathbf{S}_{1}(u * v)\right)=\Pi_{2 \pi} *(\hat{u} \cdot \hat{v})=\left(\Pi_{2 \pi} * \hat{u}\right) \cdot\left(\Pi_{2 \pi} * \hat{v}\right)$ is true because all considered functions happen to be $2 \pi$-periodizations of compactly supported functions in $(-\pi, \pi)^{2}$, namely $\hat{u}, \hat{v}$, and their product.

Proposition C. 4 (discrete camera model). Let $u \in \mathcal{B} \mathcal{L}_{0}^{2}$ and $h \in L^{1} \cap \mathcal{B} \mathcal{L}_{0}^{2}$, band-limited in $[-s \pi, s \pi]^{2}$. Then

$$
\begin{equation*}
\mathbf{S}_{1}(u * h)=\mathcal{S}_{s}(\overline{\mathbf{u}} * \mathbf{h}) \tag{C.5}
\end{equation*}
$$

where we have called $\overline{\mathbf{u}}=\mathbf{S}_{1} W_{1} H_{\frac{1}{s}} u$ and $\mathbf{h}=\mathbf{S}_{1} H_{\frac{1}{s}} h$.
Proof. We first derive the expression and then ${ }^{s}$ justify the application of each result:

$$
\begin{aligned}
\mathbf{S}_{1}(u * h) & =\mathbf{S}_{1} H_{s} H_{\frac{1}{s}}(u * h) \\
& \stackrel{(\mathrm{C} .1)}{=} \mathbf{S}_{1} H_{s} H_{\frac{1}{s}}\left(W_{s} u * h\right) \\
& \stackrel{(\mathrm{C} .3)}{=} \mathbf{S}_{1} H_{s}\left(H_{\frac{1}{s}} W_{s} u * H_{\frac{1}{s}} h\right) \\
& \stackrel{(\mathrm{C} .2)}{=} \mathbf{S}_{1} H_{s}\left(W_{1} H_{\frac{1}{s}} u * H_{\frac{1}{s}} h\right) \\
& \stackrel{(\mathrm{B.5)}}{=} \mathbf{S}_{1} H_{s} \mathbf{I}_{1} \mathbf{S}_{1}\left(W_{1} H_{\frac{1}{s}} u * H_{\frac{1}{s}} h\right) \\
& \stackrel{(\mathrm{C} .4)}{=} \mathbf{S}_{1} H_{s} \mathbf{I}_{1}\left(\mathbf{S}_{1} W_{1} H_{\frac{1}{s}} u * \mathbf{S}_{1} H_{\frac{1}{s}} h\right) \\
& \stackrel{\text { def }}{=} \mathbf{S}_{1} H_{s} \mathbf{I}_{1}(\overline{\mathbf{u}} * \mathbf{h}) \\
& \stackrel{\text { def }}{=} \mathcal{S}_{s}(\overline{\mathbf{u}} * \mathbf{h})
\end{aligned}
$$

First note that as $u \in \mathcal{B} \mathcal{L}_{0}^{2}$ and $h \in L^{1}$, we can apply (C.1) and (C.3) directly. As $W_{1} u$ is in $\mathcal{B} \mathcal{L}^{2}$, we can apply (C.2). The Nyquist theorem (B.5) is valid since $u \in L^{2}$ and $h \in L^{1}$; then $W_{1} H_{\frac{1}{s}} u * H_{\frac{1}{s}} h$ belongs to $\mathcal{B} \mathcal{L}^{2}$.

Both $W_{1} H_{\frac{1}{s}} u^{s}$ and $H_{\frac{1}{s}} h$ are band-limited finite energy functions so we are free to apply (C.4). Since the sequence $(\overline{\mathbf{u}} * \mathbf{h})$ is the sampling of the band-limited $L^{2}$ function $W_{1} H_{\frac{1}{s}} u * H_{\frac{1}{s}} h$, it belongs to $\ell^{2}$ (Lemma C.1). Finally, the interpolation $\mathbf{I}_{1}(\overline{\mathbf{u}} * \mathbf{h})$ is well defined.

Lemma C.5. Let $h \in L^{1} \cap \mathcal{B} \mathcal{L}_{0}^{2}$ and $k \in \mathcal{B} \mathcal{L}_{0}^{2}$ such that $\hat{k}(\zeta)=\frac{h(\zeta)}{h\left(\frac{\zeta}{\lambda}\right)}$. Assume $\lambda$ large enough to ensure that $\hat{h}(\zeta / \lambda)$ does not vanish in the support of $\hat{k}$. Then if $\lambda>1$, we have

$$
\lim _{n \rightarrow \infty} H_{\lambda^{n-1}} k * \cdots * H_{\lambda} k * k=h
$$

where the limit is in $L^{2} \cap C^{0}$.
Proof. Let us call $u_{n}=H_{\lambda^{n-1}} k * \cdots * H_{\lambda} k * k$. Then in the Fourier domain we have

$$
\begin{aligned}
\lim _{n \rightarrow \infty} \hat{u}_{n}(\zeta) & =\lim _{n \rightarrow \infty} \prod_{i=0}^{n-1} \hat{k}\left(\frac{\zeta}{\lambda^{i}}\right) \\
& =\lim _{n \rightarrow \infty} \frac{\hat{h}(\zeta)}{\hat{h}\left(\zeta / \lambda^{n}\right)}
\end{aligned}
$$

Since $h \in L^{1}, \hat{h} \in C^{0}$ and we have

$$
\lim _{n \rightarrow \infty} \hat{h}\left(\zeta / \lambda^{n}\right)=\hat{h}(0)=1
$$

The convergence is uniform on a fixed compact set because $\hat{h}$ is continuous and compactly supported. This implies that the convergence holds in $L^{1}$ and $L^{2}$. Therefore

$$
H_{\lambda^{n-1}} k * \cdots * H_{\lambda} k * k \xrightarrow{L^{2} \cap C^{0}} h .
$$

## Appendix D. Stability of the interimage kernel estimation.

Lemma D.1. Let A be an injective bounded linear operator (IBLO) defined on a Banach space $X$ and let $\Delta \mathbf{A}$ be a perturbation of $\mathbf{A}$ such that $\mathbf{A}+\Delta \mathbf{A}$ is also an IBLO and $\|\mathbf{A}\|\|\Delta \mathbf{A}\|<$ 1. Let $\mathbf{b} \in X$ and let $\Delta \mathbf{b}$ be a perturbation of $\mathbf{b}$. Then, the solutions of $\mathbf{x}=\mathbf{A}^{+} \mathbf{b}$ and $\mathbf{x}^{*}=(\mathbf{A}+\Delta \mathbf{A})^{+}(\mathbf{b}+\delta \mathbf{b})$ satisfy

$$
\begin{equation*}
\frac{\left\|\mathbf{x}^{*}-\mathbf{x}\right\|}{\|\mathbf{x}\|} \leq \frac{\operatorname{cond}(\mathbf{A})}{1-\left\|\mathbf{A}^{+} \Delta \mathbf{A}\right\|}\left(\frac{\|\delta \mathbf{b}\|}{\|\mathbf{b}\|}+\frac{\|\Delta \mathbf{A}\|}{\|\mathbf{A}\|}\right), \tag{D.1}
\end{equation*}
$$

where $\operatorname{cond}(\mathbf{A})=\|\mathbf{A}\|\left\|\mathbf{A}^{+}\right\|$.
Proof. First note that as $\mathbf{A}$ is full rank, the pseudoinverse is the left inverse of $\mathbf{A}$, namely $\mathbf{A}^{+} \mathbf{A}=\mathbf{I}$. Since $\|\mathbf{A}\|\|\Delta \mathbf{A}\|<1$, we have that

$$
(\mathbf{A}+\Delta \mathbf{A})^{+}=\left(\mathbf{I}+\mathbf{A}^{+} \Delta \mathbf{A}\right)^{-1} \mathbf{A}^{+},
$$

and we also have

$$
\left\|\left(\mathbf{I}+\mathbf{A}^{+} \Delta \mathbf{A}\right)^{-1}\right\|=\left\|\sum\left(\mathbf{A}^{+} \Delta \mathbf{A}\right)^{k}\right\| \leq \sum\left\|\left(\mathbf{A}^{+} \Delta \mathbf{A}\right)\right\|^{k}=\frac{1}{1-\left\|\mathbf{A}^{+} \Delta \mathbf{A}\right\|}
$$

Hence,

$$
\begin{aligned}
\mathbf{x}^{*}-\mathbf{x} & =(\mathbf{A}+\Delta \mathbf{A})^{+}(\mathbf{b}+\delta \mathbf{b})-\mathbf{A}^{+} \mathbf{b} \\
& =\left(\mathbf{I}+A^{+} \Delta \mathbf{A}\right)^{-1} \mathbf{A}^{+}(\mathbf{b}+\delta \mathbf{b})-\mathbf{A}^{+} \mathbf{b} ;
\end{aligned}
$$

therefore

$$
\begin{aligned}
\left(\mathbf{I}+\mathbf{A}^{+} \Delta \mathbf{A}\right)\left(\mathbf{x}^{*}-\mathbf{x}\right) & =\mathbf{A}^{+}(\mathbf{b}+\delta \mathbf{b})-\mathbf{A}^{+} \mathbf{b}-\mathbf{A}^{+} \Delta \mathbf{A} \mathbf{A}^{+} \mathbf{b} \\
& =\mathbf{A}^{+}(\delta \mathbf{b}-\Delta \mathbf{A x}),
\end{aligned}
$$

and then

$$
\begin{aligned}
\frac{\left\|\mathbf{x}^{*}-\mathbf{x}\right\|}{\|\mathbf{x}\|} & \leq \frac{\left\|\mathbf{A}^{+}\right\|}{1-\left\|\mathbf{A}^{+} \Delta \mathbf{A}\right\|} \frac{\|\delta \mathbf{b}\|+\|\Delta \mathbf{A} \mathbf{x}\|}{\|\mathbf{x}\|} \\
& =\frac{\operatorname{con}(\mathbf{A})}{1-\left\|\mathbf{A}^{+} \Delta \mathbf{A}\right\|} \frac{\|\delta \mathbf{b}\|+\|\Delta \mathbf{A} \mathbf{x}\|}{\|\mathbf{A}\|\|\mathbf{x}\|} \\
& \leq \frac{\operatorname{cond}(\mathbf{A})}{1-\left\|\mathbf{A}^{+} \Delta \mathbf{A}\right\|}\left(\frac{\|\delta \mathbf{b}\|}{\|\mathbf{A x}\|}+\frac{\|\Delta \mathbf{A}\|\|\mathbf{x}\|}{\|\mathbf{A}\|\|\mathbf{x}\|}\right) \\
& \leq \frac{\operatorname{cond}(\mathbf{A})}{1-\left\|\mathbf{A}^{+} \Delta \mathbf{A}\right\|}\left(\frac{\|\delta \mathbf{b}\|}{\|\mathbf{b}\|}+\frac{\|\Delta \mathbf{A}\|}{\|\mathbf{A}\|}\right) .
\end{aligned}
$$

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## REFERENCES

[1] A. Almansa, S. Durand, and B. Rougé, Measuring and improving image resolution by adaptation of the reciprocal cell, J. Math. Imaging Vision, 21 (2004), pp. 235-279.
[2] S. M. Backman, A. J. Makynen, T. T. Kolehmainen, and K. M. Ojala, Fast lens testing using random targets, in Opto-Ireland 2002: Optics and Photonics Technologies and Applications, Proceedings of SPIE 4876, 2003, pp. 1100-1109.
[3] S. M. Backman, A. J. Makynen, T. T. Kolehmainen, and K. M. Ojala, Random target method for fast MTF inspection, Opt. Express, 12 (2004), pp. 2610-2615.
[4] E. H. Barney Smith, PSF estimation by gradient descent fit to the ESF, in Image Quality and System Performance III, Proceedings of SPIE 6059, 2006, 60590E.
[5] J.-M. Bony, Cours d'analyse. Théorie des distributions et analyse de Fourier, Les Éditions de l'École Polytechnique, Palaiseau, France, 2001.
[6] F. L. Bookstein, Principal warps: Thin-plate splines and the decomposition of deformations, IEEE Trans. Pattern Anal. Mach. Intell., 11 (1989), pp. 567-585.
[7] J. Brauers, C. Seiler, and T. Aach, Direct PSF estimation using a random noise target, in Digital Photography, Proceedings of SPIE 7537, F. H. Imai, N. Sampat, and F. Xiao, eds., 2010, New York.
[8] D. Capel, Image Mosaicing and Super-Resolution (CPHC/BCS Distinguished Dissertations), SpringerVerlag, New York, 2004.
[9] B. Chalmond, PSF estimation for image deblurring, CVGIP: Graphical Models and Image Processing, 53 (1991), pp. 364-372.
[10] S. Chaudhuri and A. N. Rajagopalan, Depth from Defocus: A Real Aperture Imaging Approach, Springer-Verlag, New York, 1999.
[11] C. D. Claxton and R. C. Staunton, Measurement of the point-spread function of a noisy imaging system, J. Opt. Soc. Am. A, 25 (2008), pp. 159-170.
[12] A. Daniels, G. D. Boreman, A. D. Ducharme, and E. Sapir, Random transparency targets for modulation transfer function measurement in the visible and infrared regions, Opt. Eng., 34 (1995), pp. 860-868.
[13] M. Delbracio, A. Almansa, and P. Musé, Recovering the Subpixel PSF from Two Photographs at Different Distances, preprint, http://www.ipol.im/pub/pre/H3/ (accessed November 1, 2012).
[14] M. Delbracio, P. Musé, A. Almansa, and J. M. Morel, The non-parametric sub-pixel local point spread function estimation is a well posed problem, Int. J. Comput. Vision, 96 (2012), pp. 175-194.
[15] J. W. Goodman, Introduction to Fourier Optics, McGraw-Hill Science/Engineering/Math, New York, 1996.
[16] G. Healey and R. Kondepudy, Radiometric CCD camera calibration and noise estimation, IEEE Trans. Pattern Anal. Mach. Intell., 16 (1994), pp. 267-276.
[17] L. Hörmander, The Analysis of Linear Partial Differential Operators. I. Distribution Theory and Fourier Analysis, Grundlehren Math. Wiss., Springer-Verlag, Berlin, 1983.
[18] Imatest LLC, Imatest 3.6, http://www.imatest.com/ (2010).
[19] ISO, 12233:2000 Photography-Electronic Still-Picture Cameras-Resolution Measurements, 2000.
[20] N. Joshi, Enhancing Photographs Using Content-Specific Image Priors, Ph.D. thesis, Department of Computer Science and Engineering, University of California, San Diego, CA, 2008.
[21] N. Joshi, R. Szeliski, and D. J. Kriegman, PSF estimation using sharp edge prediction, in Proceedings of the IEEE Conference on Computer Vision and Pattern Recognition, Anchorage, AK, 2008.
[22] E. Levy, D. Peles, M. Opher-Lipson, and S. G. Lipson, Modulation transfer function of a lens measured with a random target method, Appl. Opt., 38 (1999), pp. 679-683.
[23] D. G. Lowe, Object recognition from local scale-invariant features, in Proceedings of the Seventh IEEE International Conference on Computer Vision, Vol. 2, IEEE Computer Society, Los Alamitos, CA, 1999, pp. 1150-1157.
[24] M. LuXEN AND W. FÖRSTNER, Characterizing image quality: Blind estimation of the point spread function from a single image, in Proceedings of Photogrammetric Computer Vision, Graz, Austria, 2002.
[25] A. Marion, Acquisition et Visualisation des Images, Eyrolles, Paris, 1997.
[26] J.-M. Morel and S. Ladjal, Notes sur l'analyse de Fourier et théorie de Shannon en traitement d'images, in Analyse de Fourier et traitement dimages. Journées X-UPS, Ecole Polytechnique, Palaiseau, France, 1998, pp. 37-100.
[27] J.-M. Morel and G. Yu, ASIFT: A new framework for fully affine invariant image comparison, SIAM J. Imaging Sci., 2 (2009), pp. 438-469.
[28] M. K. Ng, R. H. Chan, and W. C. Tang, A fast algorithm for deblurring models with Neumann boundary conditions, SIAM J. Sci. Comput., 21 (1999), pp. 851-866.
[29] S. C. Park, M. K. Park, and M. G. Kang, Super-resolution image reconstruction: A technical overview, IEEE Signal Processing Mag., 20 (2003), pp. 21-36.
[30] S. E. Reichenbach, R. Park, and S. K. Narayanswamy, Characterizing digital image acquisition devices, Opt. Eng., 30 (1991), pp. 170-177.
[31] D. Robinson and P. Milanfar, Statistical performance analysis of super-resolution, IEEE Trans. Image Process., 15 (2006), pp. 1413-1428.
[32] R. Sprengel, K. Rohr, and H. S. Stiehl, Thin-plate spline approximation for image registration, in Engineering in Medicine and Biology Society, 1996. Bridging Disciplines for Biomedicine, Proceedings of the 18th Annual International Conference of the IEEE, Vol. 3, 1996, pp. 1190-1191.
[33] E. M. Stein and G. Weiss, Introduction to Fourier Analysis on Euclidean Spaces, Princeton Math. Ser. 32, Princeton University Press, Princeton, NJ, 1971.
[34] H. Tian, B. Fowler, and A. E. Gamal, Analysis of temporal noise in CMOS photodiode active pixel sensor, IEEE J. Solid-State Circuits, 36 (2001), pp. 92-101.
[35] C. S. Williams and O. A. Becklund, Introduction to the Optical Transfer Function, SPIE Press Monogr. PM112, SPIE Press, Bellingham, WA, 2002.
[36] O. Yadid-Pecht, Geometrical modulation transfer function for different pixel active area shapes, Opt. Eng., 39 (2000), pp. 859-865.
[37] G. Yu and J.-M. Morel, ASIFT: An algorithm for fully affine invariant comparison, Image Processing On Line (2011); available online at http://dx.doi.org/10.5201/ipol.2011.my-asift.
[38] J. A. Zandhuis, D. Pycock, S. F. Quigley, and P. W. Webb, Sub-pixel non-parametric PSF estimation for image enhancement, IEE Proceedings: Vision, Image, and Signal Processing, 144 (1997), pp. 285-292.
[39] T. Zhao, R. Wang, Y. Liu, and F. Yu, Characteristic-analysis of optical low pass filter used in digital camera, in ICO20: Optical Design and Fabrication, Proceedings of the SPIE 6034, 2006, pp. 170-178.


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[^1]:    ${ }^{1}$ Recall that the Hilbert-Schmidt norm is $\sum_{i}\left\|\mathbf{L} e_{i}\right\|^{2}$, where $\left\{e_{i}\right\}$ is any Hilbert basis of the domain of $\mathbf{L}$. If the linear operator is a matrix, then the Hilbert-Schmidt norm is the Frobenius norm of the matrix.

